# TABLE OF CONTENTS

Numerical Simulation of High Electron Mobility Transistors based on the Spectral Element Method  
Feng Li, Qing H. Liu, and David P. Klemer ................................................................. 1144

An Efficient ACA Solution for Electromagnetic Scattering from Discrete Body of Revolution  
Zhenhong Fan, Zi He, and Rushan Chen ......................................................................... 1151

Electromagnetic Scattering from a PEMC Circular Cylinder Coated by Topological Insulator (TI)  
Anjum Shahzad, Ahsan Illahi, Shakeel Ahmed, Akhtar Hussain, and Qaisar A. Naqvi ....... 1158

Nonlinear Analysis and Performance Improvement of Amplifying Aperture Coupled Reflectarray Antenna  
Iman Aryanian, Abdolali Abdipour, and Gholamreza Moradi ........................................ 1164

Parameterized Model Order Reduction for Efficient Time and Frequency Domain Global Sensitivity Analysis of PEEC Circuits  
Luca De Camillis, Giulio Antonini, and Francesco Ferranti............................................ 1170

Finite Difference Analysis of an Open-Ended, Coaxial Sensor Made of Semi-Rigid Coaxial Cable for Determination of Moisture in Tenera Oil Palm Fruit  
Ee M. Cheng, Zulkifly Abbas, Mohamed Fareq Abdul Malek, Kim Y. Lee, Kok Y. You, Shing F. Khor, Jumiah Hassan, and Hishammudin Zainuddin ........................................ 1181

Design of PSR with Different Feed Configurations and Partition Lens System for Skin Cancer Treatment  
Petrishia Arockiasamy and Sasikala Mohan .................................................................. 1193

Design, Simulation and Fabrication of a Wide Bandwidth Envelope Tracking Power Amplifier  
Iman Aryanian, Abdolali Abdipour, and Abbas Mohammadi ......................................... 1202

Analysis of Control Variables to Maximize Output Power for Switched Reluctance Generators in Single Pulse Mode Operation  
Paiprote Thongprasri and Supat Kittiratsatcha .............................................................. 1208
Electromagnetic Coupling Analysis of Transient Excitations of Rectangular Cavity through Slot using TD-EFIE with Laguerre Polynomials as Temporal Basis Functions
Dorsaf Omri and Taoufik Aguili ................................................................. 1221

Ultra-Wideband Balanced Bandpass Filters Based on Transversal Signal-Interference Concepts
Chaoying Zhao, Wenjie Feng, Yuanchuan Li, and Wenquan Che ......................... 1232

Novel Pentagonal Dual-Mode Filters with Adjustable Transmission Zeros
Zhaojun Zhu, Shuo Liang, and Chaolei Wei ...................................................... 1238

Miniaturized Microstrip Suppressing Cell with Wide Stopband
Mohsen Hayati, Farzin Shama, and Milad Ekhteraei ........................................... 1244

A Low-pass Filter with Sharp Transition and Wide Stop-band Designed based on New Metamaterial Transmission Line
Lin Peng, Yu J. Qiu, Xing Jiang, and Cheng L. Ruan .............................................. 1250

High Selectivity Balanced Filters Based on Transversal Signal-interaction Concepts
Wenjie Feng, Meiling Hong, and Wenquan Che .................................................. 1257
Numerical Simulation of High Electron Mobility Transistors based on the Spectral Element Method

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Abstract — The spectral element method (SEM) is implemented for the numerical simulation of high electron mobility transistors (HEMTs) through a self-consistent solution of the Schrödinger-Poisson equations. The electron conduction band structure and electron density distribution are calculated and plotted, and results compared to those based on methods utilizing a finite-difference approach. Simulation accuracy and efficiency are analyzed and compared with traditional finite difference method (FDM). DC current-voltage (I-V) characteristics for the HEMT structure are simulated, based on a quasi-2D current model. The SEM approach offers advantages in speed and efficiency over FDM, while yielding results which conform well to reported experimental results. These advantages are particularly important for compound heterojunction devices with complex material profiles, for which FDM methods may be inefficient and computationally slow.

Index Terms — Heterojunction, Schrödinger equation, spectral element method, transistor.

I. INTRODUCTION

High Electron Mobility Transistors (HEMTs) – also referred to as modulation-doped FETs (MODFETs), or heterostructure FETs (HFETs) – are field-effect transistors which utilize a channel region formed by a heterojunction of (typically III-V) materials having different band gaps, in contrast to the conventional MOSFET channel formed as a three-dimensional region of doped semiconductor. The characteristics of the HEMT two-dimensional electron gas (2DEG) within the channel are typically studied by solving the Poisson-Schrödinger equations in a self-consistent matter, using various techniques such as the finite difference method [1], finite element method [2], and multigrid method [3], among others. HEMTs and related heterojunction solid-state devices can have highly complex material profiles with very small dimensions (on the scale of angstroms); accordingly, traditional numerical methods may be numerically inefficient and computationally slow, limiting their usefulness in device/circuit design and analysis applications.

The spectral element method (SEM) is a high-order finite element method which combines the advantages of the finite element method and the spectral method, resulting in computational speed advantages – by two to three orders of magnitude – when compared to conventional finite difference methods [4]. Furthermore, application of the spectral element method can result in high accuracy in numerical simulations with reasonable computational effort.

Our present research suggests that, to date, few efforts have been made to apply the spectral element method to the simulation of widely-used semiconductor devices such as the HEMT device. In this work we develop a spectral element method approach for the simulation of the HEMT structure. Although the implementation of the SEM is more complex than that of the finite difference method, we demonstrate that improvements in computational accuracy and efficiency are highly significant. This will be illustrated through comparison with computational results using the finite difference method and analysis of relative $L_2$ errors, (i.e., the root mean square of the error components). In addition, a quasi-two-dimensional scheme is used to study current-voltage characteristics of the HEMT device model in the triode region, based on the electron density distribution calculated from the Schrödinger-Poisson equations describing the HEMT structure.
d = \psi(x) + V(x)\psi(x) = E\psi(x), \quad (1)

where \psi is the wave function, E represents energy, V is potential energy, h is Planck’s constant over 2π, and m* is the effective mass for the electron.

As is typical in semiconductor device modeling, we assume that permittivity is independent of time, and that polarization due to mechanical forces is negligible. Accordingly, the one dimensional Poisson equation is written as:

\[ \frac{\partial}{\partial x} \left( \epsilon_0 \epsilon_r(x) \frac{\partial \psi}{\partial x} \right) + \rho(x) = -\frac{q}{\epsilon_0} \frac{\partial \psi}{\partial x}, \quad \text{where} \quad \epsilon_r(x) = \frac{\epsilon_{\text{vac}}}{\epsilon_{\text{vac}} - \epsilon_{\text{air}}} \epsilon_{\text{air}}, \quad \rho(x) = \frac{\partial \psi}{\partial x}. \quad (2)\]

The 3-D bulk electron density can be expressed as [6]:

\[ n_3D(x) = \frac{(2m^*)^{\frac{3}{2}}}{\pi \hbar^2} \int_0^\infty dE \frac{(E-V)^{\frac{3}{2}}}{1+e^{(E-V)/k_B T }}, \quad (7)\]

where the potential energy V may be related to the electrostatic potential \phi through the equation:

\[ V(x) = -q\phi(x) + \Delta E_c(x), \quad (8)\]

where \Delta E_c is the conduction band offset at the heterointerface.

III. NORMALIZATION AND THE STURM-LIOUVILLE DIFFERENTIAL EQUATION

The Schrödinger equation can be normalized in one dimension as [7]:

\[ \left[ -\frac{\hbar^2}{2m^*} \frac{\partial^2}{\partial x^2} + V(\tilde{x}) \right] \psi(\tilde{x}) = E\psi(\tilde{x}), \quad (9)\]

where \( \tilde{x} = \frac{x}{\Delta x} \), \( V(x) = \frac{\nu}{\nu_0} = \frac{\nu_{\text{air}} - \nu_{\text{air}}}{\nu_{\text{air}}}, \) \( E = \frac{E}{E_0}, \) \( E_0 = \frac{\pi^2 k^2}{2m^* \Delta x^2} m^* (\tilde{x}) = m^* (\tilde{x}) / m^* (0), \) with \( \Delta x \) a reference thickness, e.g., the AlGaAs spacer thickness in the HEMT structure.

The one dimensional Poisson equation can be normalized as:

\[ \left( \frac{\partial}{\partial x} \left( \epsilon_r(\tilde{x}) \frac{\partial \psi}{\partial x} \right) + \frac{\partial \psi}{\partial x} \right) = -\frac{\partial \psi}{\partial x}, \quad (10)\]

where \( \epsilon_r(\tilde{x}) = \frac{\epsilon_{\text{vac}}(\tilde{x})}{\epsilon_{\text{vac}} - \epsilon_{\text{air}}} \epsilon_{\text{air}}, \) and \( \psi(\tilde{x}) = \frac{\partial \psi}{\partial x} \rho(x). \)

These normalized Schrödinger and Poisson equations can be treated as special forms of the Sturm-Liouville differential equation:

\[ \left[ -\frac{d^2}{d\tilde{x}^2} \left( \eta(\tilde{x}) \frac{d}{d\tilde{x}} \right) + \psi(\tilde{x}) \right] u(\tilde{x}) = E u(\tilde{x}) + \psi(\tilde{x}), \quad (11)\]

which becomes the Schrödinger equation for \( \psi(\tilde{x}) = 0 \) and \( \eta(\tilde{x}) = \frac{1}{\pi^2 m^* (\tilde{x})}, \) and the Poisson equation for \( \psi(\tilde{x}) = 0, E = 0, \) and \( \eta(\tilde{x}) = \epsilon_r(\tilde{x}). \)

Finally, multiplying the above equation by the time-independent test function \( \epsilon \) on both sides, and integrating the equation over the spatial domain \( \Omega, \) a weak formulation of the Sturm-Liouville differential equation can be obtained:

\[ \int_{\Omega_k} (\eta(\tilde{x}) \frac{d}{d\tilde{x}} ) u d\tilde{x} + \int_{\Omega_k} \psi u d\tilde{x} = \int_{\Omega_k} E u d\tilde{x} + \int_{\Omega_k} \psi d\tilde{x}, \quad (12)\]

IV. DISCRETIZATION AND MATRIX FORMULATION

One characteristic of the spectral element method is adopted from the finite element method, i.e., the domain under study is divided into \( K \) elements. In implementing this division, the spatial location of any semiconductor device inhomogeneity, e.g., a material heterojunction, must be considered in order to avoid an ambiguity which occurs when different materials (e.g., AlGaAs and GaAs) appear within a single element of the numerical grid. After discretization into elements, integrations involved in the original equations may be performed individually on each element, specifically:

\[ \int_{\Omega_k} (\eta(\tilde{x}) \frac{d}{d\tilde{x}} ) u d\tilde{x} + \int_{\Omega_k} \psi u d\tilde{x} = \int_{\Omega_k} E u d\tilde{x} + \int_{\Omega_k} \psi d\tilde{x}, \quad (13)\]

for \( k = 1, \ldots, K. \)

A mathematical mapping is then implemented, and a global physical coordinate \( x \in [x_k, x_{k+1}] \) for each element \( K \) is mapped into a local coordinate \( \xi \in \Lambda = [-1, 1] \) for the Gauss-Lobatto-Legendre (GLL) integration quadrature. The mapping function takes the form of:

\[ x(\xi) = x_k + \Delta x(\xi) \frac{1}{2}, \quad (14)\]

where \( \Delta x = x_{k+1} - x_k \) is the length of element \( \Omega_k, \) and
\( x_k \) and \( x_{k+1} \) represent respectively the left and right endpoints of element \( \Omega_k \).

As usual, this coordinate transformation requires the inclusion of the so-called Jacobian \( J_k \) within the integrand:

\[
\int_{\Omega_k} f(x) dx = \int_A f^{(k)}(\xi) \left| \frac{dx^{(k)}}{d\xi} \right| d\xi,
\]

where the integrand function \( f(x) \) is an arbitrary function and the superscript \( (k) \) denotes the restriction of \( f(x) \) to element \( k \). \( J_k = \frac{dx^{(k)}}{d\xi} = \Delta x^{(k)} \) is the Jacobian for the \( k^{th} \) element. To perform the integration over \( \Lambda \), GLL quadrature is applied, reducing the integral to a finite weighted sum:

\[
\int_A f(\xi) d\xi = \sum_{i=0}^{N} w_i f(\xi_i),
\]

where \( w_i \) represents the weights of the GLL quadrature.

Following domain decomposition, a GLL interpolation scheme is applied to the function \( u \) in the one-dimensional Sturm-Liouville differential equation above, applied to each element. A test function \( v \) is also defined:

\[
u^{(k)}(\tilde{x}) = \sum_{j=0}^{N} u^{(k)}(\tilde{x}_j) b_j^{(k)}(\tilde{x}), \quad (17)
\]

where \( b_j^{(k)}(\tilde{x}) \) represents the \( N^{th} \)-order GLL interpolation polynomial:

\[
b_j^{(k)}(x) = \frac{-1}{N(N+1) L_N(\xi_j)} \frac{(1-x^2) L_N(\xi_j)}{x-\xi_j}, \quad (19)
\]

in which \( L_N \) represents the Legendre polynomial of the \( N \)th order, and \( L_N' \) its derivative.

Finally, the Sturm-Liouville differential equation can be transformed to:

\[
\sum_{i=0}^{N} \left\{ u^{(k)}(\xi_i) \cdot \sum_{i=0}^{N} \sum_{l=0}^{N} w_i b_l^{(k)}(\xi_i) \cdot \frac{d}{d\xi} \frac{d}{d\xi} \right\} + \sum_{i=0}^{N} \left\{ w_i b_l^{(k)}(\xi_i) \cdot \frac{J_k}{j_k} \right\}
\]

which can be written succinctly in matrix notation as:

\[
AU = \lambda BU + \hat{f}(\phi). \quad (21)
\]

As mentioned earlier, the Sturm-Liouville equation reduces to the Schrödinger equation for \( \bar{\rho} \equiv 0 \), which can be written in the matrix form:

\[
A\Psi = \lambda B\Psi, \quad (22)
\]

where \( A \) is the stiffness matrix, \( B \) is the diagonal matrix by virtue of the quadrature, and \( \Psi \) is the wavefunction to be determined.

Since the matrix \( B \) is a diagonal matrix, equation (22) can be written as a regular eigenvalue problem, which can be solved more efficiently:

\[
\tilde{A}\tilde{\Psi} = \tilde{\lambda}\tilde{\Psi}, \quad (23)
\]

where \( \tilde{A} = B^{-1/2}AB^{-1/2} \), and \( \tilde{\Psi} = B^{1/2}\Psi \).

Similarly, the Sturm-Liouville equation reduces to the Poisson equation for \( \varphi = 0 \) and \( \bar{E} = 0 \), which can also be expressed in matrix form:

\[
A\phi = \hat{f}(\phi), \quad (24)
\]

This equation can be solved using Newton-Raphson iteration in the usual manner:

\[
\phi(n+1) = \phi(n) - A^{-1} \left( \frac{\partial f(\phi(n))}{\partial \phi} \right) A\phi(n) - \hat{f}(\phi(n)), \quad (25)
\]

where the index \( n \) denotes the \( n^{th} \) iteration.

\section{V. BOUNDARY CONDITIONS}

At this point, we consider a typical AlGaAs/GaAs HEMT structure, with the conduction band profile schematically shown in Fig. 1. When solving the Poisson equation for HEMTs, we may choose \( E_f = 0 \); the electrostatic potential at the gate \( (x = 0) \) is determined by the Schottky barrier height and the applied gate voltage: \( \phi(x_0) = \phi_{ms} - V_{gate} \). At the gate \( (x = 0) \), \( \phi(x_0) = \phi_0 \), the value of the electrostatic potential applied to the HEMT gate, a function of the choice of Fermi level \( E_f \).

Intrinsic to the derivation of the weak formulation, Neumann-type boundary conditions are naturally included for the Schrödinger equation in the spectral element method; this is a significant advantage of the SEM approach for solving the Schrödinger equation, as compared with the FDM approach. In contrast, with FDM, the process of handling Neumann-type boundary conditions for the Schrödinger equation is awkward and requires significant numerical effort.

\begin{figure}[h]
\centering
\includegraphics[width=0.5\textwidth]{fig1}
\caption{Conduction band profile for a typical HEMT device.}
\end{figure}

\section{VI. ELECTRON CONDUCTION BAND AND ELECTRON DENSITY DISTRIBUTION}

The physical parameters of a typical AlGaAs/GaAs HEMT device used for numerical testing are summarized in Table 1 below. The device consists of a doped
Al$_{0.3}$Ga$_{0.7}$As layer, 20 nm in depth, with a doping density of $3 \times 10^{18}$/cm$^3$, above an undoped Al$_{0.3}$Ga$_{0.7}$As barrier (spacer) layer, 5 nm in depth, positioned above a deep (175 nm) GaAs buffer layer.

Figure 2 illustrates computational results for the electron conduction band energy and electron density distribution with no external bias, with results shown using both FDM- and SEM-based simulations.

Table 1: HEMT physical parameters used for simulation

<table>
<thead>
<tr>
<th>Material</th>
<th>x</th>
<th>Thickness (Å)</th>
<th>Doping (/cm$^3$)</th>
<th>Relative Dielectric Constant</th>
<th>Effective Mass ($m_0$)</th>
</tr>
</thead>
<tbody>
<tr>
<td>n+ Al$<em>x$Ga$</em>{1-x}$As</td>
<td>0.3</td>
<td>200</td>
<td>3E+18</td>
<td>12.2</td>
<td>0.092</td>
</tr>
<tr>
<td>Al$<em>x$Ga$</em>{1-x}$As</td>
<td>0.3</td>
<td>50</td>
<td>0</td>
<td>13.1</td>
<td>0.092</td>
</tr>
<tr>
<td>GaAs</td>
<td>0</td>
<td>1750</td>
<td>0</td>
<td>13.1</td>
<td>0.067</td>
</tr>
</tbody>
</table>

x represents the Al mole fraction for the Al$_x$Ga$_{1-x}$As material; $m_0$ is the effective mass for electrons.

The expected quantum well at the heterojunction ($x=250$ Angstroms) is clearly evident, as is the accumulation of electron charge carriers (the 2DEG) in the region of the quantum well. The electron distribution “spills out” slightly into the material above the heterojunction ($x < 250$ Angstroms), illustrating the need for an undoped spacer layer to avoid Coulombic interactions between electrons and ionized dopant atoms. It can be seen that FDM and SEM simulations provide consistent results with regard to electron conduction band energy and electron density distribution, supporting the validity of the SEM implementation for numerical solution of the Schrödinger-Poisson equation.

![Fig. 2. Conduction band energy value and electron density distribution for the HEMT device of Table 1, calculated from SEM and FDM simulations.](image)

**VII. SIMULATION EFFICIENCY AND ACCURACY ANALYSIS**

Visual comparisons of graphically plotted solutions (e.g., in Fig. 2) are qualitatively helpful, but lack quantitative rigor. One formal quantitative metric would be the rate of numerical convergence as a function of an increasing number of unknowns. The comparison of simulation time is also useful. Figure 3 shows a comparison of error as a function of simulation CPU time for FDM and SEM. It can be seen that, to reach similar accuracy, SEM is nearly 40 times faster than FDM for this one dimensional simulation.

![Fig. 3. Error as a function of CPU time for FDM and SEM.](image)

Given that computational (CPU) time is directly related to the number of nodes defined for the computer simulation, we can also study $L_2$ errors as a function of grid size (i.e., number of nodes) in lieu of CPU time. Furthermore, instead of directly using grid size in a direction along an axis perpendicular to the device surface, the parameter number of points per wavelength (PPW) is used for normalization.

Figure 4 shows relative $L_2$ error for a simulated static potential computed using the FDM approach, as a function of PPW. It is apparent that $L_2$ error drops below approximately 0.5% if a minimum of 100 points-per-wavelength are chosen for the FDM simulation. In contrast, the SEM-based approach (with GLL order of 2) requires only 3 points-per-wavelength to achieve this level of accuracy, as shown in Fig. 5, demonstrating a significant improvement in efficiency and accuracy over the FDM approach. It is also apparent that $L_2$ error decreases approximately linearly with increases in PPW, as evident in the log-log graphs for both FDM- and SEM-based computational approaches.

Finally, Fig. 5 illustrates that the rate of decrease in $L_2$ error as a function of increasing PPW is greater for higher-order GLLs. To achieve errors in the range 0.1% to 1%, a GLL order of $N = 4$ and PPW value of 4 points per wavelength would suffice for spectral element simulations of HEMT device structures.
Fig. 4. Relative $L_2$ error in computed electrostatic potential as a function of points-per-wavelength, for FDM-based simulations.

Fig. 5. $L_2$ error in computed electrostatic potential as a function of PPW, for SEM-based simulations having various GLL orders $N$ ($N = 1, 2$ and $4$).

VIII. HEMT CURRENT-VOLTAGE CHARACTERISTICS

The direct-current (DC) current-voltage (I-V) characteristics for a HEMT device can be calculated based on a quasi-two-dimensional (quasi-2D) drift-diffusion current model, using computational results of the electron density distribution [8, 9]. Figure 6 shows a schematic illustration of the quasi-2D current model which applies to the HEMT device structure under consideration.

The 2DEG sheet charge density is a function of voltage along the channel:

$$n_s(V) = \int n(V,z)dz,$$  \hspace{1cm} (26)

and the drain current is:

$$I_D = -qWv(x)n_s.$$  \hspace{1cm} (27)

The computation of drain current requires knowledge of the relationship between drift velocity and electric field (the $v - E$ relationship); for increased accuracy, experimental measurements of the nonlinear $v - E$ relationship (from [10]) are used for current calculations.

Device parameters for the simulated AlGaAs/GaAs HEMT are taken from [11], specifically: gate width $W = 60\mu m$, gate length $L = 0.5\mu m$, gate metallization Ti/Pt/Au, Schottky barrier height $\phi_B = 0.58V$, Si-doped $n-Al_{0.28}Ga_{0.72}As$ layer thickness $d_{gs} = 30nm$ with doping concentration $N_d = 1.5 \times 10^{18}/cm^3$, undoped $Al_{0.28}Ga_{0.72}As$ spacer layer thickness of $\delta d = 4 nm$ and mole fraction $x = 0.28$. Source and drain resistances are assumed to be $R_s = R_d = 0.05 \Omega cm$ [12, 13]. These values are highly process-dependent, thus typical values are chosen empirically here.

Based on this quasi-2D current model and knowledge of the electron density distribution (calculated from the Schrödinger-Poisson equations using the spectral element method), the current-voltage characteristics for the HEMT triode region can be determined (Fig. 7). Simulated points are shown by discrete markers in Fig. 7; the solid lines which interconnect simulation points of like gate voltage are provided for convenience in visualizing trends only. These results conform well to experimental data obtained for the triode region in Thomasian et al. [11].

IX. CONCLUSIONS

It is clear that implementation of the spectral element method in numerical simulations of the conduction band
structure and 2DEG electron distribution in HEMT devices offers significant advantages in numerical efficiency and relative accuracy when compared to less-complex methods, such as the finite difference method. Furthermore, results from SEM-based simulation can facilitate the determination of device terminal I-V characteristics in a much more computationally-efficient manner, as compared to traditional methods. Estimation of AC small-signal parameters from large-signal data now becomes numerically feasible given the greater computational speed associated with an SEM-based method. This can facilitate both small-signal analysis and design and nonlinear large-signal analysis. Given the increasing interest in applying heterostructure-based compound semiconductor devices to new application areas (e.g., optical, chemical, and biological sensors), the SEM-based approach demonstrated herein can permit efficient numerical design of complex device structures having novel material profiles.

REFERENCES


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An Efficient ACA Solution for Electromagnetic Scattering from Discrete Body of Revolution

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Abstract — Discrete body of revolution (DBoR) enhanced method of moments (MoM) is a specialized technique to analyze electromagnetic scattering from the object with discrete cylindrical periodicity. By exploiting the block circulant property of the impedance matrix of MoM, both filling time and storage requirement for the matrix are reduced. The matrix-vector product can be further accelerated by using the fast Fourier transform (FFT) technique. However, the matrix filling time and memory requirement of DBoR-FFT are the same as those of DBoR-MoM, which are still expensive when the number of unknowns in each sector becomes larger. Meanwhile, the DBoR-FFT scheme works inefficiently for the small number of periodic sectors. In this paper, the adaptive cross approximation (ACA) technique is employed to enhance the DBoR-MoM. Numerical examples are given to demonstrate the efficiency of the proposed method.

Index Terms — Adaptive cross approximation, discrete body of revolution, electromagnetic scattering, method of moments.

I. INTRODUCTION

Method of moments (MoM) is a popular tool to analyze the electromagnetic scattering from the conducting objects [1-3]. For objects with general shape, MoM costs $O(N^3)$ CPU time and $O(N^2)$ memory, where $N$ is the number of unknowns, which prohibits its application to electrically large objects.

There are mainly two types of techniques to conquer this difficulty. On one hand, a series of fast algorithms have been proposed for general geometry [2-4]. Among them, adaptive cross approximation (ACA) algorithm [4] is one of the popular techniques. It makes use of the fact that the approximate rank of submatrix is deficient when the source group and the observation group are sufficiently separated. Hence the submatrix can be computed efficiently by invoking low-rank decomposition technique. On the other hand, specialized codes are developed to save the time and memory cost of the ordinary MoM for the structures bearing the symmetry, uniformity or periodicity. Bodies of revolution [5-7], bodies of translation [8], and periodic frequency selective surface [9] are several well-known structures. Recently, discrete body of revolution (DBoR) based integral equation approach [10-14] is proposed to analyze the structures with discrete cylindrical periodicity.

Many structures encountered in practical application possess discrete cylindrical periodicity, such as windmill, turbine and jet-engine. In the original DBoR schemes [10-11], a matrix equation with multiple right-hand sides has to be solved since the decomposition of incident field are required. A direct solution scheme of DBoR is discussed in [13-14], and it usually requires a parallel out-of-core solver for the electrically large geometries whereas the in-core solution is preferred in most situations. An efficiently iterative DBoR solver, which is free of decomposition of incident field, is proposed in [12]. It exploits the block circulant property of the whole impedance matrix, thus the storage requirement and filling time of the matrix are of order $N^3/M$, where $M$ denotes the number of discrete sectors. The time complexity of one matrix-vector product scales $O(N^3)$ if FFT technique is not adopted, and scales $O[(N^2 \log M)/M]$ if fast Fourier transform (FFT) technique is adopted.

However, the efficiency of DBoR-FFT is still required to be improved since FFT works inefficiently when the number of discrete sectors $M$ is small and the storage requirement and filling time are of $O(N^3/M)$, which is still large. In this paper, a DBoR-ACA scheme is developed which exploits ACA to accelerate the solution of DBoR. Numerical experiments demonstrate that DBoR-ACA is an efficient solution scheme.

The remainder of this paper is organized as follows. In Section II, the theory and the formulations are given. Three numerical experiments are presented in Section III to show the efficiency of the proposed method. Section IV concludes this paper.
II. THEORY AND FORMULATIONS

A. DBoR-MoM and DBoR-FFT

As shown in Fig. 1, consider a DBoR geometry comprised of \( M \) discrete cylindrically periodic sectors, each sector occupying an angular width of \( \Delta \phi = 2\pi / M \). In the analysis, the mesh is generated for sector \( S_1 \) and then rotated to obtain the meshes for other sectors \( S_i \), \( i = 2, 3, ..., M \). The meshes must remain conformal on truncated boundary between two neighbor sectors for current continuity and satisfy cylindrically periodical condition to take advantage of DBoR, as discussed in [11]. The surface current density \( \mathbf{J}(r) \) is expanded by the RWG basis functions [1] divided into sectors:

\[
\mathbf{J}(r) = \sum_{i=1}^{M} \sum_{n,s} I_{is} f_{ns}(r),
\]

where \( Q \) denotes the number of unknowns in each sector, and \( N = M \times Q \) is the number of total unknowns. \( I_{is} \) and \( f_{ns} \) represent the corresponding expansion coefficient and the RWG function for \( q \)th basis function in \( m \)th sector. \( r \in S_m \) is position vector. The impedance matrix \( \mathbf{Z} \) of combined field integral equation (CFIE) is correspondingly partitioned into blocks. As shown in Eq. (2), each block is denoted as \( \mathbf{Z}_{mn} \), which represents the interactions between sector \( S_m \) and sector \( S_n \), each with the size of \( Q \times Q \):

\[
\begin{bmatrix}
\mathbf{Z}_{11} & \mathbf{Z}_{12} & \mathbf{Z}_{13} & \cdots & \mathbf{Z}_{1M} \\
\mathbf{Z}_{21} & \mathbf{Z}_{22} & \mathbf{Z}_{23} & \cdots & \mathbf{Z}_{2M} \\
\mathbf{Z}_{31} & \mathbf{Z}_{32} & \mathbf{Z}_{33} & \cdots & \mathbf{Z}_{3M} \\
\vdots & \vdots & \vdots & \ddots & \vdots \\
\mathbf{Z}_{M1} & \mathbf{Z}_{M2} & \mathbf{Z}_{M3} & \cdots & \mathbf{Z}_{MM}
\end{bmatrix}
\begin{bmatrix}
I_1 \\
I_2 \\
I_3 \\
\vdots \\
I_M
\end{bmatrix}
=
\begin{bmatrix}
V_1 \\
V_2 \\
V_3 \\
\vdots \\
V_M
\end{bmatrix}.
\]

Similar to the procedure for electric field integral equation in [12], \( \mathbf{Z}_{mn} \) for CFIE can be represented as \( \mathbf{Z}_{mn,s,t} \) since it depends only on the value of \( m-n \). Also, due to the rotational symmetry, it exists \( \mathbf{Z}_{mn,s,t} = \mathbf{Z}_{m-n, s+t} \). As a result, only \( \mathbf{Z}_{i,0} \), \( i=0, 1, ..., M-1 \), are required to be computed and stored. One can also use \( O^2 \) times FFT of \( M \)-points to compute the matrix-vector product [12] since the impedance matrix is in the block circulant form. In this letter, the scheme without FFT is referred to as DBoR-MoM whereas the scheme with FFT is referred to as DBoR-FFT.

B. DBoR-ACA

DBoR-ACA does not work well when the number of discrete sectors \( M \) is small. A scheme of DBoR-ACA is developed by employing the ACA algorithm to compress each block of \( \mathbf{Z} \). A multilevel spatial partitioning is used to group the RWG functions in each sector. The groups are recorded using octree data at all levels. The touching groups at the finest level are near groups and the others are well-separated groups. The interactions of near groups are computed via DBoR-MoM directly, while the interactions of others are accelerated by the ACA algorithm. Consider two well-separated groups, one group containing \( s \) testing functions residing in sector \( S_m \), and the other group containing \( t \) basis functions residing in sector \( S_n \). The interactions between them lead to a submatrix \( \mathbf{Z}_{mn} \), which is one of the submatrices of block \( \mathbf{Z}_{M+1,n} \). Here, the superscript \( s \times t \) denotes the size of the submatrix. Suppose that \( \mathbf{Z}_{mn} \) is rank-deficient with an effectively approximate rank \( r \). The rank \( r \) is usually far smaller than \( s \) and \( t \) when both \( s \) and \( t \) are large. By utilizing the ACA algorithm, the matrix \( \mathbf{Z}_{mn} \) can be approximated as:

\[
\mathbf{Z}_{mn} \approx \mathbf{U}_{mn}^{rs} \mathbf{V}_{mn}^{rs},
\]

where \( \mathbf{U}_{mn}^{rs} \) and \( \mathbf{V}_{mn}^{rs} \) are two decomposition matrices. The rank \( r \) is determined adaptively by ACA algorithm to satisfy the following condition:

\[
\| \mathbf{Z}_{mn} - \mathbf{U}_{mn}^{rs} \mathbf{V}_{mn}^{rs} \| \leq \tau \| \mathbf{Z}_{mn} \|,
\]

where \( \tau \) denotes the truncated tolerance of the ACA algorithm and is set as \( 10^{-3} \) in this paper. The details of the ACA algorithm to fulfill (3) are referred to [4]. DBoR-ACA fills only a fraction of entries for each block \( \mathbf{Z} \). As shown by numerical observation in [4], both the memory and CPU time requirements of the ACA algorithm scale as \( N^{5/3} \log N \) for electrically moderate size problems, while those of MoM scale \( N^2 \). Thus DBoR-ACA can reduce both the memory requirement and simulation time for DBoR-MoM.

Table 1 lists a comparison of predictions of the computational complexity for the scheme of DBoR-MoM, DBoR-FFT, and DBoR-ACA, where MVP time denotes the time required to compute matrix-vector product once.

![Fig.1. A geometry with M cylindrical periodic sectors.](image-url)
controlled by both the octree structure and the sectors. Increasing number of divided sectors brings two burdens which lessen the compressed efficiency of DBoR-ACA against the case when ACA algorithm is utilized for objects of general shape. The first one is that it produces more groups belonging simultaneously to more than one sector, which resulting in more groups with small number of unknowns. The second burden is that the size of the largest group of DBoR-ACA is reduced since the size of each sector is reduced. But even for as many as 128 sectors, the performance of DBoR-ACA has not been reduced by much.

Figure 2 (b) shows the CPU time cost of one MVP. It can be found that the computational time for DBoR-MoM and DBoR-ACA changes slightly as number of sectors increases, whereas the computational time of DBoR-FFT reduces dramatically. The complexity of DBoR-FFT is consistent with $O(N^2 \log N/M)$ for large value of number of sectors $M$, however the computational time is even greater than that of DBoR-MoM for $M < 32$. It is because the implementation of FFT with small number of points is not that efficient. In addition, it destroys the CPU cache friendly feature of the submatrices of original DBoR-MoM. It can be observed that a slightly decrease of the CPU time for DBoR-MoM for large number of sectors. This owes to a slightly decreasing of the total number of unknowns as given in Table 2. The slightly increase of the CPU time for DBoR-ACA with large number of sectors is ascribed to the same burdens for the memory requirement. It should be noted that the DBoR-ACA scheme takes more CPU time to perform one MVP than the DBoR-FFT scheme does for large $M$. For DBoR configuration of practical engineering $M$ is usually not very big, hence, DBoR-ACA is still a faster solver by considering the filling time of the impedance matrix together. Table 3 shows the case for 128 sectors. Here the total time denotes the whole analysis time including the time for pre-processing, matrix filling and solving, and RCS calculating. It can be observed that FFT takes effect in accelerating the DBoR-MoM. However, the total simulation time of DBoR-ACA is still less than that of DBoR-FFT due to saving of matrix filling time.

### III. NUMERICAL RESULTS

To demonstrate the efficiency of the DBoR-ACA scheme in comparison with DBoR-MoM and DBoR-FFT, codes are developed for all schemes and numerical examples are presented for typical geometries. In the simulations, the frequency $f$ is 300 MHz unless otherwise specified. The electric field of incident wave is $E^i = \hat{E} \exp(j2\pi\ell/\lambda)$, where $\lambda$ is the wavelength. The discrete rotational axis of DBoR is $z$ axis. CFIE is employed with combination coefficient of 0.5. The resulting matrix equations are iteratively solved by restarted GMRES [16] where the restarted number is set to be 30. The stop criterion for iteration is relative residual norm less than $10^{-5}$. The bistatic radar cross section (RCS) results are observed at the plane with fixed azimuthal angle $\varphi' = 0^\circ$ and varied polar angles $\theta'$. All the simulations are carried out in single precision arithmetic on a computer equipped with a 2.83 GHz Intel® Core2 Quad processor, with one core being used.

#### A. Efficiency test for a conducting ring

The first example is selected to test the performance of different schemes varying with number of sectors. The configuration is a conducting ring with inner radius $a=2$ m, outer radius $b=3$ m, and height $h=0.1$ m, as shown in Fig. 2 (a). To take advantage of DBoR, the mesh has to be changed each time when the number of sectors is increased. Here, the total number of unknowns is kept at a fixed level approximately. Table 2 lists the number of unknowns corresponding to each number of sectors.

<table>
<thead>
<tr>
<th>$M$</th>
<th>8</th>
<th>16</th>
<th>32</th>
<th>64</th>
<th>80</th>
<th>100</th>
<th>128</th>
</tr>
</thead>
<tbody>
<tr>
<td>$N$</td>
<td>47304</td>
<td>46560</td>
<td>46752</td>
<td>46848</td>
<td>44880</td>
<td>45600</td>
<td>45312</td>
</tr>
</tbody>
</table>

Figure 2 (a) plots the memory requirement for DBoR-MoM and DBoR-ACA. The memory requirement of DBoR-FFT is same as that of DBoR-MoM. It can be observed that the memory requirement of DBoR-MoM is scaled as $1/M$, while the DBoR-ACA grows a few larger than $O(1/M)$ as $M$ increases. The reason is that the dividing strategy of the DBoR-ACA group is

Table 1: Predictions of computational complexity

<table>
<thead>
<tr>
<th></th>
<th>DBoR-MoM</th>
<th>DBoR-FFT</th>
<th>DBoR-ACA</th>
</tr>
</thead>
<tbody>
<tr>
<td>Matrix-filling time</td>
<td>$O(N^2/M)$</td>
<td>$O(N^2/M)$</td>
<td>$O(N^{4/3}\log N/M)$</td>
</tr>
<tr>
<td>Storage of matrix</td>
<td>$O(N^2/M)$</td>
<td>$O(N^2/M)$</td>
<td>$O(N^{4/3}\log N/M)$</td>
</tr>
<tr>
<td>MVP time</td>
<td>$O(N^2)$</td>
<td>$O(N^2\log N/M)$</td>
<td>$O(N^{4/3}\log N/M)$</td>
</tr>
</tbody>
</table>
Fig. 2 Comparisons of the complexity for DBoR-MoM, DBoR-FFT, and DBoR-ACA schemes by increasing the number of divided sectors while keep approximately the total number of unknowns of 46000. (a) Memory, and (b) CPU time for one matrix-vector multiplication.

Table 3: Comparisons of the CPU time and memory requirement for the conducting ring with 128 cylindrical sectors and each with 354 unknowns.

<table>
<thead>
<tr>
<th></th>
<th>DBoR-MoM</th>
<th>DBoR-FFT</th>
<th>DBoR-ACA</th>
</tr>
</thead>
<tbody>
<tr>
<td>Memory (MB)</td>
<td>125</td>
<td>125</td>
<td>26</td>
</tr>
<tr>
<td>Matrix filling time (s)</td>
<td>72</td>
<td>73</td>
<td>13</td>
</tr>
<tr>
<td>Number of iteration</td>
<td>39</td>
<td>39</td>
<td>39</td>
</tr>
<tr>
<td>Iteration time (s)</td>
<td>105</td>
<td>19</td>
<td>42</td>
</tr>
<tr>
<td>Total time (s)</td>
<td>180</td>
<td>95</td>
<td>58</td>
</tr>
</tbody>
</table>

B. Computation complexity test for a conducting ring

The second example is a conducting ring with inner radius $a = 2$ m, outer radius $b = 4$ m, and height $h = 0.1$ m. This example is to show the complexity of various DBoR schemes for electrically increasing large problems. The ring is discretized with a mesh size of $h = 0.05$ m and the frequency $f$ is increased from 214.3 MHz to 333.3 MHz. This leads to an increase of the total number of unknowns from 54528 to 129600 as the relation $N \propto f^2$. The ring is modeling with 64 sectors. The size of group box at finest level is 0.2 m. The memory requirement and CPU time of one MVP for DBoR-MoM and DBoR-ACA are illustrated in Fig. 3 (a) and Fig. 3 (b), respectively. The memory requirement of DBoR-FFT is same as that of DBoR-MoM. It can be observed the practical implementation is consistent with the prediction of the complexity as listed in Table 1. Also both the memory requirement and CPU time cost of DBoR-ACA are less than those of DBoR-FFT when the number of unknowns becomes large in each sector.

C. Bistatic RCS for a conducting jet-engine inlet

The last example is a conducting jet-engine inlet as shown in Fig. 4 (a). The configuration has 16 sectors and each sector has 4932 unknowns. The size of group box at finest level of DBoR-ACA algorithm is 0.4 m. This example is to show the efficiency for the small number of sectors of various DBoR schemes. The geometry and dimension of one sector of jet-engine is shown in Fig. 4 (b) and of shell is shown in Fig. 4 (c). For the cylindrical shell, the radius is 2.1 m and the height is 4.0 m for the inner side and the thickness is 0.1 m. The jet-engine is placed at a distance of 0.1 m above the bottom of the shell. The bistatic RCS results are illustrated in Fig. 5 for the various DBoR schemes and fast multipole solver in FEKO®. It can be observed that they are in agreement with each other. Table 4 shows the CPU time and memory requirement for this example. It can be found that DBoR-FFT fails to accelerate DBoR-MoM while DBoR-ACA successes to spend less memory and less CPU time.
Table 4: Comparison of the CPU time and memory requirement for the jet-engine inlet with 16 sectors and each sector with 4932 unknowns

<table>
<thead>
<tr>
<th></th>
<th>DBoR-MoM</th>
<th>DBoR_FFT</th>
<th>DBoR_ACA</th>
</tr>
</thead>
<tbody>
<tr>
<td>Memory (MB)</td>
<td>3008</td>
<td>3008</td>
<td>396</td>
</tr>
<tr>
<td>Matrix filling time (s)</td>
<td>1530</td>
<td>1532</td>
<td>244</td>
</tr>
<tr>
<td>Number of iteration</td>
<td>344</td>
<td>344</td>
<td>348</td>
</tr>
<tr>
<td>Iteration time (s)</td>
<td>4953</td>
<td>9013</td>
<td>1072</td>
</tr>
<tr>
<td>Total time (s)</td>
<td>6490</td>
<td>10552</td>
<td>1321</td>
</tr>
</tbody>
</table>

IV. CONCLUSION

The DBoR-MoM has been extended to CFIE for the analysis of electromagnetic scattering from discrete body of revolution in free space. The ACA technique was exploited to accelerate both matrix-filling operation and matrix-vector product of the DBoR-MoM. Numerical examples validate the efficiency and accuracy of the proposed method in comparison with DBoR-MoM and DBoR-FFT. The numerical results suggest that DBoR-FFT fails to accelerate DBoR-MoM for the DBoR with small number of sectors whereas the proposed DBoR-ACA method is appropriate for accelerating the solution of all types of cylindrically periodic geometries. At the end, it is worthwhile to note that a faster scheme can be obtained if sparsified ACA [17] is applied into the DBoR.

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Electromagnetic Scattering from a PEMC Circular Cylinder Coated by Topological Insulator (TI)

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Abstract — Scattering of electromagnetic plane wave from a perfect electromagnetic conducting (PEMC) cylinder coated with a topological insulating (TI) material has been presented. The core (PEMC) and cladding material (TI) produce co- and cross-polarized components of electromagnetic field in response to the incident plane wave for a given polarization (TE or TM). When the value of $\theta$ is made zero, TI material becomes ordinary dielectric and the results of PEMC coated with TI with $\theta = 0$ (dielectric material) have been compared with the previously published literature and are found in good agreement. When the coating is removed, same results as that of isolated PEMC circular cylinder have also been reproduced.

Index Terms — Cladding, insulating, isolated, polarization, topological.

I. INTRODUCTION

Topological insulator’s states in 2D and 3D materials were observed theoretically in 2005 and 2007 while their experimental discovery won the 2010 Nobel Prize. Topological insulator material is currently the hottest topic in condensed matter and quantum physics, and is hard to understand. Topological insulator is a type of material that conducts electricity on its surface due to special surface electronic states. The surface states of TI are topologically protected, i.e., they cannot be destroyed by impurities. TIs are made possible due to the combination of time-reversal symmetry and interaction of spin-orbit coupling, which occurs in heavy elements like mercury and bismuth.

TIs are defined by the constitutive relations:

$$
D = \varepsilon_r \varepsilon_0 E - \varepsilon_0 \alpha \frac{\theta}{\pi} (c_0 B),
$$

(1)

$$
c_0 H = \frac{c_0}{\mu_0 \mu_r} B + \alpha \frac{1}{\pi \mu_0 \mu_r} B,
$$

(2)

where $\varepsilon_r$, $\mu_r$ are relative permeability and permittivity and $c_0$, $\mu_0$ are the permeability and permittivity of free space respectively. $c_0$ is the speed of light in vacuum, $\alpha = e^2/4\pi \varepsilon_0 \hbar c_0$ is the fine structure constant, $h$ is Plank constant, $e$ is the electric charge and $\theta$ is axion parameter which is uniquely determined by band structure. Only two values of $\theta$ are possible: $\theta = 0$ (i.e., conventional dielectric) and $\theta = \pi$ (i.e., topological insulator), which gives time reversal symmetry. When time reversal symmetry is broken, the $\theta$ is quantized in odd integer values of $\pi$ [1].

Due to these attractive properties many scientists started studying TI. Surface plasmons localized on the topologically nontrivial interface have been studied by Karch [1]. Qi and Zhang studied the theory of topological superconductors in close analogy to the theory of topological insulators [2]. Scattering results from topological insulator cylinders are very few [3-6]. Scattering by TI circular cylinder is discussed in [3]. Scattering from buried TI circular cylinder in a slightly rough surface has been investigated in [4] and buried in a semi infinite medium has been discussed in [5]. In [6], it has been shown that what will happen when a TI circular cylinder is placed in chiral medium? Electromagnetic scattering from coated cylinder is a more challenging task and is therefore addressed in the present paper.

PEMC is a new class of materials introduced by Lindell and Sihvola [7]. It is the generalization of perfect electric conductor (PEC) and perfect magnetic conductor (PMC) material. It is defined by the boundary conditions:

$$
\mathbf{n} \times (\mathbf{H} + \alpha \mathbf{E}) = 0,
$$

(3)

$$
\mathbf{n} \cdot (\mathbf{D} - MB) = 0,
$$

(4)

where $\mathbf{M}$ denotes the admittance of the PEMC boundary. $M = 0$ for PMC and $M \rightarrow \pm \infty$ for PEC. The circular cylinders are the most basic canonical shape for the study of electromagnetic waves scattering. Cylindrical geometry has a long history in EM problems [8-9, 12-18, 22-26]. Scientists had studied the problems of circular cylinder using composition of different materials, e.g., dielectric, negative refractive index materials (NRM),

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PEMC, chiral, and nilothil [10-14, 16, 17, 19-24].

In this paper, an infinite PEMC circular cylinder coated with TI material is considered. The purpose of the study is to explore important scattering characteristics and to provide physical insight of this geometry. Using the large argument approximation of Hankel function, the bi-static echo widths in the far zone are calculated. For the verification of analytical formulation and numerical code, numerical results are compared with the published work. We have used $e^{-i\omega t}$ time dependence which is suppressed throughout the analysis.

In the next few sections, analytical formulation, numerical results and discussion, and conclusions are described.

II. ANALYTICAL FORMULATION

Geometry of problem is shown in Fig. 1. Inner cylinder is PEMC while the outer cylinder shows the coating layer of TI material. PEMC circular cylinder coated with TI material is of uniform thickness. The cylinder is supposed to be of infinite length along z-axis. Radius of PEMC cylinder is 'a' while radius of PEMC cylinder coated with TI is 'b'. The region outside the coating $\rho > b$ is free space and is mentioned as region 0 with wave number $k_0 = \omega \sqrt{\mu_0 \epsilon_0}$. The region between $a < \rho < b$ is termed as region 1 with wave number as $k_1 = \omega \sqrt{\mu_1 \epsilon_1}$.

Fig. 1. PEMC Cylinder coated with topological insulator.

The polarization of the incident electric field is taken parallel to the axis of the cylinder. The incident electric field is given by:

$$E_{0x}^i = e^{ik_0 \rho \cos \phi}.$$  

The incident electric field can be written in terms of cylindrical wave function as:

$$E_{0x}^i = \sum_{n=-\infty}^{\infty} i^n j_n (k_0 \rho) e^{in\phi}. \tag{6}$$

Using Maxwell’s equations, the corresponding magnetic field in $\phi$ direction can be written as:

$$H_{1\phi}^{\phi} = \frac{E_0}{\eta_0} \sum_{n=-\infty}^{\infty} j_n (k_0 \rho) e^{in\phi}. \tag{7}$$

The scattered co-polarized electric field in region 0 can be written as:

$$E_{0x}^s = \sum_{n=-\infty}^{\infty} i^n a^n H_n^{(1)} (k_0 \rho) e^{in\phi}. \tag{8}$$

And the corresponding $\phi$ component of scattered co-polarized magnetic field can be written as:

$$H_{0\phi}^s = -\frac{E_0}{\eta_0} \sum_{n=-\infty}^{\infty} i^n a^n H_n^{(1)} (k_0 \rho) e^{in\phi}. \tag{9}$$

As the core material of the cylinder is of PEMC, so in addition to co-polarized component cross-polarized component will also appear. The scattered cross-polarized magnetic field in region 0 can be written as:

$$H_{0\phi}^c = -\frac{E_0}{\eta_0} \sum_{n=-\infty}^{\infty} i^n b^n H_n^{(1)} (k_0 \rho) e^{in\phi}. \tag{10}$$

And the corresponding $\phi$ component of scattered cross-polarized electric field can be expressed as:

$$E_{0\phi}^c = -E_0 \sum_{n=-\infty}^{\infty} i^n a^n H_n^{(1)} (k_0 \rho) e^{in\phi}. \tag{11}$$

Region 1 has two interfaces at $\rho = a$ and $\rho = b$; therefore, co- and cross-polarized electric and magnetic fields in region 1 can be expressed in terms of oppositely traveling cylindrical waves as:

$$E_{1x} = E_0 \sum_{n=-\infty}^{\infty} i^n [e^n H_n^{(2)} (k_1 \rho) + d^n H_n^{(1)} (k_1 \rho)] e^{in\phi}, \tag{12}$$

$$H_{1\phi} = -\frac{E_0}{i\eta_1} \sum_{n=-\infty}^{\infty} i^n [e^n H_n^{(2)} (k_1 \rho) + d^n H_n^{(1)} (k_1 \rho)] e^{in\phi}, \tag{13}$$

$$E_{1x} = -\frac{iE_0}{\eta_1} \sum_{n=-\infty}^{\infty} i^n [e^n H_n^{(2)} (k_1 \rho) + f^n H_n^{(1)} (k_1 \rho)] e^{in\phi}, \tag{14}$$

$$H_{1\phi} = -E_0 \sum_{n=-\infty}^{\infty} i^n [e^n H_n^{(2)} (k_1 \rho) + f^n H_n^{(1)} (k_1 \rho)] e^{in\phi}. \tag{15}$$

In above expressions $j_n(.)$ is the Bessel functions of first kind, while $H_n^{(1)}(.)$ and $H_n^{(2)}(.)$ are the Hankel functions of first and second kinds respectively. Also $a_n, b_n, c_n, d_n, e_n$ and $f_n$ are the unknown scattering coefficients. These unknown coefficients can be found by using boundary conditions at the interfaces $\rho = a$ and $\rho = b$.

At $\rho = a$, boundary conditions are:

$$H_{1z} + ME_{1z} = 0 \quad \rho = a, \quad 0 \leq \phi \leq 2\pi, \tag{16}$$

$$H_{1\phi} + ME_{1\phi} = 0 \quad \rho = a, \quad 0 \leq \phi \leq 2\pi. \tag{17}$$

At $\rho = b$, boundary conditions are:

$$H_{0\phi}^s + H_{0\phi}^c = H_{1\phi} - \frac{\theta}{\epsilon_0} E_{1\phi} \quad \rho = b, \quad 0 \leq \phi \leq 2\pi, \tag{18}$$

$$H_{0z} = H_{1z} - \frac{\theta}{\epsilon_0} E_{1z} \quad \rho = b, \quad 0 \leq \phi \leq 2\pi, \tag{19}$$

$$H_{0\phi}^s = H_{1\phi} \quad \rho = b, \quad 0 \leq \phi \leq 2\pi, \tag{20}$$

$$E_{0\phi}^c = H_{1\phi} \quad \rho = b, \quad 0 \leq \phi \leq 2\pi. \tag{21}$$

where

$$E_{0x} = E_{0x}^i + E_{0x}^s,$$  

$$H_{0\phi} = H_{0\phi}^c + H_{0\phi}^s.$$

By the application of above boundary conditions at interface $\rho = a$ and $\rho = b$, a linear matrix is obtained in terms of the unknown scattering coefficients. Solution of
this matrix gives unknown scattering coefficients. The values of \( a_n \) and \( b_n \) give us co- and cross-polarized components of scattered field due to PEMC cylinder coated with TI.

### III. BACK SCATTERING CROSS-SECTIONS \((\sigma)\)

The ratio of the total power scattered by the scatter to the incident power per unit area on the scatterer is called back scattering cross-section and is given as:

\[
\sigma = 2\pi\rho \frac{W^s}{W_I} = 2\pi\rho \frac{|E^s|^2}{|E^I|^2}.
\]

For parallel polarization, the normalized bi-static echo width (RCS) of the co-polarized and cross-polarized field components is given by:

\[
\frac{\sigma_{co}}{\lambda_0} = \frac{2}{\pi} \sum_{n=-\infty}^{\infty} a_n e^{in(\theta)} \frac{\rho}{\rho},
\]

\[
\frac{\sigma_{cross}}{\lambda_0} = \frac{4}{\pi} \sum_{n=-\infty}^{\infty} b_n e^{in(\theta)} \frac{\rho}{\rho},
\]

where \( a_n \) is the scattering coefficient of co-polarized field and \( b_n \) is the scattering coefficient of cross-polarized field. For perpendicular polarization, the duality principle for the above formulation may be used.

### IV. NUMERICAL RESULTS AND DISCUSSION

In this section numerical results are described. The numerical results are based on the above analytical formulations for PEMC coated with TI material. For Figs. 2-8, \( k_0a = 1.05, k_0b = 2.1, e = 1.6*10^{-19} \)C, speed of light is \( c_0 = 3*10^8 m/s \) and permeability \( \mu = 1 \).

In Fig. 2 echo width of PEMC circular cylinder coated with TI material has been plotted with observation angle from 0 to \( 2\pi \) radians. In this figure, \( M\eta_1 = 0, \epsilon = 9.8 \) and \( \theta = 0 \), which is a case for PMC circular cylinder coated with dielectric material. This result when compared with [24], excellent agreement is found.

Fig. 2. Echo width from PMC coated with dielectric material with \( \epsilon = 9.8 \).

In Figs. 3 and 4, the numerical result is repeated for \( M\eta_1 = \infty \), \( \epsilon = 9.8 \) and \( \theta = 0 \) and \( M\eta_1 = 1, \epsilon = 9.8 \) and \( \theta = 0 \). Comparison is made for PEC circular cylinder coated with dielectric material. Again, the obtained nice comparison and validated our formulation. With this confidence, the numerical code has been run for different values of \( M\eta_1 \) and \( \theta \), while \( \epsilon = 100 \) which is the characteristic value of TI material (i.e., \( \epsilon = 50 \) to 100) and Figs. 5-8 are obtained.

In Figs. 5-8, scattering behavior of PEMC circular cylinder coated with TI material has been highlighted. With these results one can understand the composition of the highly focused material, i.e., TI material with PEMC (which is the most fundamental material for electromagnetic analysis).

Figure 5 represents echo width of co- and cross-polarized components for PMC circular cylinder coated with TI material. In this figure, \( M\eta_1 = 0, \epsilon = 100 \) and \( \theta = 41\pi \), which is the case of PMC circular cylinder coated with TI material. In this plot, cross-polarized component has also been appeared along with co-polarized component. This cross-polarized component is due to TI material. Figure 6 shows the case when PEMC
circular cylinder coated with TI material has been taken. For this figure, parameters are taken as $M\eta_1 = 1$, $\epsilon = 100$ and $\theta = 41\pi$. On comparing Fig. 5 and Fig. 6, it has been observed co-polarized component is same, while cross-polarized component of Fig. 5 is greater than the cross-polarized component of Fig. 6. This greater contribution in the cross-polarized component is because of PEMC core when coated with TI material.

![Fig. 5. Echo width of co- and cross-polarized components for PMC with $\epsilon = 100$.](image1)

![Fig. 6. Echo width of co- and cross-polarized components for PEMC with $\epsilon = 100$.](image2)

Figure 7 shows the comparison between co-polarized components for different of $\theta$, i.e., for $\theta = 0$ and $\theta = 41\pi$; when PEMC circular cylinder coated with TI material is considered. It is observed that the behavior of co-polarized component for $\theta = 0$ is same as co-polarized component for $\theta = 41\pi$ near 0-80 and 310-360 but different for 80-310 regardless of the amplitude.

![Fig. 7. Echo width of co- and co-polarized components for different values of $\theta$ when core is PEMC with $\epsilon = 100$.](image3)

![Fig. 8. Echo width of cross- and cross-polarized components for different values of $\theta$ when core is PEMC with $\epsilon = 100$.](image4)

Figure 8 shows the comparison between cross-polarized components for $\theta = 0$ and $\theta = 41\pi$ respectively, when PEMC circular cylinder coated with TI material is considered. It is observed that the behavior of cross-polarized component for $\theta = 0$ is different from co-polarized component for $\theta = 41\pi$ near 0-80 and 310-360 but same for 80-310 regardless of the amplitude.

V. NUMERICAL VERIFICATION

In Fig. 2 the code has been verified with the limiting parameters $M\eta_1 = 0$, $\epsilon = 9.8$ and $\theta = 0$ and compared with the literature [24]. The code is further verified in Figs. 3 and 4, the parameters used are: $M\eta_1 \rightarrow \infty$, $\epsilon = 9.8$ and $\theta = 0$ and $M\eta_1 = 1$, $\epsilon = 9.8$ and $\theta = 0$. Comparison is made with [24]. In both the cases excellent agreement is found. The proposed study can also be verified with the help of experiments as well as commercially available simulation software which will be our task.

VI. CONCLUSIONS

In the present paper analytical formulation of a perfect electromagnetic conducting (PEMC) circular cylinder of infinite length coated with a topological insulating (TI) material has been presented. The core (PEMC) and cladding (TI) material produces co- and cross-polarized components of electromagnetic magnetic field in response to the incident plane wave for a given polarization (TE or TM). By coating TI material on PEMC circular cylinder, again co- and cross-polarized components of the scattered field has been obtained. When the value of $\theta$ is made zero, TI material becomes...
ordinary dielectric and the results of PEMC coated with dielectric material have been reproduced. When the coating is removed, the same results, as that of isolated PEMC circular cylinder, have been reproduced. Making \( M_{\eta_1} \rightarrow \infty \) or 0, PEC and PMC coated with TI material results. By using both aforementioned conditions, results of PEC and PMC coated with ordinary dielectric material have been obtained. Thus, in short, it can be said that the present problem is the most fundamental and generalized which contains all the special cases, i.e., PEC coated with dielectric, PMC coated with dielectric, PEMC coated with dielectric, PEC coated with TI, PMC coated with TI, PEMC coated with TI material can be conveniently achieved. It can also be concluded that the behavior of co- and cross-polarized components vary both in amplitude and shape with variation in geometrical parameters, i.e., \( M_{\eta_1}, \theta \) and \( \epsilon \).

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Nonlinear Analysis and Performance Improvement of Amplifying Aperture Coupled Reflectarray Antenna

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Abstract — Amplifying reflectarray antenna can be used to increase the overall gain of the antenna in large distance communication systems. However, amplifying reflectarray antenna may become nonlinear in some incident powers which may lead to performance degradation. In this paper nonlinear behavior of an amplifying reflectarray antenna is studied and a new method is proposed to improve the performance of the antenna. Nonlinear analysis of the active unit cell is performed using harmonic balance method considering nonlinear model of the amplifier. Then, the effect of nonlinear element in radiation pattern of the antenna is studied. Aperture coupled patch structure is used to analyze amplifying unit cell. Finally, an amplifying reflectarray antenna considering nonlinear behavior of the active elements is designed and the proposed balanced amplifier structure is used to improve performance of the amplifying antenna.

Index Terms — Active antenna, antenna array, harmonic balance method, nonlinear analysis, reflectarray antenna.

I. INTRODUCTION

Printed reflectarray has some advantages compared to the usual reflectors, four of which - i.e., saving volume, simplifying the mechanical design, applying easily to deployable reflectors, and capability of integrating active elements by the antenna structure are of great importance. Different unit cell shapes are proposed to improve the reflectarray antenna performance which introduce the required phase-shift on the reflected field to produce a focused or shaped beam. Required phase shift can be obtained using resonating patches [1] or by a transmission line of proper length connected [2] or aperture-coupled to the patches [3, 4] with different size or using active elements like PIN diodes [5] or varactor diodes [6].

Using high gain antenna for large distance communication is necessary to improve performance of communication link. In these cases, usually reflector antenna or phased array antenna is used. However, manufacturing reflector antenna is difficult especially in high frequencies and phased array antenna may have some problems like unwanted radiation from the feed network. Amplifying reflectarray antenna is proposed in [3, 4, 7] which uses amplifier in each unit cell to increase gain of the antenna. Using an amplifier in the antenna structure results in difficulties in the antenna design and some issues should be determined like stability of the antenna. Moreover, the active element acts nonlinear and this necessitates the nonlinear analysis of the antenna structure. So, [8] studies nonlinear analysis of amplifying reflectarray antenna and [6] studies nonlinear analysis of reflectarray antenna containing varactor diodes.

This paper shows the importance of nonlinear analysis of active reflectarray antenna, and also the influence of nonlinear element in radiation pattern is clarified. The main output of this work is that by the explained method, the impact of nonlinearities on the performance of reflectarrays can be investigated. Furthermore, any active reflectarray cell having active device by any nonlinear model can be used in the analysis and the impact of the model parameters can be studied. Also, a new structure using balanced amplifier is used to improve antenna performance. This performance improvement will be cleared by designing a sample antenna with and without using the proposed cell.

To analyze the active reflectarray antenna, first the cell removing the amplifier is simulated using HFSS software considering infinite array approach to obtain the passive unit cell scattering parameters in which the amplifier is replaced by a two port network and two spatial ports modelled as Floqute port are assumed representing two orthogonal polarizations. So, a 4 port network is obtained which its scattering parameters are known. In the next step, an amplifier which has nonlinear model is connected to the 4 port network and the active cell performance is studied to obtain the cell amplitude response by varying the incident power to the cell. Finally, obtained nonlinear response of the cell is used to design the antenna. Verification of the nonlinear response of the unit cell is done by ADS simulation.
This process is carried out for a sample active unit cell in the center frequency of 6.2 GHz and detailed steps are explained.

**II. UNIT CELL MODELING**

In aperture coupled microstrip antenna structure, each cell consists of a microstrip line coupled to the radiating patch on the opposite side of the substrate via an aperture in the ground plane as shown in Fig. 1. In this paper cross-polarized element configuration is used like [3, 4] to prevent instability, where the incident and scattered fields are orthogonally-polarized. Unit cell consists of a dual-polarized aperture coupled microstrip patch and an amplifier connected between the two ports in the microstrip line path. Also, as in this paper our goal is to evaluate the performance of the active element, an ideal phase shifter is used to control the phase of the reflected signal. Parameters of the unit cell are given in Table 1. Dielectric constant of top substrate is 3.02 with a height of 1.524 mm, and dielectric constant of bottom substrate is 6.15 with a height of 1.28 mm.

The passive part of the cell is modelled as a 4 port network in which ports 3 and 4 are spatial ports modelled as Floqute port [9], and the amplifier is connected between ports 1 and 2. So, the passive part of the unit cell removing the amplifier is simulated using HFSS software supposing infinite array to obtain 4 port scattering parameters as shown in Fig. 2.

Next, active element is connected between ports 1 and 2 of the obtained 4 port network as shown in Fig. 2 to obtain the active cell response in linear or nonlinear states. Active element used in this work is NE4210 which has nonlinear TOM model [10] and can be simulated in ADS software. Nonlinear TOM model is shown in Fig. 3 which has two nonlinear capacitances of $C_{gs}$ and $C_{ps}$, and one nonlinear current source of $I_{ds}$. Relations for current source of $I_{ds}$, and capacitances of $C_{gs}$ and $C_{ps}$ of TOM model are given in (1) to (5) and parameters of the nonlinear model of NE4210 are given in Table 2.

![Fig. 1. Antenna unit cell schematic.](image)

![Fig. 2. Four port modelling of the unit cell.](image)

![Fig. 3. Nonlinear TOM model.](image)

Table 1: Unit cell parameters

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>d</td>
<td>33.56 mm</td>
</tr>
<tr>
<td>p1</td>
<td>14.3 mm</td>
</tr>
<tr>
<td>P2</td>
<td>13.9 mm</td>
</tr>
<tr>
<td>S1</td>
<td>6.3 mm</td>
</tr>
<tr>
<td>S2</td>
<td>8.8 mm</td>
</tr>
<tr>
<td>W1</td>
<td>0.75 mm</td>
</tr>
<tr>
<td>t1</td>
<td>6 mm</td>
</tr>
<tr>
<td>t2</td>
<td>2.5 mm</td>
</tr>
<tr>
<td>W2</td>
<td>1.87 mm</td>
</tr>
<tr>
<td>X1</td>
<td>1.5 mm</td>
</tr>
<tr>
<td>X2</td>
<td>5.7 mm</td>
</tr>
</tbody>
</table>

Table 2: Parameters of the nonlinear model

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>$V_{gs}$</td>
<td>-0.798 (V)</td>
</tr>
<tr>
<td>$C_{gs0}$</td>
<td>0.36 pF</td>
</tr>
<tr>
<td>$C_{gs}$</td>
<td>0.014 pF</td>
</tr>
<tr>
<td>$\beta$</td>
<td>0.0952 (A/V$^0$)</td>
</tr>
<tr>
<td>$\Delta_1$</td>
<td>0.3 (V)</td>
</tr>
<tr>
<td>$T_{g}h$</td>
<td>0.5 (1/W)</td>
</tr>
<tr>
<td>$\Delta_2$</td>
<td>0.6 (V)</td>
</tr>
<tr>
<td>$T_{g}t$</td>
<td>0.065</td>
</tr>
<tr>
<td>$R_g$</td>
<td>8 Ohm</td>
</tr>
<tr>
<td>$Q$</td>
<td>2.5</td>
</tr>
<tr>
<td>$R_d$</td>
<td>0.5 Ohm</td>
</tr>
<tr>
<td>$C_{ds}$</td>
<td>0.12 pF</td>
</tr>
<tr>
<td>$R_s$</td>
<td>3 Ohm</td>
</tr>
<tr>
<td>$R_{db}$</td>
<td>5 KOhm</td>
</tr>
<tr>
<td>$C_{bs}$</td>
<td>1 nF</td>
</tr>
<tr>
<td>$V_{b1}$</td>
<td>0.6 (V)</td>
</tr>
</tbody>
</table>

Current source of $I_{ds}$ in TOM model is given as:

$$I_{ds} = \frac{I_{ds0}}{1+T_{g}h \times V_{ds} \times I_{ds0}}$$  \hspace{1cm} (1)

where
The nonlinear behavior of the cell.

IV. PERFORMANCE IMPROVEMENT OF THE CELL

In this part a new configuration using balanced amplifier is introduced to improve stability of the designed amplifying reflectarray antenna as shown in Fig. 5. Also, this configuration increases the power compression point of the cell which improves the antenna gain in nonlinear states. Using balanced amplifier in amplifying reflectarray antenna has some advantages. First, if the amplifiers are identical, the VSWR from the balanced structure is near 1 and it improves stability of the structure which is a problem in active reflectarray antenna. Moreover, output power is twice that achieved from the single amplifier, and if one of the amplifiers fails, the balanced amplifier unit will still work with reduced gain. Another important advantage of using balanced amplifier in amplifying reflectarray antenna is the ability of easily cascading active unit with other units, like phase shifter, since each unit is isolated by the coupler.

III. HARMONIC BALANCE ANALYSIS

Using harmonic balance method [11], the nonlinear analysis of the unit cell is performed by dividing the unit cell into two nonlinear and linear networks as shown in Fig. 4. Nonlinear part consists of nonlinear capacitances and nonlinear current source. Also, linear part consists of 4-port network of the unit cell and passive elements. Voltages of \( V_1, V_2 \) and \( V_3 \) of Fig. 4 should be evaluated so that \( I_i + \dot{I}_i = 0 \). Harmonic balance equation can be solved with different methods, among which the Newton-Raphson [11] technique is the most common technique and is used in this paper. Assuming nonlinear TOM model for the transistor, total directivity of the unit cell will be obtained in Section V for different incident power which will show the nonlinear behavior of the cell.
measurement gives the distance from the center of the Smith chart to the nearest output (load) stability circle. This stability factor is given by:

\[
M = \frac{1 - |S_{11}|^2}{|S_{22} - \text{conj}(S_{11})^*\Delta + |S_{12}|^*S_{21}|},
\]

(10)

where \( \Delta \) is determinant of the S-parameter matrix. Having \( M > 1 \) is the single necessary and sufficient condition for unconditional stability of a 2-port network. M factor for one stage amplifier is given in Fig. 6, which shows that the active part may become unstable in the frequency of operation.

\[
(10)
\]

Fig. 6. M factor for one stage amplifier and balanced amplifier.

M factor for balanced amplifier is shown in Fig. 6 which has minimum of 1.01 and shows improvement in stability of the active part. Also, output power of the cell versus incident power for first design and improved cell is shown in Fig. 7 and gain of the cell versus incident power for first design and improved cell is shown in Fig. 8.

\[
(11)
\]

VI. SAMPLE ANTENNA DESIGN

In this section it is shown that the power received in each cell is different and as a result each cell may have different gain and phase. Phase of the received field from the feed antenna at each cell is shown in Fig. 9.

Knowing the magnitude of y polarization electric field in each cell \( \hat{E}_y^{R}(u,v) \), power delivered to each cell is obtained as:

\[
P(x,y) = \frac{\left(\hat{E}_y^{R}(u,v)\right)^2}{2\epsilon_0\eta_0}.
\]

(11)

To simulate the active antenna, a 59.2 cm * 59.2 cm antenna is designed in the center frequency of 6.2 GHz by focal length of 74 cm. If we assume center of the antenna as center of Cartesian coordinates, feed antenna is placed in (-29.6 cm, 0, 74 cm). Assuming transmitted feed antenna power in a way that the power distribution on the antenna surface is like Fig. 10, most active elements become nonlinear, and the impact of nonlinearity of each cell should be considered. For this reason amplitude and phase differences caused by the
nonlinear amplifier in each cell should be considered.

Fig. 10. Supposed power on the antenna surface.

Considering amplitude behavior of the unit cell shown in Fig. 8, and the power distribution on antenna surface shown in Fig. 10, amplitude error of each unit cell is calculated as shown in Fig. 11.

Fig. 11. Amplitude error of each unit cell.

The errors in the amplitude and phase of the reflected signal from each cell can cause gain reduction. However, phase error is low in comparison to amplitude error and can be neglected in this scenario. Antenna directivity with and without considering the nonlinear effect of active elements is shown in Fig. 12, when feed power is so that some cells are in nonlinear region and maximum difference between linear and nonlinear analysis can be obtained. In this case, nonlinear analysis shows degradation in gain which cannot be assessed by linear analysis. So, because of the nonlinear behavior of the active elements, maximum gain of the designed antenna decreases from 41 dBi to 37.5 dBi as shown in Fig. 12. Therefore, to assess the pattern of the active reflectarray antenna correctly, for all feed power, the nonlinear impact of amplifier should be considered. It is worth mentioning when feed power is so that all cells are in linear region, linear and nonlinear simulations have the same results.

Using the balanced amplifier by nonlinear response shown in Fig. 8 and considering the power distribution like Fig. 10, the antenna is analyzed again which shows that the antenna gain is increased to 40.2 dBi where gain reduction is decreased to about 0.8 dB and 2.7 dB improvement is reached.

Fig. 12. Performance improvement of the antenna using balanced amplifier when feed power is so that some cells are in nonlinear region.

**VII. CONCLUSION**

It is shown that in some cases, nonlinear analysis of amplifying reflectarray antenna is needed. This paper uses a method combining linear full-wave simulations with the harmonic balance method to predict impact of nonlinearities on the unit cell characteristics of active reflectarrays, as well as on the pattern produced by the reflectarrays. Next, result of nonlinear analysis has been used to design a sample antenna which shows that predicting pattern of the antenna with linear modelling of the active element has error. Finally, a new structure is proposed to improve performance of amplifying reflectarray antenna which improves stability of the antenna and increases total gain of the antenna when incident power is such that the active elements are in nonlinear state. Using the proposed cell, antenna gain reduction in nonlinear state is decreased.

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Parameterized Model Order Reduction for Efficient Time and Frequency Domain Global Sensitivity Analysis of PEEC Circuits

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Abstract — This paper presents a new parameterized model order reduction technique to efficiently perform global time- and frequency-domain sensitivity analysis of electromagnetic systems over the design space of interest. The partial element equivalent circuit (PEEC) method is adopted to build the electromagnetic system model at a set of initial samples in the design space. The block Laguerre-SVD algorithm is proposed to reduce the size of the original equations of the PEEC-based equivalent circuit along with those describing the port voltage and current sensitivities. Then, a multivariate cubic spline interpolation method is used to build a parameterized compact model of port voltages and currents along with their corresponding sensitivities over the entire design space of interest. Finally, two numerical examples are presented, which confirm the accuracy and efficiency of the proposed method.

Index Terms — Parameterized model order reduction, partial element equivalent circuit, sensitivity analysis, time- and frequency-domain circuit simulation.

I. INTRODUCTION

The need to improve the performances of electromagnetic (EM) structures during their early design stage has made sensitivity analysis a necessary tool. The sensitivities represent the system response gradients in the design parameter space, where the design parameters are related to the geometry and/or the materials of the EM structure.

The simplest way to compute sensitivities is represented by the perturbation method, which requires to analyze the EM structure for two different values of each design parameter for a specific nominal point in the design space. It is computationally expensive and often inaccurate, therefore impractical when the number of design parameters to take into account is large and a global sensitivity analysis over the design space of interest is required. Recently, significant progress has been made towards the development of sensitivity analysis approaches to be used along with EM simulators, involving conducting and dielectric objects, both in time- and frequency-domain [1–8]. Differential and integral equation-based methods have been considered for sensitivity analysis [9, 10]. Typical fields of applications are optimization of microwave devices, modeling of signal integrity (SI)/power integrity (PI) problems, control of crosstalk for electromagnetic compatibility (EMC) purposes. These techniques usually turn to be highly demanding in terms of both CPU time and memory resources, since they perform the sensitivity analysis using EM solvers and/or manipulating matrices describing the EM system which are typically very large.

Among EM methods, the partial element equivalent circuit (PEEC) [11] has gained increasing popularity because of its ability to transform the EM system under examination into an equivalent circuit [11–16] that can be represented by modified nodal analysis (MNA) matrix circuit equations [17], studied by means of Kirchhoff principles and simulated using circuit solvers. The PEEC method uses a circuit interpretation of the electric field integral equation (EFIE) [18].

In the context of sensitivity analysis, a PEEC-based method to carry out parameterized sensitivity analysis of EM systems that depend on multiple design parameters has been proposed in [19]. The PEEC method is used to compute state-space matrices of the MNA equations for a set of values of design parameters (e.g., geometrical and substrate parameters). An interpolation process provides parameter-
ized models of these matrices as functions of design parameters [20]. The proposed interpolation scheme is able to compute derivatives of EM matrices, which are needed to perform the system sensitivity analysis. Thus, the algorithm provides sensitivity information over the entire design space of interest (global sensitivity), and not only around one operating point (local sensitivity). Although the method [19] is very accurate and more efficient with respect to the perturbative approach, it suffers from a high computational cost when the size of the MNA matrices of the PEEC circuits becomes large. In [21], a parameterized sensitivity analysis based on a parameterized model order reduction (PMOR) technique is presented. The finite element method is used to generate the equations of the original network, a multiparameter moment matching PMOR technique and an adjoint variable method are used to calculate frequency-domain sensitivities.

In this paper, we propose a new parameterized model order reduction (PMOR) technique to efficiently perform global time- and frequency-domain sensitivity analysis of electromagnetic systems over the design space of interest. The PEEC is adopted to generate a set of PEEC MNA equations and corresponding state-space matrices at a set of design space points. For each of these points in the design space, the block Laguerre-SVD algorithm is proposed to reduce the size of the original equations of the PEEC-based equivalent circuit along with those describing the port voltage and current sensitivities. Then, a methodology based on a multivariate cubic spline interpolation is used to build a parameterized compact model of port voltages and currents along with their corresponding sensitivities over the entire design space of interest. The proposed technique shows a significantly improved efficiency when performing a global sensitivity analysis with respect to the method [19], while maintaining a high accuracy.

The paper is organized as follows. Section II briefly describes the PEEC formulation and the sensitivity formulation while Section III presents the proposed parameterized model order reduction algorithm. Finally some numerical examples are presented in Section IV to validate the proposed technique.

II. PEEC SENSITIVITY FORMULATION

In what follows, we consider a quasi-static PEEC formulation [12]. The Galerkin’s approach is applied to convert the continuous electromagnetic problem described by the EFIE to a discrete problem in terms of electrical circuit quantities, i.e., currents \( i(t) \) and node potentials \( v(t) \). An example of PEEC electrical quantities for a conductor elementary cell is illustrated in the Laplace domain, in Fig. 1 where the current-controlled voltage sources \( sL_p,i_j I_j \) and the current-controlled current sources \( I_{cc} \) model the magnetic and electric field coupling, respectively. Let us denote with \( n_n \) the number of the nodes and with \( n_b \) the number of branches where currents flow. Among this latter, \( n_c \) and \( n_d \) represent the branches of conductors and dielectrics, respectively. Furthermore, let us assume to be interested in generating an admittance (\( Y \)) representation having \( n_p \) output currents \( i_p(t) \) under voltage excitation \( v_p(t) \). Using the MNA formulation [17], the following admittance representation is obtained [20]:

\[
\begin{bmatrix}
    P & 0_{n_n,n_b} & 0_{n_n,n_d} & 0_{n_p,n_p} \\
    0_{n_b,n_n} & L_p & 0_{n_b,n_d} & 0_{n_p,n_p} \\
    0_{n_d,n_n} & 0_{n_d,n_b} & C_d & 0_{n_p,n_p} \\
    0_{n_p,n_n} & 0_{n_p,n_b} & 0_{n_p,n_d} & 0_{n_p,n_p}
\end{bmatrix}
\begin{bmatrix}
    q(t) \\
    d/dt v_d(t) \\
    d/dt i_k(t) \\
    i_k(t)
\end{bmatrix}
= 0_{n_n,n_n} - PA^T 0_{n_n,n_d} PK^T
- AP R \Phi 0_{n_p,n_p}
- 0_{n_d,n_n} - \Phi^T 0_{n_d,n_b} 0_{n_d,n_d} 0_{n_p,n_p}
- K P 0_{n_p,n_n} 0_{n_p,n_b} 0_{n_p,n_d} 0_{n_p,n_p}

\begin{bmatrix}
    0_{n_p, n_n} & 0_{n_p, n_b} & 0_{n_p, n_d} & 0_{n_p, n_p}
\end{bmatrix}
\begin{bmatrix}
    q(t) \\
    i(t) \\
    v_d(t) \\
    i_k(t)
\end{bmatrix}
+ \begin{bmatrix}
    0_{n_p, n_n} & 0_{n_p, n_b} & 0_{n_p, n_d} & 0_{n_p, n_p}
\end{bmatrix}
\begin{bmatrix}
    v_p(t)
\end{bmatrix}
, (1)
\]

\[
i_p(t) = \begin{bmatrix}
    0_{n_p, n_n} & 0_{n_p, n_b} & 0_{n_p, n_d} & 0_{n_p, n_p}
\end{bmatrix}^T
\begin{bmatrix}
    q(t) \\
    i(t) \\
    v_d(t) \\
    i_k(t)
\end{bmatrix}
, (2)
\]

where \( P \in \mathbb{R}^{n_n \times n_n} \) and \( L_p \in \mathbb{R}^{n_b \times n_b} \) are the coefficients of potential and partial inductance matrices, respectively, \( R \in \mathbb{R}^{n_b \times n_b} \) is a diagonal matrix containing the resistances of volume cells and \( C_d \in \mathbb{R}^{n_d \times n_d} \) is the excess capacitance matrix describing the polarization charge in dielectrics [22]. \( A \in \mathbb{R}^{n_n \times n_n} \) is the connectivity matrix, while \( K \in \mathbb{R}^{n_p \times n_n} \) is a selection matrix introduced to define the port voltages in terms of node potentials:

\[
v_p(t) = K \cdot v(t).
\]
In (1), \( q(t) \in \mathbb{R}^{n_x \times 1} \) represents the charges on the conductors, \( i(t) \in \mathbb{R}^{n_i \times 1} \) is the vector of volume currents, \( v_d(t) \in \mathbb{R}^{n_v \times 1} \) describes the voltage drop across the excess capacitance and \( i_k(t) \in \mathbb{R}^{n_p \times 1} \) represents the port currents. \( I_{n_p \times n_p} \) is the identity matrix of dimension equal to the number of ports and \( \Phi \) is:
\[
\Phi = \begin{bmatrix} 0_{n_p, n_p} & I_{n_p, n_p} \end{bmatrix}.
\]

The vector \( i_p(t) \) describes the \( n_p \) port currents that are of opposite sign with respect to \( i_k(t) \). The system of equations (1)-(2) is typically ill-conditioned because the values of charges are usually much smaller than those of currents and voltages. In order to mitigate such a problem, a scaling scheme can be adopted [20]. Equations (1)-(2) can be easily translated from the \( Y \) representation to an impedance \( (Z) \) representation [19] that can be expressed in a compact form as:
\[
C \dot{x}(t) = -G x(t) + B i_p(t),
\]
\[
v_p(t) = L^T x(t),
\]
where \( C \in \mathbb{R}^{n_x \times n_x}, G \in \mathbb{R}^{n_x \times n_i}, B \in \mathbb{R}^{n_x \times n_p}, L = B, x(t) = [q(t) \ i(t) \ v_d(t)]^T \in \mathbb{R}^{n_p \times 1} \) and \( n_s = n_n + n_b + n_d \) is the number of state variables [19].

Considering now the influence of the design parameters \( g = (g_1, \ldots, g_M) \), the formulation (5) becomes:
\[
C(g) \dot{x}(t, g) = -G(g) x(t, g) + B(g) i_p(t),
\]
\[
v_p(t, g) = L^T(g) x(t, g).
\]

In [19], the sensitivity of the voltage outputs with respect to \( M \) design parameters \( g = (g_m)_{m=1}^M \) was computed deriving (6). Merging the original system and corresponding sensitivity system, a new full system can be written as:
\[
\begin{bmatrix} C(g) & 0 \\ C(g)C(g) \end{bmatrix} \begin{bmatrix} \dot{x}(t, g) \\ \dot{x}(t, g) \end{bmatrix} = -\begin{bmatrix} G(g) & 0 \\ G(g)G(g) \end{bmatrix} \begin{bmatrix} x(t, g) \\ \dot{x}(t, g) \end{bmatrix} + \begin{bmatrix} B(g) & 0 \\ B(g)B(g) \end{bmatrix} \begin{bmatrix} i_p(t, g) \\ \dot{i}_p(t, g) \end{bmatrix},
\]
\[
\begin{bmatrix} v_p(t, g) \\ \dot{v}_p(t, g) \end{bmatrix} = \begin{bmatrix} L(g) \hat{L}(g) \end{bmatrix}^T \begin{bmatrix} x(t, g) \\ \dot{x}(t, g) \end{bmatrix}.
\]

where \( \hat{\cdot} \) denotes the derivatives with respect to the design parameters. Equation (7) can be solved, applying appropriate termination conditions, by the means of differential equations solvers, upon the knowledge of the system matrices and their derivatives. However, the simulations become slow and difficult to manage when the dimensions of the original PEEC matrices become large. Therefore, it is fundamental to obtain a parameterized reduced order model able to reduce the CPU time effort needed to carry out the desired simulations.

### III. Parameterized Model Order Reduction Algorithm

In this section, we describe the proposed PMOR algorithm applied to the system (7) in order to perform parameterized (global) time- and frequency-domain sensitivity analysis with respect to design parameters in a more efficient way in comparison with the technique [19], where a parameterized sensitivity analysis was performed using interpolation models of the original PEEC matrices without any order reduction scheme.

#### A. Block model order reduction

The first step of the proposed PMOR algorithm is to generate a set of PEEC matrices \( \{P(g_k), L_p(g_k), C_d(g_k), R(g_k)\}_{k=1}^{K_{tot}} \) corresponding to a set of \( K_{tot} \) initial samples \( g_k \) in the design space. We assume that a topologically fixed discretization mesh is used and that it is independent from the specific design parameter values as in [19]. When geometrical parameters are modified, the mesh is only locally stretched or shrunk. Therefore, the PEEC matrices \( A, \Phi, K \) are uniquely determined by the circuit topology and do not depend on \( g \). Then, a model order reduction (MOR) is proposed to generate the Krylov matrix \( K_q(g_k) \) of the system (7) for each initial sample in the design space. In [23, 24], block structure preserving MOR methods were presented, where blocks were derived based on the specific application. This concept of block structure preserving MOR is used in this paper to generalize the Laguerre-LSVD MOR (LSVD-MOR) algorithm [25]. The standard LSVD-MOR is listed in Algorithm 1, where \( \alpha \) is a positive scaling parameter, \( q - 1 \) is the order of approximation and:
\[
K_q = \begin{bmatrix} R^{(0)}_1 & R^{(1)}_1 & \ldots & R^{(q-1)}_1 \\ R^{(0)}_2 & R^{(1)}_2 & \ldots & R^{(q-1)}_2 \\ \vdots & \vdots & \ddots & \vdots \\ R^{(0)}_{M} & R^{(1)}_{M} & \ldots & R^{(q-1)}_{M} \end{bmatrix},
\]

is the Krylov matrix of order \( q - 1 \) [25]. The LSVD-MOR algorithm can be extended to a block LSVD-MOR method, considering the block form of the matrices in (7). If we replace the set of matrices \( \{C, G, B, L\} \) in Algorithm 1 with that block form, the step \( k = 0 \) of the standard LSVD-MOR algorithm can be re-written as:
\[
\begin{bmatrix} G + \alpha C & 0 \\ G + \alpha \hat{C}G + \alpha C \end{bmatrix} \begin{bmatrix} R^{(0)}_{11} & R^{(0)}_{12} \\ R^{(0)}_{21} & R^{(0)}_{22} \end{bmatrix} = \begin{bmatrix} B & 0 \\ 0 & \hat{B} \end{bmatrix}.
\]

After some manipulation it follows:
\[
\begin{bmatrix} R^{(0)}_{11} & R^{(0)}_{12} \\ R^{(0)}_{21} & R^{(0)}_{22} \end{bmatrix} = \begin{bmatrix} G^{-1}_N B & 0 \\ G^{-1}_N B + G^{-1}_N \hat{B}G^{-1}_N B \end{bmatrix} = \begin{bmatrix} R^{(0)}_{11} & R^{(0)}_{12} \\ R^{(0)}_{21} & R^{(0)}_{22} \end{bmatrix},
\]
where we have defined $G_N$ and $\tilde{G}_N$ as
\[ G_N = G + \alpha C \]
\[ \tilde{G}_N = G + \alpha \tilde{C}. \] (11)
At the step $k = 0$, the Krylov matrix $K_q$ is composed of two components:
\[ \begin{bmatrix} R_1^{(0)} = G_N^{-1}B \\ R_2^{(0)} = \tilde{G}_N^{-1}B + G_N^{-1}\tilde{B} \end{bmatrix} \] (12)
The first component is related to the Krylov matrix of the original system (5) for $k = 0$, while the second one is the derivative of this first component:
\[ \begin{bmatrix} R_2^{(0)} = \tilde{R}_1^{(0)} \end{bmatrix} \] (13)
This means that the Krylov matrix of the original system and its derivative are enough to finalize the first step of the block LSVD-MOR algorithm. The other steps for for $k > 0$ lead to:
\[ \begin{bmatrix} G + \alpha C & 0 \\ G + \alpha \tilde{G} + \alpha \tilde{C} \end{bmatrix} \begin{bmatrix} R_1^{(k)} & R_1^{(k)} \\ R_2^{(k)} & R_2^{(k)} \end{bmatrix} = \begin{bmatrix} G - \alpha C & 0 \\ G - \alpha \tilde{G} - \alpha \tilde{C} \end{bmatrix} \begin{bmatrix} R_1^{(k-1)} & R_1^{(k-1)} \\ R_2^{(k-1)} & R_2^{(k-1)} \end{bmatrix}. \] (14)
Likewise, after some manipulations:
\[ \begin{bmatrix} R_1^{(k)} & R_1^{(k)} \\ R_2^{(k)} & R_2^{(k)} \end{bmatrix} = \begin{bmatrix} G_N^{-1}0 & R_1^{(k-1)} \\ R_2^{(k-1)} \end{bmatrix} = \begin{bmatrix} G_N^{-1}0 & R_1^{(k-1)} \\ R_2^{(k-1)} \end{bmatrix}, \]
where we have defined $G_R$ and $\tilde{G}_R$ as:
\[ G_R = G - \alpha C \]
\[ \tilde{G}_R = G - \alpha \tilde{C}. \] (16)
The blocks of the Krylov matrix for $k > 0$ are also composed of two components:
\[ \begin{cases} R_1^{(k)} = G_N^{-1}G_R R_1^{(k-1)} \\ R_2^{(k)} = G_N^{-1}G_R R_2^{(k-1)} + \left( G_N^{-1}G_R + G_N^{-1}\tilde{G}_R \right) R_1^{(k-1)} \end{cases} \] (17)
As previously, equation (17) shows that the first component $R_1^{(k)}$ is the component of the Krylov matrix of the original system (5) for $k > 0$, while the second component $R_2^{(k)}$ is the corresponding derivative:
\[ \begin{bmatrix} R_2^{(k)} \end{bmatrix} = \begin{bmatrix} \tilde{R}_1^{(k)} \end{bmatrix}. \] (18)
According to Algorithm 1, the Krylov matrix $K_q$ reads:
\[ K_q = \begin{bmatrix} R_1^{(0)} & R_1^{(1)} & \cdots & R_1^{(q-1)} \\ R_2^{(0)} & R_1^{(1)} & \cdots & R_1^{(q-1)} \\ \vdots & \vdots & \ddots & \vdots \\ R_2^{(0)} & R_1^{(1)} & \cdots & R_1^{(q-1)} \end{bmatrix} \] (19)
and by a column permutation:
\[ K_q = \begin{bmatrix} R_1^{(0)} & \cdots & R_1^{(q-1)} \\ \vdots & \ddots & \vdots \\ R_2^{(0)} & \cdots & R_1^{(q-1)} \end{bmatrix}, \] (20)
where $R_1^{(i)} = R^{(i)}$ is the $i$-th component of the Krylov matrix of the original system and $R_2^{(i)} = \tilde{R}^{(i)}$ is its derivative (see (13) and (18)). For each initial sample $g_k$ the corresponding Krylov matrix $K_q(g_k)$ is computed using the block LSVD-MOR method.
To compute the reduced system of (7), an orthonormalization step is applied to the Krylov matrix $K_q$ using a Singular Value Decomposition (SVD) algorithm to obtain the projection matrix:
\[ V \rightarrow V \Sigma U^T = \text{SVD}(K_q), \] (21)
that is then used for the congruence transformations to compute the reduced matrices. In order to preserve the block structure of the system (7), the projection matrix $V \in \mathbb{R}^{2n_r \times n_r}$ is partitioned in a block fashion [23, 24]:
\[ V = \begin{bmatrix} V_1 \\ V_2 \end{bmatrix}, \] (22)
where $n_r = q \cdot n_p$ denotes the column size of the Krylov matrix $K_q$. The projection matrix blocks may not have a full column rank $n_r$, in particular the first block $V_1$, due to the presence of a block of zeros. An orthonormalization step is applied to $V_1$ in order to obtain a matrix $\tilde{V}_1$ whose columns span the same space as the columns of $V_1$, while having however a full column rank [24]. The same operation is performed on $V_2$ to obtain $\tilde{V}_2$. The second block $V_2$ has generally full column rank $n_r$. After this step the column size of the two blocks $V_1$ and $V_2$ may differ with respect to the column size of $V_1$ and $V_2$. However, this does not influence the reduction of the full system (7). Finally the projection matrix is written as [23, 24]:
\[ \tilde{V} = \begin{bmatrix} \tilde{V}_1 & 0 \\ 0 & V_2 \end{bmatrix}. \] (23)
In order to reduce the overall system in (7), the projection matrix (23) is used to compute the reduced matrices by congruence transformations:
\[ \begin{bmatrix} C_{r,1} & 0 \\ C_{r,2} & C \end{bmatrix} = \begin{bmatrix} \tilde{V}_1^T & 0 \\ 0 & V_2^T \end{bmatrix} \begin{bmatrix} C_0 & \tilde{V}_1 \\ \tilde{C}C & 0 V_2 \end{bmatrix}, \] (24)
Algorithm 2 lists the steps of the proposed block LSVD-MOR method to perform in order to obtain
the set of reduced matrices for the reduced system. Concerning the proposed PMOR algorithm, the block LSVD-MOR method is used to generate the Krylov matrix $K_i(g_k)$ of the system (7) for each initial sample in the design space.

**B. Interpolation models**

Once the set of matrices \{P(g_k), L_p(g_k), C_d(g_k), R(g_k), K_i(g_k)\} for each point of interest in the design parameter space, is available, the corresponding interpolation models are built. Figure 2 shows an example for the case of one design parameter. Each interpolation model is built starting from a set of samples indicated with red dots (•) in Fig. 2. For each (•) sample in the design space, a set of PEEC matrices and a Krylov matrix are computed and the corresponding models, that cover the entire design space, are built by the means of an interpolation scheme. For interpolation purposes the multivariate cubic spline interpolation method [26], which is well-known for its stable and smooth characteristics, is used. The proposed interpolation scheme is continuous in the first and second order derivatives and can be used in the general case of an M-dimensional (M-D) design space.

First, the interpolation models \{P(g), L_p(g), C_d(g), R(g)\} are computed while guaranteeing positive definiteness and semidefiniteness matrix properties [19, 20]. At this step, according to Algorithm 2, it could be possible to compute a set of reduced matrices for each point of interest in the design space by using the block LSVD-MOR method previously described. However, to improve the efficiency of the proposed PMOR method, we choose to create an interpolation model for the $K_i(g)$ matrix, starting from the corresponding data samples $K_i(g_k)$. From the model $\tilde{K}_i(g)$, the values of the projection matrices $V_1(g)$ and $V_2(g)$ can be computed for each point of interest in the design space.

Inspecting equation (19), only the interpolation models of the matrices \{${R}_1^{(i)}(g_k)$\} for each point of interest in the design space.

and \{${R}_2^{(i)}(g_k)$\} for each point of interest in the design space. Furthermore, there is still one degree of freedom about how to compute the matrices \{${R}_2^{(i)}(g_k)$\} for each point of interest in the design space.

1. As described in Algorithm 2;

2. directly as derivative of \{${R}_1^{(i)}(g_k)$\} for each point of interest in the design space. This choice is preferred since it allows storing only the model \{${R}_1^{(i)}(g_k)$\} and then saving memory resources, while keeping a similar accuracy.

Using the interpolation models \{P(g), L_p(g), C_d(g), R(g)\} and \{${R}_1^{(i)}(g_k)$\} for each point of interest in the design space. Applying the congruence transformations, the sets of reduced matrices \{C_{r,1}(g), G_{r,1}(g), B_{r,1}(g), L_{r,1}(g)\} for each point of interest in the design space. The reduced version of (7) can be written as:

\[
\begin{bmatrix}
C_{r,1}(g) & 0 \\
C_{r,2}(g) & C_{r,2}(g)
\end{bmatrix}
\begin{bmatrix}
x_r(t, g) \\
\hat{x}_r(t, g)
\end{bmatrix}
= -\begin{bmatrix}
G_{r,1}(g) & 0 \\
G_{r,2}(g) & G_{r,2}(g)
\end{bmatrix}
\begin{bmatrix}
x_r(t, g) \\
\hat{x}_r(t, g)
\end{bmatrix}
+ \begin{bmatrix}
B_{r,1}(g) & 0 \\
B_{r,2}(g) & B_{r,2}(g)
\end{bmatrix}
\begin{bmatrix}
i_p(t, g) \\
\hat{i}_p(t, g)
\end{bmatrix}
+ \begin{bmatrix}
L_{r,1}(g) & L_{r,2}(g) \\
0 & L_{r,2}(g)
\end{bmatrix}^T
\begin{bmatrix}
0 \\
x_r(t, g) \\
\hat{x}_r(t, g)
\end{bmatrix}.
\]

These reduced matrices are used to perform both time- and frequency-domain sensitivity analysis, with appropriate termination conditions.

**IV. NUMERICAL EXAMPLES**

Two numerical examples are proposed to validate the proposed PMOR technique for sensitivity analysis. Parameterized time- and frequency-domain sensitivity analyses are performed with the proposed PMOR technique and the results are compared with the approach proposed in [19] and with the perturbative approach (with respect to each parameter $g_m$) that in time- and frequency-domain reads:

\[
\hat{v}_{p,g_m}(t, g) = \frac{v_p(t, g_1, \ldots, g_m + \Delta g_m, \ldots, g_M) - v_p(t, g)}{\Delta g_m}.
\]

\[
\hat{Z}_{g_m}(s, g) = \frac{Z(s, g_1, \ldots, g_m + \Delta g_m, \ldots, g_M) - Z(s, g)}{\Delta g_m}.
\]
where $m = 1, \ldots, M$, $M$ and $\Delta g_m$ represent the number of parameters and the increment, respectively. The accuracy of the perturbative approach depends on the choice of the increment $\Delta g_m$: if the increment is not small enough, the estimation of the derivative is not accurate, while if the perturbation is very small compared with the nominal value, numerical problems may occur due to numerical noise. This may lead to inaccurate computation of the system sensitivities. In contrast, thanks to the interpolation models, the methods presented in this paper and in [19] lead to more accuracy and numerical stability, since the derivatives are computed from continuously differentiable polynomials built by means of spline functions. The method [19] is denoted as Full Parameterized while the proposed method is denoted as Block PMOR in what follows.

In the numerical results (see Tables 1-2), for the Full Parameterized and Block PMOR models, the model evaluation CPU time indicates the average time needed to evaluate the corresponding parameterized models in a point of the validation grid in order to obtain a set of PEEC matrices. Moreover, for the Block PMOR model, the SVD operation is also part of the model evaluation. For the Perturbative Approach, the model evaluation CPU time refers to the average time needed to compute a set of PEEC matrices by a PEEC solver at and around a point in the validation grid, which are then used for a finite difference calculation. Once the parameterized models are evaluated, or PEEC matrices have been computed (perturbative approach), they can be used to carry out sensitivity analysis in frequency- and time-domain. For each of the three methods, the average time needed to perform the sensitivity analysis in a point of the validation grid is denoted as simulation CPU time.

Numerical simulations have been performed on a Linux platform on an AMD FX(tm)-6100 Six-Core Processor 3.3 GHz with 16 GB RAM.

A. Metallic enclosure coupled to a transmission line

In the first example a metallic enclosure coupled to a transmission line is studied. The cross section is shown in Fig. 3. The geometrical dimensions are $w_c = 1 \text{ mm}$, $d_c = 5 \text{ mm}$, $s_1 = 1 \text{ mm}$, $d_1 = 30 \text{ mm}$. A set of PEEC matrices is computed over a grid of $6 \times 6$ values of the conductor length $l_c \in [25 - 65] \text{ mm}$ and $s_c \in [7 - 22] \text{ mm}$ in order to build the aforementioned parameterized reduced order model. Furthermore, a parameterized full model has been built by means of the method [19] for the comparison. The order of the full models is 2870 while the reduced order is 471. Parameterized time- and frequency-domain sensitivities are performed over a validation grid of $5 \times 5$ values of $l_c \in [29 - 61] \text{ mm}$ and $s_c \in [8.5 - 20.5] \text{ mm}$. The obtained results are then compared with the ones obtained by the perturbative model (26), (27). For the time-domain results, the bottom conductor is excited by a smooth pulse voltage source with amplitude $1 \text{ V}$, rise/fall times $\tau_r = \tau_f = 1.5 \text{ ns}$, width $6 \text{ ns}$ and internal resistance $R_T = 50 \Omega$. All the ports are terminated on $50 \Omega$ resistances.

Figures 4-5 and 6-7 show time- and frequency-domain results that confirm the high accuracy of the proposed approach. Table 1 clearly shows the computational advantage of the proposed PMOR technique: the CPU time required to perform both time- and frequency-domain sensitivity simulations is considerably improved with respect to the two other compared approaches for a sensitivity analysis around a nominal design point and therefore also for a global sensitivity analysis. It is important to note that the Block PMOR and Full Parameterized methods require a one-time effort to build the corresponding parameterized models that then allow global sensitivity simulations. This CPU time for the Block PMOR method is equal to 8236 s and for the Full Parameterized method is equal to 7513 s.

![Fig. 3. Metallic enclosure coupled to a transmission line (example A).](image-url)
Fig. 4. Time-domain voltage sensitivity with respect to $l_c$ at the port 1 (example A).

Fig. 5. Time-domain voltage sensitivity with respect to $s_c$ at the port 2 (example A).

Fig. 6. Magnitude of the sensitivity of $Z_{11}$ with respect to $s_c$ (example A).

Fig. 7. Magnitude of the sensitivity of $Z_{12}$ with respect to $l_c$ (example A).

Fig. 8. Three conductors microstrip (example B).

B. Three conductors microstrip

Three coupled microstrips are modeled in this example and Fig. 8 shows the corresponding cross section. The geometrical dimensions are $w_c = 178 \mu m$, $t_c = 35 \mu m$, $d = 3 \, mm$ and the length of the lines is $l = 40 \, mm$. The parameterized reduced order and parameterized full models have been built starting from a set of PEEC matrices computed over a grid of $6 \times 6$ values of $s_c \in [0.1 \, mm, 0.4 \, mm]$ and $h \in [0.1 \, mm, 0.3 \, mm]$. The order of the full model is 3360 while the reduced order is 577. Parameterized time- and frequency-domain sensitivities are performed over a validation grid of $5 \times 5$ values of $s_c \in [0.13 \, mm, 0.37 \, mm]$ and $h \in [0.12 \, mm, 0.28 \, mm]$. The obtained results are then compared with the perturbative approach (26), (27). For the time-domain results, a smooth pulse voltage source with amplitude 1V, rise/fall times $\tau_r = \tau_f = 1.5 \, ns$, width 6 ns and internal resistance $R_T = 50 \, \Omega$ is applied on the first conductor. All the ports are terminated on 50 $\Omega$ resistances. Time-domain sensitivity results are shown in Figs. 9, 10 while frequency-domain sensitivity results are shown in Figs. 11, 12. As in the previous example, the results confirm the high accuracy of the proposed method. Table 2 shows the simulation time comparison that clearly confirms that the proposed PMOR method considerably reduces the CPU time.
required for a sensitivity analysis with respect to the other two methods around a nominal design point and therefore for a global sensitivity analysis. As in the previous example, an initial computational effort is required to compute the Block PMOR and Full Parameterized models that is equal to 9306 s for the Block PMOR and to 8456 s for the Full Parameterized method.

V. CONCLUSIONS

In this paper we have presented a new PMOR technique to perform both time- and frequency-domain global sensitivity analysis of PEEC circuits. It is based on the PEEC method, the block Laguerre SVD model order reduction technique and interpolation schemes. Two numerical examples confirm the high modeling capability and the improved efficiency of the proposed approach with respect to existing sensitivity analysis methods.

Table 2: Simulation time comparison (Example B).

<table>
<thead>
<tr>
<th></th>
<th>Block PMOR</th>
<th>Full Parameterized</th>
<th>Perturbative Approach</th>
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</thead>
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<td><strong>Time</strong></td>
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<td></td>
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<td>699 s</td>
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<tr>
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<td>48 s</td>
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<tr>
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<td>10 s</td>
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<tr>
<td>Simulation</td>
<td>52 s</td>
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<tr>
<td>Total</td>
<td>63 s</td>
<td>601 s</td>
<td>1556 s</td>
</tr>
</tbody>
</table>
VI. APPENDIX

This appendix illustrates the pseudo-code of the Laguerre-SVD and the Block Laguerre-SVD algorithms.

Algorithm 1 Laguerre-SVD Algorithm

Select values for $\alpha$ and $q$

$R^{(0)} \leftarrow (G + \alpha C)R^{(0)} = B$

for $k = 1 \rightarrow q-1$

$R^{(k)} \leftarrow (G + \alpha C)R^{(k)} = (G - \alpha C)R^{(k-1)}$

end for

$K_q = [R^{(0)}, R^{(1)}, \ldots, R^{(q-1)}]$

$V\Sigma U^T = \text{SVD}[K_q]$

$C_r \leftarrow V^T C V$

$G_r \leftarrow V^T G V$

$B_r \leftarrow V^T B_r$

Algorithm 2 Block Laguerre-SVD Algorithm

Select values for $\alpha$ and $q$

$G_N \leftarrow G + \alpha C$

$G_R \leftarrow G - \alpha C$

$R_1^{(0)} \leftarrow G_N^{-1} B$

$R_2^{(0)} \leftarrow G_N^{-1} B + \hat{G}^{-1} B$

for $k = 1 \rightarrow q-1$

$R_1^{(k)} \leftarrow G_N^{-1} G_R R_1^{(k-1)}$

$R_2^{(k)} \leftarrow \left( G_N^{-1} G_R + G_N^{-1} \hat{G}^{-1} \right) R_1^{(k-1)}$

$G_N^{-1} G_R R_2^{(k-1)}$

end for

$K_q = [R_1^{(0)}, \ldots, R_1^{(q-1)}, 0, \ldots, 0, R_2^{(0)}, \ldots, R_2^{(q-1)}]$

$n_r = q \times n_p$

$V = [V_1, V_2]$

for $i = 1, 2$

If rank $V_i = r_{V_i} \leq n_r$ determine an $n_s \times r_{V_i}$ matrix $\tilde{V}_i$ with colspan $\tilde{V}_i = \text{colspan } V_i$ and rank $V_i = r_{V_i}$

end for

$\tilde{V} = [\tilde{V}_1, 0]

0 \tilde{V}_2$

Apply congruence transformations (24)

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REFERENCES


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Finite Difference Analysis of an Open-Ended, Coaxial Sensor Made of Semi-Rigid Coaxial Cable for Determination of Moisture in *Tenera* Oil Palm Fruit


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**Abstract** — In this paper, the use of the Finite Difference Method (FDM) is proposed to determine the reflection coefficient of an open-ended coaxial sensor for determining the moisture content of oil palm fruit. Semi-rigid open-ended coaxial sensor is used in conjunction with Vector Network Analyzer for reflection coefficient measurement of oil palm fruit. Moisture content in oil palm fruit determine optimum harvest time of oil palm fruit. Finite difference method is then used to simulate measured reflection coefficient due to different moisture contents in oil palm fruit at various stages of ripeness. The FDM results were found to be in good agreement with measured data when compared with the quasi-static and capacitance model. Overall, the mean errors in magnitude and phase for the FDM were 0.03 and 3.70°, respectively.

**Index Terms** — Finite difference method, moisture content, oil palm fruit, open-ended coaxial sensor, reflection coefficient.

**I. INTRODUCTION**

**A. Background of oil palm**

The oil palm, *Elaeis guineensis* Jacq, is indigenous to West Africa where the cultivation area is from Sierra Leone, Liberia, the Ivory Coast, Ghana, Cameroon and extended to the equatorial regions of the Republics of Congo and Zaire [1].

The harvesting period begins around 24 to 30 months after planting [2] and each palm can produce between 8 to 15 fresh fruit bunches (FFB) per year. The weight of each bunch is about 15 to 25 kg each and this depends on the planting material and age of the palm. Each FFB contains about 1000 to 1300 fruit. Each fruit consists of 3 layers, which are the fibrous mesocarp layer, the endocarp (shell) and the kernel (Fig. 1).

Palm oil is obtained from the fleshy mesocarp, which is composed of 45-55 per cent oil by weight [3]. The *Tenera* has been the preference for the palm oil industry because of its thin shell and high oil content in the thick mesocarp structure.
B. Conventional technique of determining the fruit ripeness

There are several techniques to gauge the oil palm fruit ripeness. The visible symptoms to determine fruit ripeness include the color change of the fruit [4]-[5], the percentage or number of detached fruit per bunch [6] and the fruit ability to float on water, or so called floatation technique [7]. However, they are unreliable due to their inconsistencies and inaccuracies.

C. The relationship between moisture content and ripeness of oil palm fruit

The current research in gauging the ripeness of oil palm fruit is via examination of the amount of moisture content in mesocarp of an oil palm fruit. Ariffin et al. [8] states that the moisture content in mesocarp of oil palm fruit can be used as an indicator to determine the fruit ripeness. It was found that the moisture content is higher in unripe oil palm fruit at the early stage of fruit development. The water in the mesocarp decreases gradually during fruit ripening which coincides with the oil accumulation approximately from week 12 to week 15 after anthesis (Fig. 2). The amount of water in fresh mesocarp decreases rapidly to 40% in the ripe fruit from week 16 to week 17 after anthesis. The water content will then decreases slowly from week 18 to week 24. The moisture content decrease is almost about the same time as the accumulation of oil in the mesocarp. Hence, there is a close relationship between the moisture content (mc) and oil content (oc) in mesocarp. This phenomenon is helpful to gauge the fruit ripeness. Hartley [1] states that the mass fraction of oil and mass fraction of water in the mesocarp can be expressed linearly. This relationship are visualized in Fig. 3.

\[
\text{m.c.} = \frac{m_{\text{before dry}} - m_{\text{after dry}}}{m_{\text{before dry}}} \times 100\% ,
\]

where \( m_{\text{before dry}} \) and \( m_{\text{after dry}} \) are the weight of fruit sample before dried and after dried, respectively.
B. Simulation and measurement of open-ended coaxial sensor on oil palm fruit

Open-ended coaxial sensors have been used extensively to measure the reflection coefficient of oil palm fruit [10]-[12]. The probe associated with such a sensor is made of an RG-402 semi-rigid cable, normally operating at 2 GHz. The stage of fruit ripeness is determined by the percentage of moisture content. As the moisture content (or permittivity, \( \varepsilon_r \)) of the fruit changes, the values of reflection coefficient measured by the sensor also changed.

Unfortunately, both the quasi-static model (a.k.a. the admittance model) and the capacitance model assume that the thickness of the sample under consideration is infinite [13]. Therefore, these models are inappropriate for characterizing a thin sample or any sample with finite thickness, such as oil palm fruit. However, the dimensions of the sample must be taken into account in FDM calculation [14]-[15]. For instance, the length and thickness of the fruit are considered in FDM.

C. The moisture content and the dielectric properties in oil palm fruit

The moisture content of agricultural products is one of the most important parameters for determining the quality of the products. This information is required to determine the optimum time for harvesting and safe storage.

The standard oven-drying method is tedious and time-consuming, and they are not suitable for use in agro-production application. Hence, the development of a rapid test method, such as microwave method, is a pressing need in the industry. The complex dielectric permittivity, \( \varepsilon' \) is often expressed by equation (2):

\[ \varepsilon' = \varepsilon'' - j\varepsilon'' \]  

where \( \varepsilon' \) is related to the ability of the material to store energy (dielectric constant) and \( \varepsilon'' \) is the loss factor which is the dissipation of energy in the material. The permittivity of oil palm fruit [17] can be expressed as:

\[ \sqrt{\varepsilon'_{w}} = \varepsilon_r \sqrt{\varepsilon'_{w}} + \varepsilon_r \sqrt{\varepsilon'_t} + \varepsilon_r \sqrt{\varepsilon'_f} \]  

where \( \varepsilon_{w}, \varepsilon_{t}, \) and \( \varepsilon_{f} \) are the volume fraction of water, fiber, and oil, respectively, and \( \varepsilon'_{w}, \varepsilon'_{t}, \) and \( \varepsilon'_{f} \) are the corresponding complex permittivities. It has been shown that both \( \varepsilon'_{w} \) and \( \varepsilon'_{0} \) are essentially constant throughout the frequency range between DC and 10 GHz with \( \varepsilon'_{w} = 2.2 - 0.06j \) and \( \varepsilon'_{0} = 2.3 - 0.02j \). The values of \( \varepsilon'_{w} \) are obtained from the Cole-Cole model [18]:

\[ \varepsilon'_{w} = \varepsilon_{in} + \frac{\varepsilon_{in} - \varepsilon_{o}}{1 + \left( j \omega \tau \right)^{1-\alpha}}, \]  

where \( \alpha' \) is the distribution parameter, which is an empirical constant. Thus, the palm oil mixture consists of three main components, i.e., \( \varepsilon_{o}, \varepsilon_{w}, \) and \( \varepsilon_{t} \), and the relationship between them is:

\[ \varepsilon = 1 - \varepsilon_{w} - \varepsilon_{t}. \]  

Since \( \varepsilon_{t} = 0.16 \) [1], \( \varepsilon_{w} \) can be calculated as:

\[ \varepsilon_{w} = \frac{(m.c.)\varepsilon_{w} + \varepsilon_{o} - \varepsilon_{o}Y_{f}}{\rho_{o} - (m.c.)\rho_{o} + (m.c.)\rho_{f}}, \]  

where the densities \( \rho_{o}, \rho_{t}, \) and \( \rho_{0} \) are 1, 0.92 and 0.93 respectively and \( m.c. \) is the moisture content. The volume fraction of oil and water can be found by using Equation (5) and Equation (6), respectively. In Equation (7), the relative moisture content in the wet basis can be determined in terms of the mass of water, oil, and fiber, which are represented by \( m_{w}, m_{o}, \) and \( m_{f} \), respectively [19]:

\[ m.c. = \frac{m_{w}}{m_{w} + m_{o} + m_{f}} \times 100\%. \]  

Hence, the permittivity of the oil palm fruit can be calculated using the mixture model [12].

Figure 4 shows the permittivity of oil palm fruit for \( m.c. \) between 20% and 90%. The abnormal behavior of \( \varepsilon' \) with \( m.c. \) below 30% is due to bound water [20]-[21].

\[ \Gamma = \frac{Z_{L} - Z_{0}}{Z_{L} + Z_{0}} = 1 - \frac{Y_{f}}{Y_{L}}, \]  

where \( Z_{0} \) is the 50 \( \Omega \) characteristic impedance of the coaxial sensor. The normalized admittance, \( \tilde{Y} \) [12], [22], is established by two terms, i.e., normalized conductance, \( G(0)/Y_{0} \), and susceptance \( B(0)/Y_{0} \). \( \tilde{Y} \) can be expressed as:

\[ \tilde{Y} = \frac{Y_{o} - jY_{f}}{Y_{o} + jY_{f}}. \]
was 3.29 is the relative permittivity of the sample and the filling of the coaxial line.

F. Iteration method in solving finite difference method (FDM)
The computation work of FDM involves large system of simultaneous equations, and iterative method was used to overcome these. Iterative method uses the approximation from previous computation to calculate the next approximation. This computation is carried out iteratively until its value converges.

Initial values of the potentials were set at the free nodes which equals to zero or to any reasonable value. For example, we set 1 V at the excitation plane and 0 V at the ground conductor or perfect electric conductor (PEC). These potential values are arranged to form a matrix. Maintaining the potentials at the fixed nodes constant at all times, then applying the equation:

\[ V_{i,j} = \frac{1}{4} (V_{i+1,j} + V_{i-1,j} + V_{i,j+1} + V_{i,j-1}) \]

Fig. 5. Finite difference solution pattern: finite-difference, five-node molecule.

G. Application of the concept of finite-difference in the coaxial sensor and sample
Plane CD is the boundary between two different materials, i.e., the Teflon in the coaxial line and the sample. At the dielectric boundary (Fig. 6), the boundary condition,

\[ D_{1n} = D_{2n} \]
must be imposed where \( D_{1n} \) and \( D_{2n} \) are the normal components of the electric flux density at dielectric filler in coaxial line and in the sample being tested, respectively. This condition is based on Gauss’ Law for electric fields, i.e.,

\[
\oint D \cdot dl = \oint E \cdot dl = Q_{\text{enc}} = 0,
\]

since no free charge is deliberately placed on the dielectric boundary. Substituting \( E = -\nabla V \) in Equation (19) gives:

\[
0 = \oint \varepsilon \nabla V \cdot dl = \oint \varepsilon V^{\prime} \cdot dl,
\]

where \( \frac{\partial V}{\partial n} \) denotes the derivative of \( V \) normal to the contour \( l \). Applying Equation (20) to the interface in Fig. 6 yields:

\[
V_0 = \frac{\varepsilon_1 - \varepsilon_2}{2(\varepsilon_1 + \varepsilon_2)} V_1 + \frac{\varepsilon_2}{2(\varepsilon_1 + \varepsilon_2)} V_3 + \frac{1}{4} V_2 + \frac{1}{4} V_4.
\]

The finite difference potential results on plane CD in Fig. 6 (circular ring potential in the area of the cross section of coaxial sensor, \( V_{\text{ring}} \)) were computed. The total potential, \( V_{\text{area}} \), and the total charge, \( Q_{\text{area}} \), at the area of the aperture of the probe can be determined easily by using Equations (22) and (23), respectively [23]:

\[
V_{\text{area}} = \int_a^b V_{\text{ring}} dp,
\]

\[
Q_{\text{area}} = \varepsilon \int_a^b V_{\text{ring}} \rho dp.
\]

where \( \rho \) is the radius at aperture of the coaxial probe, \( a \) is the inner radius of the coaxial probe, and \( b \) is the outer radius of the coaxial probe. The normalized and characteristic admittance are expressed as:

\[
\tilde{Y} = \frac{j\omega C}{Y_0},
\]

\[
\tilde{Y}_0 = \frac{2\pi}{\varepsilon_0 \varepsilon_r \ln \left( \frac{b}{a} \right)}.
\]

where \( \varepsilon_0 \) is the permittivity in free space, \( \varepsilon_r \) is the relative permittivity of the coaxial line (PTFE), and \( \mu_0 \) is the free space of permeability. The reflection coefficient, \( \Gamma \) is obtained from Equation (8).

III. RESULTS AND DISCUSSION

A. Magnitude of the reflection coefficient

The results comparison for the measured and calculated values of the reflection coefficient at various percentages of moisture content in oil palm fruit is shown in Fig. 7.

The whole results suggested that the magnitude of the reflection coefficient decreases as the moisture content of the fruit increased [24]. The results obtained using the mixture model indicated that complex permittivity, \( \varepsilon^* \) increased when the moisture content is high. This relationship, which is due to the high degree mismatch of impedance, is clearly shown in Fig. 8. Increases in \( \varepsilon^* \) could cause the sample’s impedance, \( Z_L \) to decrease. The admittance model can be used to calculate this.

In summary, increasing the moisture content causes the complex permittivity to increase, as Fig. 4 shows. Hence, this condition results in the decrease of impedance, which, in turn, causes the magnitude of the reflection coefficient, \( |\Gamma| \), to decrease. Figure 8 shows this relationship as a 3D line plot.
It can be noticed that the magnitude of the reflection coefficient, $|\Gamma|$ decreases gradually against moisture content, $mc$ (%). The trend line in Fig. 2 that represents water and oil content seems unchanged in this mc range. The fruit seems to be at constant water and oil level. The mean magnitude error of FDM, admittance model and capacitance model were close to the mean magnitude error that is presented by the fitting line shown in Fig. 9, i.e., 0.01. The insignificant change in moisture content yield to the insignificant change in their magnitude of reflection coefficient as well. It can be proved by the sensitivity in Fig. 10. Figure 10 indicates the sensitivity of $\frac{d|\Gamma|}{d(mc)}$ in region 1. It can be noticed that the sensitivity is kept constant when the mc increases from 20% to 40%. It means that it is best represented as a linear relationship. It has been proved by the fitting linear equation in Fig. 9. The sensitivity value is -0.0017 and it is very small. This can be explained by Fig. 2. In Fig. 2, the range of water content which is between 20% to 40% shows the insignificant change when the fruit exceeds week 17 after anthesis.

According to the admittance model, decreasing $Z_L$ (as a result of a greater dielectric constant) results in a decrease in the magnitude of the reflection coefficient. The relationship between the normalized admittance, \[ \bar{Y}_L = \frac{Z_0}{Z_L} \] and the reflection coefficient, $\Gamma$ is shown by the Equation (25). $Z_0$ is the 50 Ω characteristic impedance of the coaxial sensor. Figure 8 shows the relationship between magnitude of reflection coefficient, magnitude of impedance, and moisture content.

The FDM, admittance model and capacitance model produced trends that were similar to the measurement results. The magnitudes that were acquired by FDM showed better agreement with the measured data than the admittance model or the capacitance model [19]. The FDM provided a mean error of 0.03 for the moisture content ranging from 20% to 90%. The mean errors produced by the admittance model and the capacitance model are 0.06 and 0.05, respectively. The poor accuracies of the admittance model and the capacitance model were due to the assumptions that were made in the models. In both models, it is assumed that the thickness of the sample is infinite [13]. Therefore, neither one of them is suitable for use in characterizing a thin sample or any sample with a finite thickness, such as the sample of oil palm fruit. However, the dimensions of the sample, i.e., its length and width, must be taken into account in the FDM calculation. The PML is necessary to truncate the computation region of the material, in order to retain the practicability of the computation.

Among the three models, the FDM approach had the best agreement with the measured values of the reflection coefficient, as shown in Fig. 7.

Figure 9 represents a portion of Fig. 7, which designated as region 1 in Fig. 7. Region 1 is in mc range from 20% to 40%. Meanwhile, the region where the mc range is from 40% to 90% is designated as region 2. These two mc ranges are important to study the period after anthesis. The relationship of water content in fruit and the period after anthesis can be referred to Fig. 2.

Measurement data shows $|\Gamma|$ decreases gradually when moisture content increases. Referring to Fig. 2, it can be observed that the mc in the range of 20% to 40% is within 18 weeks to 24 weeks after anthesis. During this period, the water content and oil content show insignificant change. The fruit accumulates maximum amount of oil content in this mc range. It can be used to determine the optimum of harvesting time of oil palm fruit. Therefore, the relationship of $|\Gamma|$ against moisture content can be used to predict moisture content upon the knowledge of $|\Gamma|$.

The relationships between $|Z_L|$, $|\Gamma|$, and moisture content.

![Fig. 7. The comparison between measured $|\Gamma|$ with calculated results obtained from finite difference method (FDM), admittance model and capacitance model at 2 GHz.](image)

![Fig. 8. The relationships between $|Z_L|$, $|\Gamma|$, and moisture content.](image)
Figure 11 represents region 2 in Fig. 7. The range 40% to 90% moisture content is within week 12 to week 16 after anthesis. It can be observed in Fig. 2 as well. The water content and oil content change drastically during week 16 to week 17. The water content starts to decrease, whereas oil content starts to rise on week 16 after anthesis. This is difficult to predict because the moisture content has an abrupt change. Hence, it can be observed that the error of FDM, admittance model and capacitance model are larger than the case in region 1 (Fig. 9), namely 0.06, 0.11 and 0.10. When FDM, admittance model and capacitance model are compared to each other, it can be found that FDM shows the best agreement with measured data with the smallest error, 0.06 during week 12 to 17 after anthesis. The fitted line of measure data is best represented as quadratic equation $|\Gamma| = (6 \times 10^{-6}) (mc)^2 - 0.004 (mc) + 1.111$. Hence, the sensitivity equation [25] can be represented by $\frac{d|\Gamma|}{d(mc)} = (1.2 \times 10^{-7}) (mc) - 0.004$ and it is visualized in Fig. 12. Even though the sensitivity decreases when the moisture contents increases from 40% to 90%, however, the variation of sensitivity with mc is not drastic. Although the sensitivity decreases, it is still greater than the sensitivity in region 1 as shown in Fig. 10. Overall, the sensor has higher sensitivity $\frac{d|\Gamma|}{d(mc)}$ for moisture content greater than 40% (region 2) if compared with Fig. 10 and it is commendable as this coincides with the drastic change in moisture content from unripe fruits to the ripe stage.

**B. Phase of reflection coefficient**

The variation of phase with moisture content, $mc$, is shown in Fig. 13. Phase is highly influenced by the complex permittivity, $\varepsilon^*$ and the thickness of the sample.
In addition, the length of the coaxial line and the thickness of the fruit can cause a phase shift. The phase shift in Fig. 13 shows good agreement with the measured data when compared with the phase on plane AB. When the length of coaxial line varies from 0.5 cm to 10 cm, it can be observed that the error in FDM shows the smallest when the length is 6 cm as seen in Fig. 14. However, the measured length of coaxial sensor from the caliper shows 5.655 cm. This deviation may be due to the inhomogeneity of the fruit in terms of permittivity. The FDM results deviated from the measured data because FDM only considers a homogeneous sample calculation. The phase of reflection coefficient from FDM still shows the best results for 6 cm coaxial sensor when compared to admittance model and capacitance model which have extended to plane AB as well by using technique of de-embedding of coaxial probe [27]. The effects of length of the open-ended, coaxial sensor towards reflection coefficient in reflection measurement had been reported [28]. Error shown by FDM is 45.6 degrees on the plane CD. After the plane is extended from plane CD to AB, the error is reduced to 7.80 degrees with similar condition. It is expected that the measurement plane must coincide with the calibration plane, since the calibration is done on plane AB. The mean phase error of admittance model (25.0 degrees of mean error) and capacitance model (27.1 degrees of mean error) are higher at plane CD if compared with the mean phase error at plane AB. After the plane CD is extended to plane AB, the error of admittance model is reduced to 17.3 degrees, while capacitance model is reduced to 15.0 degrees. After the comparison was done, the FDM on plane AB shows the best agreement with measured data. The poor accuracy in admittance model and capacitance model are due to the assumption made in both models. As mentioned previously, in both of these models, it was assumed that the thickness of sample of fruit was infinite. Therefore, neither of these two models is suitable for characterizing thin sample or any sample with finite thickness, such as the oil palm fruit. Figure 15 represents region 1 in Fig. 13. In Fig. 9, the magnitude of measured data, admittance model and capacitance model shows insignificant change with moisture content range 20% to 40%. The phase of admittance model and capacitance model for moisture content between 20% and 40% is almost constant as shown in Fig. 15. However, the measured phase decreases with equation \( \phi = 0.0117(mc) \cdot 1.2846(mc) - 30.665 \). The mean phase error for FDM, admittance model and capacitance model on plane AB are 3.59 degrees, 4.27 degrees and 4.44 degrees respectively. They have the mean phase error that is close to the fitting line which shows 3.13 degrees of mean error. Comparing FDM with admittance model and capacitance model, FDM has better agreement with the measured phase. The FDM, admittance model and capacitance model on plane CD (measurement plane) show a larger mean phase error if compared with the models on plane AB. It is due to the measured magnitude are collected on calibration plane but not the measurement plane. The differentiation of the fitting line equation with moisture content is:

\[
\frac{d\phi}{d(mc)} = 0.0234(mc) - 1.2846,
\]

or so-called sensitivity for moisture content range 20% to 40% as shown in Fig. 16. It is dissimilar to Fig. 10 because of the sensitivity in Fig. 16 decreases from 0.3 to 0.8 for moisture content range 20% to 40%, however, the sensitivity in Fig. 16 decreases insignificantly from 0.0035 to 0.0029. For this reason, the measured phase has higher sensitivity than the measured magnitude in region 1.

![Fig. 13. Comparison of phase of reflection coefficient among measured data, FDM, admittance model and capacitance model.](image-url)
Figure 17 represents region 2 in Fig. 13. The mc range 40% to 90% is within week 12 to week 16 after anthesis. During this period, the condition in Fig. 17 is similar to Fig. 11 because they show a similar trend. The water content starts to decrease, whereas oil content starts to rise in week 16 after anthesis. It can be observed that the error of FDM, admittance model and capacitance model are larger than the case in Fig. 15, namely 8.51 degrees, 22.29 degrees and 18.97 degrees. In Fig. 15, FDM still shows the best agreement with measured data. It has the smallest error compared with admittance model and capacitance model. Unlike the case in Fig. 15, the results of admittance model and capacitance model deviated from measured phase in mc range from 40% to 90%. The admittance model and capacitance model have larger mean phase error, namely 22.29 degrees and 18.97 degrees, respectively. The fitting equation that represents the trend of measured phase is \( \phi = 0.0015(mc)^2 - 0.9145(mc) - 27.928 \) as shown in Fig. 17, while the relationship between sensitivity and mc is \( |d\phi/d(mc)| = (0.0030)(mc) - 0.9145 \).
shows higher than the magnitude in region 2 (Fig. 17). This implies that a small change in moisture content can be easily detected by the phase of reflection coefficient when compared with magnitude of reflection coefficient. This can help to estimate the moisture content accurately [23].

In region 2 where $mc > 40\%$, the sensitivity is higher than region 1. The sensitivity can be expressed as

$$\left| \frac{d \phi}{d(mc)} \right| = -0.0245(mc) + 2.25$$

(Fig. 18). It is in line with the response of moisture content to the weeks after anthesis as shown in Fig. 2, where the variation of $mc$ becomes drastic when $mc > 40\%$.

![Fig. 18. Sensitivity, $\left| \frac{d \phi}{d(mc)} \right|$ for $mc$ range that exceeds 40%.

IV. CONCLUSION

In this work, the complex reflection coefficient was analyzed computationally with FDM on an aperture coaxial sensor. The accuracy of this analysis was investigated by comparing calculated (FDM, the admittance model and the capacitance model) with measured reflection coefficients (measured using a Vector Network Analyzer). Figures 7 and 13 indicate that the FDM was more accurate than the admittance model and the capacitance model.

REFERENCES


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Design of PSR with Different Feed Configurations and Partition Lens System for Skin Cancer Treatment

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Abstract — The Prolate Spheroidal Impulse Radiating Antenna (PSIRA) focuses pulsed radiation in the near field with a small beam width allowing the application of non-invasive skin cancer treatment. In this proposed work, SWB (Slanted Wire Bicone), EPH (Elliptical Profile Horn), TSVS (Tapered Slot Vivaldi Shape) and Tapered Arm Conical Plate (TACP) feed antenna configurations for Prolate Spheroidal Reflector (PSR) are explored to enhance the spatial resolution on biological target. The feed antenna is placed at the first focal point and the target is located at the second focal point of the PSR. Next, the resolution of focused electric field on the target is enhanced by using near field focusing lens. A 3 layer partition lens system is placed before the target to reduce the spot size of the focused field on the target. The delivery of subnanosecond pulses using reflector in conjunction with and without partition lens system on the biological target is compared for all feed antenna configurations. Tapered Arm Conical Plate (TACP) fed PSR with the 3 layer partition lens system greatly reduces the spot size to 0.5 cm along lateral direction and 1 cm along axial direction. The enhancement in spatial resolution is much favorable to reduce the damage of healthy tissues during cancer treatment.

Index Terms — Near field focusing lens, non-invasive cancer treatment, prolate spheroidal impulse radiating antenna, subnanosecond electric pulse, ultra wide band antenna.

I. INTRODUCTION

Recent research shows that fast, intense electromagnetic pulses can be used to kill melanoma cells [1, 2]. The conventional treatment for melanoma is an invasive method which delivers electric pulses in nanosecond range using electrodes into the skin. The non-invasive technique involves PSIRA to deliver an electric pulse. The non-invasive treatment limits surgery, reduces the pain, scarring and mortality of the patients while remaining cost effective and safe. From the past researches, it is observed that if an intense electric field is applied to cancer cells, it introduces a programmed cell death called apoptosis [3].

Xiao et al. showcased the high power electromagnetic pulses can treat melanoma cells [4, 5]. The intense pulses with subnanosecond range is preferred to obtain the higher probability of penetration into the interior of the cells. The preferred duration of subnanosecond pulse is in the range of 100-200 ps making it possible to focus the radiation on the target efficiently and produce small spot size on the target. Baum et al. designed PSIRA to deliver subnanosecond pulse. The PSIRA acts as a high power radiation source for melanoma treatment [6]. The PSIRA is based on the two foci of an ellipse. It radiates subnanosecond intense pulses from the first focus to the second focus. The feed of the PSIRA is an ultra-wideband antenna. The spatial resolution on the target is enhanced by placing lens system before the target. Kumar et al. designed a graded index lens system which is located before the dielectric slab (ɛ_r = 9). The spot diameter obtained from the numerical simulation is 1.187 cm [7].

In the proposed work, four types of PSIRA with partition lens are designed and their performance is compared in terms of spatial resolution on the target.

This paper is organized as follows. Section II presents the design of PSR. In Section III, the different feed configuration for PSR is discussed. Section IV discusses the partition lens system design. The skin model is presented in Section V. Section VI presents the comparative results of the numerical simulations of the PSIRA and focusing lens system.

II. PROLATE SPHEROIDAL REFLECTOR DESIGN

The Impulse Radiating Antenna (IRA) is a class of focused aperture antennas that have been used extensively for the generation and radiation of ultra wide band electromagnetic pulses [8, 9]. The IRA can effectively focus its radiation in the near field by using Prolate Spheroidal Reflector (PSR). A fast rising transient pulse is propagated from the first focus of the PSR. The wave is then refocused on the second focus of the PSR to obtain a very narrow impulsive waveform.

The PSIRA consists of two main parts, feed arms and a reflector. The schematic diagram of Prolate
Spheroidal Reflector is shown in Fig. 1. PSR is an elliptical reflector. It has two focal points. The feed antenna is placed at the first focal point. The target (skin) is placed at the second focal point. The radiated electromagnetic wave from the first focal point is concentrated at the second focal point where the target is placed. PSR is designed using the reflector geometry proposed by Xiao et al. [6]. The semi major axis \( a \) of the reflector is 29.8 cm. The semi minor axis \( b \) is 25 cm. It has two foci. The focal distance \( z_0 = \sqrt{a^2 - b^2} \), i.e., 16.3 cm. The second focal point is closer to the near field region with focal distance less than \( 2D^2/\lambda \), where \( D \) is aperture diameter and \( \lambda \) is the wavelength. The distance between the two focal point is 32.6 cm.

\[ x^2/a^2 + (y-h)^2/b^2 = 1. \]  
\[ m = e^2 = 1 - b^2/a^2, \]

e-eccentricity of the ellipse.

The length of the arc described by the ellipse is given by \( E(m) \), the complete elliptic integral of the first kind,

\[ E(m) = \int_0^{\pi/2} \sqrt{1 - m \sin^2 \theta} \, d\theta. \]

The simulation setup is shown in Fig. 2 (b). The semi major axis \( (a) \) of elliptical profile is 15 cm and semi minor axis \( (b) \) is 16 cm. At the excitation point the spacing between the plates is properly designed in order to avoid the higher order modes. For the design of horn structure the higher order modes are undesirable. So the feed gap has to be less than the wavelength at the highest operating frequency. The edges of the plates are tapered. The curvature of the tapering is optimized to completely remove the fluctuations in the main lobe at higher frequencies.

III. DESIGN OF FEED ANTENNA CONFIGURATIONS OF PSR

The radiation characteristics of IRA at high frequencies are very sensitive to the design of feeding structure. The design of different feed antenna structures for PSR is presented in this section.

A. Slanted Wire Bicone (SWB) fed PSR

The Slanted Wire Bicone (SWB) fed PSIRA design is presented. It is much suitable for ultra wide band application such as intense subnanosecond pulse radiation. The design of SWB is derived from bicone antenna [10]. The radiation is focused in one direction by slanting the two cones in the desired direction by the angle of 30°, which is shown in Fig. 2 (a). In order to construct a light weight feed structure, the wire bicone is chosen instead of solid structure. The inner diameter of conical structure is 2 cm and the outer diameter is 22 cm. The excitation gap between the cone is optimized to 1 cm. The entire simulation setup is shown in Fig. 2 (b).

B. Elliptical Profile Horn (EPH) fed PSR

The design of Elliptic Profile Horn (EPH) feed for PSR is presented. The EPH feed is simple and is designed from the single quadrant of the elliptical profile, which is shown in Fig. 3 (a). In this work the longitudinal section of the ellipse in the y-z plane is chosen, because it is easy to bend the surface smoothly at both the low impedance (driving point) as well as free space ends of the horn [11].

Two significant variables in the design of the elliptical profile horn are overall length \( a \) (semi major axis of the ellipse) and the height \( h \) (separation between the plate which is determined from the semi minor axis \( (b) \)).

The coordinates of the ellipse can be obtained from:

\[ x^2/a^2 + (y-h)^2/b^2 = 1. \]  
\[ m = e^2 = 1 - b^2/a^2, \]

e-eccentricity of the ellipse.

The length of the arc described by the ellipse is given by \( E(m) \), the complete elliptic integral of the first kind,

\[ E(m) = \int_0^{\pi/2} \sqrt{1 - m \sin^2 \theta} \, d\theta. \]
C. Tapered Slot Vivaldi Shaped (TSVS) fed PSR

The traditional Vivaldi feed arm structure for parabolic reflector has a low reflection loss in the entire frequency band but low radiation efficiency. The Tapered Slot Vivaldi Shaped (TSVS) feed arm shown in Fig. 4 (a) is designed in order to improve the radiation characteristics of the PSIRA. The Prolate Spheroidal Reflector with Tapered Slot Vivaldi Shaped feed arm is shown in Fig. 4 (b). It consists of two pairs of feed arms. The feed arms are separated by the arm angle $\varphi=70^\circ$ from the horizontal axis. The optimized arm angle ($\varphi=70^\circ$) is used because the antenna has uniform aperture field distribution in the entire frequency band of interest [12].

The TSVS feed is a combination of Vivaldi antenna and TEM coplanar transmission line. The Vivaldi shape is obtained by blending the edge of the coplanar transmission line with the radius($r$) of 7 cm. The rounded portion of the TSVS feed is optimized to reduce the current reflection from the end of the antenna and to improve the radiation of low frequencies.

Two parts of the Vivaldi shape are joined at one end of the arm and the other end of the arm is arranged in such a way that it looks like a linearly tapered slot. The length of the slot, $L=17.6$ cm. The width of the tapered slot, $W=4$ cm. The thickness of the arm is considered as 1 mm. The pair of two feed arms is placed at the first focal point of the PSR. The upper and lower feed arms are separated by the optimized feed gap of 1 cm.

D. Tapered Arm Conical Plate (TACP) fed PSR

The TACP fed PSR is constructed with a diameter of 50 cm and four tapered conical plate feed arm with separation angle of $\varphi_o=60^\circ$. The design of a conical structure mainly depends on the three angles $\beta_0$, $\beta_1$, and $\beta_2$. The center angle is $\beta_0$, and $\beta_1$ and $\beta_2$ are the lower and upper side angle of the feed arm, which is shown in Fig. 5 (a). The conically symmetric TEM feeding structure parameters [$\beta_0$, $\beta_1$, $\beta_2$] are related to the equivalent longitudinally symmetric structure parameters (semi major axis $a$, semi minor axis $b$, diameter $D$, focal length $F$). The center angle is specified as $\beta_0$:

$$\beta_0 = \arctan \left( \frac{1}{\frac{2F}{D} - \frac{D}{2F}} \right).$$  (4)

$$\beta_1 = 2 \arctan \left( \frac{m^{1/4}}{\tan \left( \frac{\beta_0}{2} \right)} \right),$$  (5)

$$\beta_2 = 2 \arctan \left( \frac{m^{1/4}}{\tan \left( \frac{\beta_1}{2} \right)} \right).$$  (6)

$\beta_1$ and $\beta_2$ are the lower and upper side angle of the feed arm.

Figure 5 (b) shows the PSIRA with TACP feed. These arms are terminated by 200 $\Omega$ load which are used as a low frequency matching circuit. Different termination loads have been tried [13] and it has been observed that there is no resistive termination load that can match the feeding arms with the reflector to remove the standing waves. It means that there is some stored energy around the arm-reflector junction. Because of that stored energy at the junction, the observed impedance is a complex value. It is not easy to match the feeding arms with the reflector for the entire frequency band. In order to reduce the energy stored around the junction of the arm-reflector and to improve the gain of the IRA, the end part of the feeding arms are tapered.
E. Impedance and radiation characteristics of different feed configurations

The impedance and radiation characteristics of different feed configurations are presented. The spot size is one of the most important parameters to be considered for non-invasive cancer treatment. The spot is measured from the focal waveform at half power points.

The return loss and VSWR are calculated from Figs. 6 (a) and 7 (a) for SWB fed PSR. The return loss is -10 dB for the bandwidth of 600 MHz to 10 GHz. Within the band of 600 MHz to 10 GHz, the VSWR is lower than 2. The radiated pulse is focused to the second focal point where the target (skin) is placed. The focal waveform is shown in Fig. 8 (a). The maximum electric field measured is 49 V/m. The Full Width Half Maximum (FWHM) is 65 ps.

The impedance characteristics of EPH fed PSIRA are shown in Figs. 6 (b) and 7 (b). The return loss is -10 dB for the bandwidth of 430 MHz to 15 GHz. The maximum electric field near second focus is measured as 100 V/m. The focal waveform is shifted towards the aperture of the reflector which is shown in Fig. 8 (b). The FWHM calculated from the focal waveform is 60 ps.

In TSVS fed PSR, the return loss measured is -10 dB for the bandwidth of 500 MHz to 10 GHz from Fig. 6 (c). The return loss of -5 dB is obtained for the spectrum of 350 MHz to 10 GHz. Within the band of 500 MHz to 9.5 GHz, the VSWR is lower than 2 which is better than the case of traditional IRA, which is shown in Fig. 7 (c). The spot size measured from the focal wave form (Fig. 8 (c)) is 90 ps.

The return loss and VSWR are measured from Figs. 6 (d) and 7 (d) for TACP fed PSR. The feed arm is well matched for the bandwidth of 1.5 GHz to 10 GHz. The FWHM measured from focal waveform Fig. 8 (d) is 100 ps, which is the same as the rise time of the input signal.

IV. DESIGN OF PARTITION LENS

The design of the 3 layer partition lens is presented. The fast and intense electromagnetic pulse is illuminated to the target (skin) which is located at the second focal point. The major problem for concentrating the fields at the second focal point is the reflection. The dielectric property of the target medium and the medium though, which the incident wave propagates are different which causes reflections. The partition lens is used to focus the field at the second focal point. The addition of partition lens before the target leads to better match the wave to the target. The larger electric field and reduced spot size are obtained.

A. Partition lens design

The design of a partition dielectric lens is shown in Fig. 9. The partition lens is designed based on Fermat’s Aplanatic principle and Fresnel law of refraction [14, 15]. The design contour of each layer is determined from the following equation:

\[ y_i = f_i(x) = \sqrt{(n^2 - 1)x^2 - 2n(n-1)Fx + (n-1)^2F^2}, \tag{7} \]

where \( n \) is the refractive index of the lens, \( F \) is the focal length.
The thickness of the lens is varied from \( t_i \) to \( t_i \). The relative dielectric constant \( \varepsilon_r = 2 \) for all three layers. In the lens design the phase difference between the electromagnetic waves plays an important role. The layers of the lens are designed with different focal length and different optical path. The electromagnetic waves through different parts of the lens have different phase. The partition lens is designed such that the phase difference between the electromagnetic waves through the different parts of the lens is an integer multiple of \( 2\pi \) till they arrive at the focal point.

\[ L_i = F_i + n \times t_i, \quad (8) \]

\( F_i + t_i = \text{Constant}, n \) is the refractive index of the lens, \( F_i \) is the focal length of the \( i^{th} \) layer, \( t_i = \) thickness of the \( i^{th} \) layer.

The appropriate focal length for each layer is chosen to make sure that the difference of optical length between each layer is an integer multiple of the wavelength (3 cm). The design parameters are shown in Table 1 for 3 layers. The inner and outer radius is used to decide the area of each layer. This will have an effect on the focusing ability of the partition lens. The area of the partition is optimised in order to obtain reduced spot size and enhanced electric field at the second focus. The intensity of the wave decreases from the reflector and increases at the geometric focus. Adding partition lens allows one to increase the field intensity near the geometric focus, but the strongest intensity is still found along the axis and increases as the wave penetrates deeper into the tissue.

The proposed lens is used to solve the thickness and volume problem of the short focus dielectric lens. The field at the second focus is higher because the phase difference between electromagnetic waves through the different part of the lens is chosen as integer multiples of \( 2\pi \).

### Table 1: Design parameter for partition lens

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Layer-1</th>
<th>Layer-2</th>
<th>Layer-3</th>
</tr>
</thead>
<tbody>
<tr>
<td>Dielectric constant ( (\varepsilon_r) )</td>
<td>2</td>
<td>2</td>
<td>2</td>
</tr>
<tr>
<td>Focal length (cm)</td>
<td>9</td>
<td>6</td>
<td>3</td>
</tr>
<tr>
<td>Outer radius (cm)</td>
<td>12</td>
<td>13.5</td>
<td>15</td>
</tr>
<tr>
<td>Inner radius (cm)</td>
<td>0</td>
<td>12</td>
<td>13.5</td>
</tr>
</tbody>
</table>

### V. NUMERICAL SIMULATIONS AND DISCUSSION

#### A. Electromagnetic simulation

The entire simulation setup is shown in Figs. 10-13. The CST Microwave Studio, 3-D Finite Integration Time Domain (FDTD), commercially available software is used for simulation. The CST Microwave Studio is a module in CST which is dedicated to fast and very accurate electromagnetic simulations of high frequency problems. This module contains different solvers for simulations of structures, both in time and frequency domain. The CST transient solver is suitable for wide band antenna simulations and electrically large structures. The simulation code is computed using CST transient solver. The PSIRA with the partition lens system and skin model for four types of configuration is shown in Figs. 10-13. The three layer focusing lens has its center at the second focal point. The radius and focal length is obtained from Table 1. The two layer skin model is used as target medium. The reflector and feed arms are assumed to be perfect electric conductors, the focusing lens and skin model are assumed to be lossless and dispersion less. The PSIRA is fed with the input of 1 V, 100ps rise time Gaussian signal.

#### Table 2: Dielectric properties of skin model

<table>
<thead>
<tr>
<th>Properties</th>
<th>Skin</th>
</tr>
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<tbody>
<tr>
<td>Dielectric constant</td>
<td>38</td>
</tr>
<tr>
<td>Conductivity (S/m)</td>
<td>0.2</td>
</tr>
<tr>
<td>Material density (Kg/m³)</td>
<td>1100</td>
</tr>
<tr>
<td>Thermal conductivity (W/K/m)</td>
<td>0.5</td>
</tr>
<tr>
<td>Heat capacity (KJ/K/kg)</td>
<td>3.35</td>
</tr>
</tbody>
</table>

Fig. 9. Design of partition lens from hemisphere dielectric lens.

Fig. 10. Simulation design of SWB fed PSIRA with partition lens system and skin model.
Fig. 11. Simulation structure of EPH fed PSIRA with partition lens system and skin model.

Fig. 12. Simulation setup of TSVS fed PSIRA with partition lens system and skin model.

Fig. 13. Simulation design of TACP fed PSIRA with partition lens system and skin model.

B. Comparative spot size analysis

The spot size of the radiated pulse at the second focus plays a vital role in the treatment of skin cancer. The enhanced spatial resolution on the target reduces the damage of healthy tissue. The electric field component that contributes mostly to the electric field at the second focal point is the Y component. The amplitude of the electric field along X and Z directions are negligible. The electric field distribution is measured around the second focal point by placing the probe at regular spacing where the target is located.

This section presents the numerical simulation results of different feed antenna configurations for PSR with focusing lens. The electric field distribution is measured along lateral as well axial direction. The FWHM is obtained from the electric field distribution curve which is shown in Figs. 14-17 for all feed configurations. The beam width of the radiated electric field with and without lens is compared. The field intensity is maximum at the focal point and it is reduced to 50% with ±1 cm near the focus for SWB, and TSVS fed PSR. In EPH fed PSR without lens configuration, the radiated impulse is not exactly focused on the second focal point. The peak of the impulse electric field on the z axis is shifted slightly from the geometric focus towards the reflector, which is observed in the modeled result (Fig. 15(a)). This is because the impulse decreases inversely with the distance while it is focused in space. At the focal point, even though coherent combinations of waves are obtained, the impulse electric field is still smaller than the nearby locations toward the reflector due to a large impulse width, cδ (in spatial units). In order that the maximum impulse amplitude occurs at the geometric focus, the impulse width needs to be small compared to both 2z0 and 2b (similar discussion in the frequency domain can be seen in [23]). This shift of the impulse electric field is overcome by using partition lens. A pulse with faster rise time is allowed the shift of the focal spot towards the geometric focus. In TACP fed PSR maximum field intensity is obtained at the second focal point and it is reduced to 50% within ±0.25 cm near the focal point.

Fig. 14. Electric field distribution of SWB fed PSR at the second focal point: (a) without lens and (b) with partition lens.

Fig. 15. Electric field distribution of EPH fed PSR at the second focal point: (a) without lens and (b) with partition lens.
Fig. 16. Electric field distribution of TSVS fed PSR at the second focal point: (a) without lens and (b) with partition lens.

Fig. 17. Electric field distribution of TACP fed PSR at the second focal point: (a) without lens and (b) with partition lens.

Table 3 summarizes the radiation characteristics of all types of radiators. Their performance is compared in terms of electric field and spot size on the target.

Table 3: Comparison of all types of PSIRA with and without partition lens system

<table>
<thead>
<tr>
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<th></th>
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<tbody>
<tr>
<td>Reflector diameter (cm)</td>
<td>50</td>
<td>50</td>
<td>50</td>
<td>50</td>
<td>50</td>
<td>50</td>
<td>50</td>
<td>50</td>
</tr>
<tr>
<td>F/D</td>
<td>0.33</td>
<td>0.33</td>
<td>0.33</td>
<td>0.33</td>
<td>0.33</td>
<td>0.33</td>
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<td>No. of arms</td>
<td>-</td>
<td>-</td>
<td>4</td>
<td>4</td>
<td>-</td>
<td>-</td>
<td>4</td>
<td>4</td>
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<tr>
<td>Separation between arms</td>
<td>-</td>
<td>-</td>
<td>70°</td>
<td>60°</td>
<td>-</td>
<td>-</td>
<td>70°</td>
<td>60°</td>
</tr>
<tr>
<td>Input Impedance (Ω)</td>
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<td>200</td>
<td>200</td>
<td>100</td>
<td>100</td>
<td>200</td>
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<td></td>
</tr>
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<td>Axial (cm)</td>
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<td>13</td>
<td>14</td>
<td>5</td>
<td>8</td>
<td>7</td>
<td>2.5</td>
<td>1</td>
</tr>
<tr>
<td>Lateral (cm)</td>
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<td>7.5</td>
<td>8</td>
<td>1</td>
<td>2</td>
<td>1</td>
<td>0.75</td>
<td>0.5</td>
</tr>
</tbody>
</table>

VII. CONCLUSION

The Prolate Spheroidal Reflector (PSR) with different feed antenna configurations are designed to delivered subnanosecond pulses for non-invasive skin.
cancer treatment. The 100ps electric pulse is launched from the feed antenna and is focused at the second focal point where the target is placed. Further, deep near field focusing is obtained using partition lens. The lens modifies the electric field distribution on the target along lateral as well as axial direction. The radiation characteristics of all PSIRA configurations with and without lens system are compared. The TACP fed PSR with partition lens system produces narrow spot size of 0.5 cm along lateral direction and 1 cm along axial direction. The enhanced resolution is obtained on the target using TACP fed PSR with partition lens system. Adding a dielectric lens allows one to increase the field intensity near the geometric focus which is more beneficial to reduce the input voltage requirement to obtain high electric field and to enhance the spatial resolution for the treatment of skin cancer.

REFERENCES


Design, Simulation and Fabrication of a Wide Bandwidth Envelope Tracking Power Amplifier

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Abstract — In this paper an envelope tracking power amplifier is designed and implemented using MRF6S27015N MOTOROLA transistor in LDMOS technology. First, the amplifier is designed using load pull simulation and its parameters are optimized to increase the power added efficiency. Next, envelope detector and envelope amplifier are designed and simulated and finally ET is applied to the amplifier which results in more than 50% of PAE in a wide range of input power with bandwidth of 200 MHz. Envelope detector circuit is fabricated using schotkey diode and envelope amplifier is manufactured using a mosfet, an op amp, and a comparator.

Index Terms — Envelope amplifier, envelope elimination and restoration, power amplifier.

I. INTRODUCTION

Power amplifier is the most important part in a communication system because it uses the most amount of energy in the system. So, good efficiency of power amplifier can decrease amount of heat generated, and therefore boost performance of the total system. Back off may be used to maintain linearity of the power amplifier, but this may result in decreasing the efficiency. Some techniques are proposed to overcome this problem like LINC, ΣΔ modulation, Doherty, and EER Polar modulator. The first two techniques show high linearity with a moderate efficiency [1, 2] and the latter two techniques show high efficiency with a moderate linearity characteristic [3, 4]. However, applying the digital predistortion (DPD) technique to the latter two techniques can enhance the moderate linearity adequately. Envelope tracking power amplifiers are used to increase efficiency of the power amplifier [5-10]. ET uses a linear PA and a controlled supply voltage, which tracks the input envelope. When the supply voltage tracks the instantaneous envelope modulation signal, it is called Wide Bandwidth ET (WBET) [11]; when the supply voltage tracks the long-term average of the input envelope power, it is called Average ET (AET) [12]; when the supply voltage switches to different step levels according to the input envelope power, it is called Step ET (SET) [13]. EER (envelope elimination and restoration) uses a combination of a high efficiency switch-mode PA with an envelope re-modulation circuit [14]. To employ high efficiency operation of EER and at the same time reduce the strict necessities of bandwidth and time-alignment, the “hybrid” EER structure was proposed in [15]. Total system efficiency is determined by the product of the envelope amplifier efficiency and the RF transistor drain/collector efficiency. As a result, a high-efficiency envelope amplifier is vital for the EER/ET system. High efficiency envelope amplifier is usually realized by a DC/DC converter, where the switching frequency is required to be several times the signal bandwidth. For narrow bandwidth applications, most high efficiency switching mode DC/DC converters are realized by traditional delta modulation [16] or pulse width modulation (PWM) [17] modulators. Envelope amplifier block diagram is shown in Fig. 1, which consists of an OPAMP to realize the voltage source, and a MOSFET to realize the current source. This envelope amplifier consists of a voltage source and a current source. Although the current source has high efficiency, the voltage source has low efficiency. Therefore, we like to get most of the output current from the current source. To control the current of the voltage source, we use a hysteretic current feedback control to realize the soft power division between switch stage and linear stage amplification. The load voltage is controlled by the linear voltage source, and the load current is a mixture of the linear stage current and the switch stage current.

II. ET AMPLIFIER DESIGN

A. RF amplifier design

Si-LDMOS is a popular device choice for base-station high-power amplifiers, since LDMOS technology can provide reliable and cost effective solutions [18]. Envelope tracking techniques, in which a wideband envelope amplifier makes variable supply biasing to the RF stage, have established excellent performance using
a variety of device technologies including Si LDMOS and GaN FETs [19]. Nonetheless, the RF PA should have suitable characteristics to be appropriate for the envelope tracking operation and achieve the optimal performance such as low deviation of the output capacitance, since a large variation of the voltage dependent output capacitance will corrupt the average efficiency as the optimum impedance matching for the output of the PA changes with the supply voltage [17].

Besides, extra nonlinearity like AM-PM distortion from the nonlinear capacitance and AM-AM distortion from the envelope amplifier, and memory effects due to the limited bandwidth of the RF PA and the envelope amplifier is produced by the dynamic supply biasing. However, digital pre-distortion (DPD) techniques may be used to correct the nonlinearity of the dynamically biased amplifier. In this paper, the amplifier is designed and implemented using the MRF6S27015N MOTOROLA transistor in LDMOS technology and is simulated using ADS2008 software where nonlinear analysis is performed using harmonic balance method [20]. The design of the amplifier is at the central frequency of 2.1 GHz and the bandwidth of the RF amplifier is 200 MHz, which is high with respect to the other works. The amplifier is biased in class AB. Output current diagrams of the transistor is used to choose appropriate Vgs, which can be selected between 2.3 volt and 3.8 volt to work in class AB and is optimized to increase the efficiency.

Load pull simulation is used to select the optimum output impedance seen from the output of the transistor that increases the efficiency which is $5.7 + j12.6$. Power added efficiency and gain of the transistor when it sees output impedance of $5.7 + j12.6$, is shown in Fig. 2. Our aim in design of the power amplifier is to increase the efficiency. So, we optimized the amplifier to reach our goal and we changed Vgs, Vds, width and length of the matching transmission lines to reach to a good efficiency. Also, matching of the transistor is optimized in ADS software. Table 1 shows line width and line length of the optimized transmission lines. RO4003 substrate is used with $\varepsilon_r = 3.5$ and $\tan \delta = .0027$. Input matching and output matching schematics are shown in Fig. 3 and Fig. 4. Optimising the matching circuit in ADS software, input and output return losses are shown in Fig. 5 and Fig. 6.
Fig. 6. Output matching.

Increasing the output power of the transistor will increase the PAE. However, our input signal has a high peak to average ratio and probability of the peak power is very low. Consequently, it is wise to design the matching for the case that happens most of the time, which is mean power. Therefore, because Vds changes with respect to the input power, we optimized the power amplifier for the Vds related to the mean power.

Table 1: Size of the matching transmission lines

<table>
<thead>
<tr>
<th>Line Number</th>
<th>Line Length (mil)</th>
<th>Line Width (mil)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Z1</td>
<td>940</td>
<td>25</td>
</tr>
<tr>
<td>Z2</td>
<td>360</td>
<td>85</td>
</tr>
<tr>
<td>Z3</td>
<td>170</td>
<td>145</td>
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<tr>
<td>Z4</td>
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<td>85</td>
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<td>Z5</td>
<td>370</td>
<td>800</td>
</tr>
<tr>
<td>Z6</td>
<td>136</td>
<td>800</td>
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<td>Z7</td>
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<td>20</td>
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<td>Z8</td>
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<td>Z9</td>
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<td>900</td>
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<tr>
<td>Z10</td>
<td>100</td>
<td>805</td>
</tr>
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<td>Z11</td>
<td>200</td>
<td>805</td>
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<td>Z12</td>
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<td>Z15</td>
<td>705</td>
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<td>100</td>
<td>44</td>
</tr>
<tr>
<td>Z18</td>
<td>900</td>
<td>20</td>
</tr>
</tbody>
</table>

B. Changing Vds

In the next step, we changed Vds of the power amplifier from 8v up to 28v by 4v to show the possibility of increase in PAE by envelope tracking. Power added efficiency and output power of the RF power versus input power at center frequency of 2.1 GHz by changing Vds from 8v up to 28v by 4v is shown in Fig. 7 and Fig. 8 respectively. When the input power is 10 dBm and Vds is 8v, this simulation shows about 17% of more PAE than the condition of Vds=28v.

Figure 7 shows that we can increase P1dB of the transistor when the input power is large by applying ET.

Figure 7. PAE versus input power at center frequency by changing Vds.

C. Applying ET

When Vds is 8v, the PAE decreases in the peak power, and when Vds is 28v the PAE is low in low power. To overcome this problem, we can apply envelope tracking to our amplifier. By changing the Vds with respect to the input power, we can see that the PAE remains above 50% for a wide range of input power. Figure 9 shows PAE versus input power at center frequency by applying envelope tracking respectively.

Fig. 8. Output power versus input power at center frequency by changing Vds.

Fig. 9. PAE versus input power at center frequency by applying envelope tracking.
D. Envelope detector

Envelope detector is shown in Fig. 10, which consists of a Schottky diode and an LC circuit by \( L=190 \) (nH) and \( C=18 \) (pF). HSMS286K Schottky diode produced by Agilent is used. Envelope detector is simulated in ADS software by the signal shown in Fig. 11 and the output signal is obtained.

![Fig. 10. Envelope detector.](image1)

Fig. 10. Envelope detector.

![Fig. 11. Input and output signal of the envelope detector.](image2)

Fig. 11. Input and output signal of the envelope detector.

III. FABRICATION AND TEST

Manufactured circuit is shown in Fig. 12 which consists of 3 parts: 1) Wilkinson power divider, 2) envelope detector, and 3) transistor. In part 1 a Wilkinson power divider is used to divide the input signal to two equal parts. One part is fed to the transistor and one part is fed to an envelope detector circuit.

![Fig. 12. Manufactured circuit.](image3)

Fig. 12. Manufactured circuit.

Manufactured envelope amplifier circuit is shown in Fig. 13 and its block diagram is shown in Fig. 1, which consists of a mosfet, an op-amp, and a comparator where their part number is given in Table 2. Also, Table 3 compares previous works with this paper.

![Fig. 13. Manufactured envelope amplifier circuit.](image4)

Fig. 13. Manufactured envelope amplifier circuit.

Gain versus input power at center frequency by changing Vds, output power versus input power at center frequency by changing Vds, and PAE versus input power at center frequency by changing Vds are shown in Fig. 14 up to Fig. 16.

![Fig. 14. Gain versus input power at center frequency by changing Vds.](image5)

Fig. 14. Gain versus input power at center frequency by changing Vds.

<table>
<thead>
<tr>
<th>Table 2: Envelope amplifier</th>
</tr>
</thead>
<tbody>
<tr>
<td>Device</td>
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<tr>
<td>OP-AMP</td>
</tr>
<tr>
<td>MOSFET</td>
</tr>
<tr>
<td>COMPARATOR</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Table 3: Comparison to previous envelope amplifiers</th>
</tr>
</thead>
<tbody>
<tr>
<td>Power (dBm)</td>
</tr>
<tr>
<td>-------------</td>
</tr>
<tr>
<td>[4]</td>
</tr>
<tr>
<td>[5]</td>
</tr>
<tr>
<td>[6]</td>
</tr>
<tr>
<td>[7]</td>
</tr>
<tr>
<td>[8]</td>
</tr>
<tr>
<td>[9]</td>
</tr>
<tr>
<td>[10]</td>
</tr>
<tr>
<td>This work</td>
</tr>
</tbody>
</table>
IV. CONCLUSION

In this research an ET amplifier in the frequency range of 2 GHz to 2.2 GHz has been designed and fabricated using MRF6S27015N MOTOROLA transistor in LDMOS technology which has 10 watts output power. Also, implementation of the envelope amplifier and envelope detector has been described. It has been shown that the PAE remains above 50% for a wide range of input power for a bandwidth of 200 MHz.

REFERENCES


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Analysis of Control Variables to Maximize Output Power for Switched Reluctance Generators in Single Pulse Mode Operation

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Abstract — This paper presents an analytical modeling method of optimal control variables to maximize the output power for switched reluctance generators (SRGs) in single pulse mode operation. A method to obtain the phase current equation used to determine the optimal control variables is proposed. The phase current equation is derived from the phase voltage equation in combination with the inductance model. The inductance model proposed in this paper is applied from the flux linkage function. The characteristics of the phase current and the energy conversion relations are analyzed to determine the optimal phase current shape. The analytical results indicate that the optimal shape can be generated when the SRG is controlled with the optimal control variables. An 8/6 SRG experimental setup is used to validate the proposed method. The optimal control variables obtained from the proposed method are used to control the SRG. Based on the experimental results, the SRG can produce the maximum output power.

Index Terms — Control variables, optimal phase current shape, switched reluctance generator.

I. INTRODUCTION

A switched reluctance generator (SRG) is a potential candidate in various applications, such as an automotive starter/generator [1, 2] an engine starter/generator [3, 4] and for variable speed wind energy [5] because it has a simple structure and low cost, is fault tolerant with a rugged structure, and involves easy starting/generating realization, high speed adaptability, with a high generation efficiency.

Its highly nonlinear nature is the main problem of the SRG, since the behavior of the SRG cannot be described by mathematical equations using conventional methods for a suitable controller design [6]. The SRG model is used for simulation to determine the relationship between output power and control variables since there is no analytical equation with which to determine the output power based on design parameters and control variables [7]. The dynamic model of an SRG using a cubic spline technique has been proposed to find the flux linkage, inductance, torque, and output power [8]. A model SRG based on the Finite Element Method (FEM) and power control methods for a small wind power generation system has been proposed in [9]. The SRG model is used to find the output power curve versus the shaft speed. This curve is analyzed to determine the optimal switching on-off angle for maximum efficiency in the system.

The output power of the SRG in single pulse mode can be controlled by adjusting the excitation angles, with the turn-on/off angle being fixed while the turn-off/on angle is adjusted, or by adjusting both the turn-on and turn-off angles. Constant output power control of the SRG has been proposed by controlling the turn-on angle with a fuzzy logic algorithm while the turn-off angle is fixed [10]. The optimal excitation angles for output power control using automatic closed loop control have been proposed so that the optimal turn-off angle in terms of power and speed is determined from an analytical fit curve [11], while the optimal turn-on angle is automatically adjusted based on the closed loop power control to regulate the output power. The optimal excitation angles have been proposed for maximum system efficiency calculated using the ratio of the two flux linkages [12]. The minimum torque ripple occurs in this case. With two flux linkages, one is the position at which the stator and rotor pole corners begin overlap and the other is the position at maximum value. A Modified Angle Position Control (MAPC) method has been proposed to determine the optimal shape of the phase current [13]. The optimal turn-on angle is fixed and the optimal turn-off angle can be determined by the analytical model of the SRG for the maximum energy conversion [14, 15].

A mathematical model for analyzing control variables and describing the behavior of the SRG and the flux linkage versus current characteristics calculation is essential. There are at least two methods to obtain flux linkage versus current characteristics—an analytical approach based on the FEM and an experimental
approach based on direct measurements. The flux linkage model based on the FEM has well known reliability, however it requires intensive computation and many details of the machine geometry and structure [16]. Analytical nonlinear models of flux linkage have been described in [17-20] that are accurate and reliable. The model based on machine geometry introduced in [17] is complicated and depends on flux linkage at aligned andunaligned positions, and a position-dependent function. The position-dependent term has a physical significance in that its coefficient needs to be related to the machine geometry. The model proposed in [18] is a little complex because the flux linkage curve is divided into 2 parts, namely, linear and nonlinear. However, it only requires the flux linkage versus current characteristics at the aligned andunaligned positions. The model described in [19] based on a Fourier series with a limited number of terms is complex since it is necessary to know the flux linkage versus current characteristics at the aligned,unaligned, and midway positions. The coefficients in terms of the Fourier series depend on the flux linkage positions at aligned,unaligned, and midway positions so that the flux linkage at the aligned and midway positions can be calculated via curve fitting based on an arc-tangent function. The Stiebler model proposed [20] is simple in that it is composed of an angular function and aligned andunaligned flux linkage. However, it is proposed in a per-unit system.

An analytical modeling method of the optimal control variables to maximize the output power of the SRGs in single pulse mode operation is presented in this paper. The control variables comprise a dc bus voltage, a shaft speed or angular velocity, and excitation angles. This paper proposes a method to obtain the phase current equation used to determine the optimal control variables. The phase current equation is derived from the phase voltage equation in combination with the inductance equation used to determine the optimal control variables. The phase current equation is expressed as:

$$u = Ri + \frac{d\lambda(i, \theta)}{dt}.$$  \hspace{1cm} (1)

The voltage equation at constant speed is given by:

$$u = Ri + \frac{dL(i, \theta)}{dt} + e,$$  \hspace{1cm} (2)

where $u$ represents the dc bus voltage, $i$ is the phase current, $R$ and $L$ are the phase of resistance and inductance, respectively, $\omega$ is the angular velocity, and the back emf is defined as:

$$e = i\omega \frac{\partial L(i, \theta)}{\partial \theta}.$$  \hspace{1cm} (3)

The energy converted is the area enclosed by the loci which is expressed as:

$$W = \oint \lambda di = \oint id\lambda.$$  \hspace{1cm} (4)

The SRG requires an excitation source in order to generate electrical energy. The SRG (phase A) is excited by the asymmetrical bridge converter as shown in Fig. 2. This converter is used as the dc source [21] for the exciting phase A of the SRG through two switches as shown in Fig. 2 (bottom) and demagnetizing the same phase through two diodes as shown in Fig. 2 (top).

In Fig. 2, the current builds in the SRG phase winding when the controllable switches are closed and no energy is supplied to the load. When the controllable switches are opened, the stored energy is supplied to the load through the two diodes. The average load current can be defined as:

$$I_L = \frac{1}{2\pi/N_r} \left( \theta_e \int_{\theta_{on}}^{\theta_{off}} d\theta - \theta_{off} \int_{\theta_{on}}^{\theta_{on}} d\theta \right),$$  \hspace{1cm} (5)

where turn-on $\theta_{on}$ and turn-off $\theta_{off}$ angles represent the controllable switches which are closed and opened, respectively, $\theta_e$ is the angle at which the phase current is depleted and it is given as $2\theta_{off} - \theta_{on}$. $\theta$ is the rotor position and $N_r$ is the number of rotor poles.

II. PRINCIPLE OF SRG IN SINGLE PULSE MODE OPERATION

A 4-phase 8/6 SRG is used in this paper which is driven by a 4-phase asymmetrical bridge converter as shown in Fig. 1. When $S_A$ and $S_A'$ are both on, the phase A voltage is $u$. If $S_A$ and $S_A'$ are both off, the...
The electric average power of the SRG is the summation of the output power of each phase in one revolution which is given by:

$$P_{out} = I_u u.$$  \hspace{1cm} (6)  

The main electrical losses of an SRG are copper loss and iron loss. The copper loss $P_{Cu}$ depends on the rms phase current $I_{rms}$ on the range $\theta_{off} \leq \theta \leq \theta_e$ \[22\] which is expressed as:

$$P_{Cu} = N_{ph} I_{rms}^2 R,$$  \hspace{1cm} (7)  

and

$$I_{rms}^2 = \frac{1}{2\pi / N_r} \int_{\theta_{off}}^{\theta_e} i^2 d\theta.$$  \hspace{1cm} (8)  

The iron loss is in proportion to the excitation magnetic motive force and the stroke frequency. It is not uniformly distributed in the core since the flux shape is non-sinusoidal and the flux harmonic spectrum differs in various parts of the magnetic spectrum. The iron loss \[23\] can be approximately calculated as:

$$P_C = K_h f^a B_m^b + K_e f^2 B_m^2,$$  \hspace{1cm} (9)  

where $f$ is the stroke frequency, $K_h$ and $K_e$ are the hysteresis and eddy-current loss coefficients, respectively, $a$ and $b$ are the constants of the exponent, and $B_m$ is the amplitude of flux density for sinusoidal variation.

### III. ANGLE POSITION CONTROL METHOD

The control variables of the SRG are the dc bus voltage $u$, the angular velocity $\omega$, the phase current $i$, and turn-on/off angle $\theta_{on}$ and $\theta_{off}$. The Angle Position Control (APC) method can control the phase current shape by adjusting $\theta_{on}$ and $\theta_{off}$ while $u$ and $\omega$ are constant. The output power can be adjusted by the phase current. The advantages of the APC method \[13\] are that the optimal $\theta_{on}$ and $\theta_{off}$ can improve efficiency, the multiple phases can be conducted at the same time, and the torque adjustment range is wide.

The effect of $\theta_{on}$ and $\theta_{off}$ on the phase current shape using the APC method is illustrated in Fig. 3, where $\theta_{on}$ is fixed and $\theta_{off}$ is adjusted as shown in Fig. 3 (a) and $\theta_{on}$ is fixed and $\theta_{off}$ is adjusted as shown in Fig. 3 (b).

![Fig. 2. Power generation process for the SRG in single pulse mode.](image)

![Fig. 3. Phase current shapes using the angle position control method.](image)
4 is in the range \( \theta_{off} \leq \theta \leq \theta_p \) and \( \theta_p \) equals \((\beta_s + \beta_m)/2\), where \( \beta_s \) is the rotor pole arc and \( \beta_m \) is the stator pole arc. Considering (10), if the back emf is smaller than the dc bus voltage, then \( di/d\theta < 0 \). The phase current shape in this case is shown in Fig. 4 (a). If the back emf is equal to the dc bus voltage, then \( di/d\theta = 0 \). In this case, the phase current shape is shown in Fig. 4 (b). If the back emf is bigger than the dc bus voltage, then \( di/d\theta > 0 \) and the phase current shape is shown in Fig. 4 (c).

![Fig. 4. Three kinds of phase current and flux linkage at different turn-on and turn-off angles with the same maximum value of the phase current.](image)

From (9), the iron loss depends on the maximum flux linkage. The maximum value of the flux linkage in Fig. 4 occurs for \( \theta_{off} \). The copper loss depends on the rms phase current which can be quantified by (7).

The energy conversion loops for 3 kinds of \( i \) and \( \lambda \) by the loci are shown in Fig. 5. The maximum output power can be produced when the phase current is controlled in the shape of a flat top (Fig. 4 (b)). This result has been confirmed by [24].

![Fig. 5. Energy conversion loops by the loci with the same maximum value of the phase current.](image)

### IV. PROPOSED METHOD FOR ANALYZING THE OPTIMAL CONTROL VARIABLES

An analytical modeling method of the optimal control variables to maximize output power of the SRGs in single pulse mode operation is presented in this paper. This paper proposes a method to obtain the phase current equation used to determine the optimal control variables. The phase current equation will be derived from the phase voltage equation in combination with the inductance model. The inductance model is applied from the flux linkage function. Finally, the optimal shape of the phase current is used to determine the optimal control variables.

#### A. Flux linkage model

The flux linkage model in a real system as shown in Fig. 6 has been developed from the flux linkage function in a per-unit system introduced by Stiebler [20]. It requires the geometrical parameters of an SRG at the aligned and unaligned rotor positions. These parameters are easily determined using an experiment or the FEM. The parameters comprise inductance at positions of aligned \( L_a \) and unaligned \( L_u \), and flux linkage at points \( s \) and \( m \) as shown in Fig. 6.

The flux linkage function in Fig. 6 is composed of the linear and saturated regions. The saturated region begins at point \( s \) and finishes at point \( m \). The flux linkage of the saturated region can be determined using a Froelich function [25] \( \lambda = (i/(a + b)) \), where \( a \) and \( b \) are constants as the slope and intercept, respectively. The constants \( a \) and \( b \) can be determined by substituting the \( \lambda_{as} \cdot i_s \) of the point \( s \) and \( \lambda_{am} \cdot i_m \) of the point \( m \) into the Froelich function.

The proposed model of the flux linkage can be
expressed as:

$$\lambda(i, \theta) = L_a i + (L_a - L_u) \frac{i}{a + bi} f(\theta), \quad (11)$$

where \( \theta_k \) is the effective overlap position of the stator and rotor poles, and the angular function is given by:

$$f(\theta) = \begin{cases} 
0.5 + 0.5 \cos \left( \frac{\pi}{\theta_k} \right), & -\theta_k \leq \theta \leq \theta_k \, \\
0, & \text{else}
\end{cases} \quad (12)$$

To verify the proposed method, an 8/6 SRG is used to determine its geometrical parameters using the FEM with its specifications as shown in Table 1.

The relationship between the flux linkage and current at rotor positions 0°, 15°, and 30° is obtained using the FEM and are shown in Table 2.

The parameters obtained using the FEM for calculation in this paper consist of \( L_a = 470 \mu H \), \( L_u = 42 \mu H \), \( \theta_k = 27^\circ \), \( a = 0.65 \), and \( b = 0.155 \).

Figure 7 shows the resultant magnetization curve at rotor positions 0°, 15°, and 30° obtained from the analytical model (11) compared with the FEM which demonstrates the validity of the proposed model.

### Table 1: Specifications of the candidate SRG

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
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</thead>
<tbody>
<tr>
<td>Outer diameter of stator</td>
<td>150 mm</td>
</tr>
<tr>
<td>Inner diameter of stator</td>
<td>70 mm</td>
</tr>
<tr>
<td>Stack length</td>
<td>72 mm</td>
</tr>
<tr>
<td>Length of air gap</td>
<td>0.5 mm</td>
</tr>
<tr>
<td>Number of phases</td>
<td>4</td>
</tr>
<tr>
<td>Stator/Rotor pole arc</td>
<td>23°/23.5°</td>
</tr>
<tr>
<td>Number of stator poles/rotor poles</td>
<td>8/6</td>
</tr>
<tr>
<td>Rated voltage/power/speed</td>
<td>48 V/2.3 kW/6000 rpm</td>
</tr>
</tbody>
</table>

### Table 2: Analytical results obtained using the FEM

<table>
<thead>
<tr>
<th>Current (A)</th>
<th>Rotor Position (Mech. Degree)</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>0°</td>
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<tr>
<td>50</td>
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<td>40</td>
<td><img src="image4.png" alt="Image" /></td>
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<tr>
<td>10</td>
<td><img src="image13.png" alt="Image" /></td>
</tr>
</tbody>
</table>

**B. Proposed model of phase current**

The phase inductance involves much more than a comparison with the mutual inductance, as the mutual inductance is neglected [26, 27]. It is known by \( L(i, \theta) = \lambda(i, \theta)/i \). Therefore, based on (11) and (12), the phase inductance is;

$$L(i, \theta) = L_a + \frac{L_a - L_u}{a + bi} f(\theta), \quad (13)$$

where the angular function \( f(\theta) \) is in the range \(-\theta_k \leq \theta \leq \theta_k\).

The inductance profile is a periodic function with period of \( 2\pi / N_r \) or the rage from \(- \pi / N_r \) to \( \pi / N_r \). Consequently, the phase inductance model proposed in this paper is divided into three regions depending on the
phase current and rotor position as shown in Fig. 8. It can be expressed as:

\[
L(i, \theta) = \begin{cases} 
    L_u & \frac{-\pi}{N_r} \leq \theta < -\theta_k^i \\
    L_u - \frac{a - \frac{u}{a + bi}}{f(\theta)} & -\theta_k^i \leq \theta \leq \theta_k^i \\
    L_u & \theta_k^i < \theta \leq \frac{\pi}{N_r}
\end{cases} 
\]

Fig. 8. Phase inductance profile is divided into three regions.

Figure 9 depicts the phase inductance of the candidate SRG obtained using the proposed analytical model (14) and the FEM so that the characteristics of the phase inductance versus the current and rotor position closely match each other.

![Phase inductance profile](image)

Fig. 9. Phase inductance of the candidate SRG obtained using the mathematical model and the FEM.

The expression of the phase current is obtained by substituting the inductance model (14) into the phase voltage Equation (10). It can be expressed as:

\[
i = \begin{cases} 
    u(\theta - \theta_{on}) & \theta_{on} \leq \theta < \theta_{off} \\
    u(\theta_{e} - \theta) & \theta_{off} \leq \theta \leq \theta_{e} \\
    0 & \text{else}
\end{cases} 
\]

The phase torque is given by:

\[
T(i, \theta) = \frac{1}{2} \frac{\partial L(i, \theta)}{\partial \theta} = \frac{(a + bi)f'(\theta) + bf(\theta)}{2} \left( L_u - L_u \right) f(\theta) i + \frac{u}{\omega} + c
\]

\[
+ (2bL_u + d)(a + bi)^2 \left( L_u - L_u \right)^2,
\]

where

\[
c = \begin{cases} 
    -a & \theta_{on} \leq \theta < \theta_{off} \\
    a & \theta_{off} \leq \theta \leq \theta_{e} \\
    0 & \text{else}
\end{cases}
\]

\[
d = aL_u + (L_u - L_u) f(\theta) - \frac{b}{\omega} h,
\]

\[
h = \begin{cases} 
    u(\theta - \theta_{on}) & \theta_{on} \leq \theta < \theta_{off} \\
    u(\theta_{e} - \theta) & \theta_{off} \leq \theta \leq \theta_{e} \\
    0 & \text{else}
\end{cases}
\]

C. Analysis of optimal control variables

To obtain the maximum output power, the optimal control variables are required from Equation (15) as mentioned in the previous topic. The maximum value of the phase current in Fig. 4 can occur when \( \theta \) is in the range \( \theta_{off} \leq \theta \leq \theta_{p} \). Therefore, the maximum value of the phase current based on (15) can be given as:

\[
i_{\text{max}} = \frac{u(\theta_{e} - \theta)}{\omega} \left[ L_u + \frac{L_u - L_u}{a + bi_{\text{max}}} f(\theta) \right].
\]
as shown in Fig. 4 (b). This result has been confirmed by [24]. Therefore, the shape of the phase current in Fig. 4 (b) is used to determine the optimal control variables so that the maximum value exists in the interval $\theta_{off}$ to $\theta_p$. The maximum value of the phase current at $\theta = \theta_{off}$ or $i_{\text{max}1}$ can be known by substituting $\theta = \theta_{off}$ into (17) which is given by:

$$i_{\text{max}1} = \frac{u(\theta_{off} - \theta_{on})}{\omega L_u + \left(\frac{L_a - L_u}{a + b_{\text{max}1}}\right) \left(1 + \cos\left(\frac{\theta \pi}{\theta_{off} \theta_k}\right)\right)}.$$  \tag{18}

Furthermore, the maximum value of the phase current at $\theta = \theta_{p}$ or $i_{\text{max}2}$ can be determined by substituting $\theta = \theta_{p}$ into (17) which is expressed as:

$$i_{\text{max}2} = \frac{u(2\theta_{off} - \theta_{on} - \theta_p)}{\omega L_u + \left(\frac{L_a - L_u}{a + b_{\text{max}2}}\right) \left(1 + \cos\left(\frac{\theta \pi}{\theta_{p} \theta_k}\right)\right)}.$$  \tag{19}

Now $i_{\text{max}1}$ is equal to the $i_{\text{max}2}$ since the shape of phase current is flat topped. The optimal turn-on angle can be calculated by substituting $i_{\text{max}1}$ into $i_{\text{max}2}$.

Consequently,

$$\theta_{\text{opt}on} = \frac{\theta_{\text{opt}off} (L_u + 2q f_1(\theta) - q f_2(\theta)) - \theta_3 (L_u + q f_1(\theta))}{q f_1(\theta) - f_2(\theta)},$$  \tag{20}

where

$$q = \frac{L_a - L_u}{a + b_{\text{max}}},$$

$$f_1(\theta) = 0.5 + 0.5 \cos\left(\frac{\theta_{\text{optoff}} \pi}{\theta_k}\right),$$

$$f_2(\theta) = 0.5 + 0.5 \cos\left(\frac{\theta_p \pi}{\theta_k}\right).$$

Based on (15) in the range $\theta_{off} \leq \theta \leq \theta_p$, the position of $\theta$ at the maximum current point can be determined by $\frac{di}{d\theta} = 0$:

$$\pi(L_u - L_u) \sin\left(\frac{\theta \pi}{\theta_k}\right) = 2 \left(\frac{u}{\omega}\right) \theta_k (a + b_{\text{max}}).$$  \tag{21}

Then, the optimal turn-off angle can be found by substituting $\theta_{off}$ into $\theta$:

$$\theta_{\text{opt}off} = \frac{\theta_k \sin^{-1}\left(\frac{2u \theta_k}{\omega \pi (L_a - L_u)}\right)}{a + b_{\text{max}}}. $$  \tag{22}

Ultimately, as $\theta_3$, $u$, and $a_{\text{max}}$ are defined, the control variables of the SRG for maximum output power can be calculated as follows:

i. The angular velocity $\omega$ can be determined by substituting values of $u$, $i_{\text{max}}$, and $\theta = \theta_{\text{p}}$ into (21).

ii. The value of $\theta_{\text{opt}off}$ can be found by applying the values of $\theta_3$, $u$, $i_{\text{max}}$, and $\omega$ into (22).

iii. The value of $\theta_{\text{opt}on}$ can be determined from (20).

V. ANALYTICAL AND EXPERIMENTAL RESULTS

To verify the proposed method, an 8/6 SRG system is set up as shown in Fig. 10. A 3-phase induction motor is used as the prime mover so that its speed is controlled by an inverter. The parameters of the SRG are shown in Table 1. A battery rated at 12 V and 120 A is used as the constant dc bus voltage $u$. The average torque $T_m$ of the prime mover is measured by a rotational torque transducer which is connected between the prime mover and the SRG. The shaft speed or angular velocity $\omega$ and aligned position $\theta_a$ are detected by a resolver mounted on the SRG. The SRG is driven by a 4-phase asymmetrical bridge converter so that excitation angles are created by a TMS320F28027. The $R_L$ equals 1.25 $\Omega$ and is used as the resistive load.

![Experimental setup](image-url)

Fig. 10. Experimental setup: (a) 8/6 SRG with a resolver, (b) prime mover, (c) asymmetrical bridge converter and TMS320F28027 DSP controller, (d) variable speed inverter, (e) 12V, 120A battery, (f) resistive load, and (g) torque meter.
Figure 11 shows the schematic layout of the experimental setup so that the mechanical input power can be calculated by:

\[ P_{in} = T_m \omega. \]  

(23)

The efficiency of the system is defined as:

\[ \eta = \frac{P_{out}}{P_{in}}. \]  

(24)

where \( P_{out} \) is the electrical output power and \( P_{in} \) is the mechanical input power.

In this paper, the parameters used for analysis comprise

\[ L_u = 42 \mu H, \quad L_a = 470 \mu H, \quad \beta_s = 23^\circ, \quad \beta_r = 23.5^\circ, \quad a = 0.65, \quad \text{and} \quad b = 0.155. \]

The relationship between the system efficiency of the SRG and the 3 kinds of phase current are investigated, if the \( I_L \) equals to 27.78 A obtained from (6). Analytical results based on the mathematical models (5), (8), (12), and (15) and Figs. 12-14 show the power generation waveforms when the turn-on angle and angular velocity have been adjusted to control \( I_L = 27.78 \) A. The values of \( \theta_{on}, \theta_{off}, \omega, i_{max}, \lambda_{max} \), and \( I_{rms} \) are summarized in Table 3.

Table 3: Results obtained from analytical model

<table>
<thead>
<tr>
<th>Case</th>
<th>( \theta_{on} ) (°)</th>
<th>( \theta_{off} ) (°)</th>
<th>( \omega ) (rad/s)</th>
<th>( i_{max} ) (A)</th>
<th>( \lambda_{max} ) (Wb)</th>
<th>( I_{rms} ) (A)</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>-9.20</td>
<td>9</td>
<td>524</td>
<td>55</td>
<td>0.0193</td>
<td>26.81</td>
</tr>
<tr>
<td>2</td>
<td>-7.80</td>
<td>9</td>
<td>586</td>
<td>40</td>
<td>0.0132</td>
<td>23.85</td>
</tr>
<tr>
<td>3</td>
<td>-8.15</td>
<td>9</td>
<td>605</td>
<td>50</td>
<td>0.0145</td>
<td>24.54</td>
</tr>
</tbody>
</table>

Based on (7) and (9), the copper loss and iron loss depend on \( I_{rms} \) and \( \lambda_{max} \), respectively. In Table 3, the maximum efficiency of the system occurs in case 2 since the copper loss and iron loss are lowest.

The experimental results in Figs. 15-17 show the waveforms of the average torque of the prime mover, phase current, dc bus voltage, and load current. Their values are summarized in Table 4.

The efficiency of the system can be determined using (24). In Table 4, the maximum efficiency of the system occurs in case 2 where the shape of the phase current is flat topped. This result corresponds with the result obtained from the proposed analytical model.

The relationship between the output power of the SRG and the 3 kinds of phase current is investigated...
where the maximum value of the 3 kinds of the phase current is controlled at 40 A by adjusting the control variables. The analytical results, the shapes of the phase inductance, phase flux linkage, phase current, phase torque, and load current obtained from analytical models (14), (11), (15), and (16), respectively, are shown in Figs. 18 (a)-(c). The energy conversion loops are shown in Fig. 18 (d) with the maximum output power occurring in case b. In this case, the phase current shape is flat topped. The control variables are summarized in Table 5.

**Fig. 15.** Case 1: $i_{\text{max}} = 55$ A, $\theta_{\text{on}}$ and $\theta_{\text{off}}$ are -9.20° and 9°.

**Fig. 16.** Case 2: $i_{\text{max}} = 40$ A, $\theta_{\text{on}}$ and $\theta_{\text{off}}$ are -7.80° and 9°.

**Fig. 17.** Case 3: $i_{\text{max}} = 50$ A, $\theta_{\text{on}}$ and $\theta_{\text{off}}$ are -8.15° and 9°.

<table>
<thead>
<tr>
<th>Case</th>
<th>$\omega$ (rad/s)</th>
<th>$T_m$ (N.m)</th>
<th>$P_{\text{in}}$ (W)</th>
<th>$u$ (V)</th>
<th>$I_L$ (A)</th>
<th>$P_{\text{out}}$ (W)</th>
<th>$\eta$ (%)</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>524</td>
<td>2.58</td>
<td>1351.9</td>
<td>36.2</td>
<td>26.7</td>
<td>965.5</td>
<td>71.4</td>
</tr>
<tr>
<td>2</td>
<td>586</td>
<td>2.04</td>
<td>1195.4</td>
<td>36.1</td>
<td>26.8</td>
<td>967.5</td>
<td>80.9</td>
</tr>
<tr>
<td>3</td>
<td>605</td>
<td>2.11</td>
<td>1276.6</td>
<td>36.3</td>
<td>26.6</td>
<td>965.6</td>
<td>75.6</td>
</tr>
</tbody>
</table>

**Table 5:** Three cases of control variables

<table>
<thead>
<tr>
<th>Case</th>
<th>$u$ (V)</th>
<th>$\omega$ (rad/s)</th>
<th>$i_{\text{max}}$ (A)</th>
<th>$\theta_{\text{on}}$ (°)</th>
<th>$\theta_{\text{off}}$ (°)</th>
<th>$I_L$ (A)</th>
<th>$P_{\text{out}}$ (W)</th>
</tr>
</thead>
<tbody>
<tr>
<td>a</td>
<td>36</td>
<td>513</td>
<td>40</td>
<td>-7.8</td>
<td>5.5</td>
<td>16.70</td>
<td>601.2</td>
</tr>
<tr>
<td>b</td>
<td>36</td>
<td>586</td>
<td>40</td>
<td>-7.8</td>
<td>9</td>
<td>27.80</td>
<td>1000.8</td>
</tr>
<tr>
<td>c</td>
<td>36</td>
<td>648</td>
<td>40</td>
<td>-7.8</td>
<td>9.5</td>
<td>24.68</td>
<td>888.5</td>
</tr>
</tbody>
</table>
Three cases of energy conversion loop

Fig. 18. Relationship between $P_{out}$ and 3 kinds of $i$.

The experimental results in Fig. 19 show the waveforms of the phase voltage, phase current, dc bus voltage, and load current. The 3 cases of output power are: case a = 582.8 W, case b = 967.5 W, and case c = 857.9 W. The output power obtained from the measurement is less than the output power obtained from the analytical model since the resistance of the phase windings in the analytical model is neglected. The SRG can produce the maximum output power in case b so that this result corresponds with the analytical result.

(a) Case a: $i_{max} = 40$ A, $\theta_{on} = -7.8^\circ$, $\theta_{off} = 5.5^\circ$, $I_L = 16.1$ A, and $u = 36.2$ V

(b) Case b: $i_{max} = 40$ A, $\theta_{on} = -7.8^\circ$, $\theta_{off} = 9^\circ$, $I_L = 26.8$ A, and $u = 36.1$ V

(c) Case c: $i_{max} = 40$ A, $\theta_{on} = -7.8^\circ$, $\theta_{off} = 9.5^\circ$, $I_L = 23.7$ A, and $u = 36.2$ V

To maximize the output power, the optimal control variables $\omega$, $\theta_{off}$, and $\theta_{on}$ can be calculated as follows:

i. The angular velocity $\omega$ can be determined by substituting $u$, $i_{max}$, and $\theta = \theta_p$ into (21).

ii. The value of $\theta_{off}$ can be found by using the values of $\theta_p$, $u$, $i_{max}$, and $\omega$ in (22).

iii. The value of $\theta_{on}$ can be determined from (20).

Table 6 shows the control variables obtained by the proposed model where $\theta_p$, $u$, and $i_{max}$ are defined as: $\theta_p = 23.25^\circ$, $u = 24$ V, 36 V, and 48 V, and $i_{max} = 30$ A, 40 A, and 50 A.

<table>
<thead>
<tr>
<th>Case</th>
<th>$\theta_p$ ($^\circ$)</th>
<th>$u$ (V)</th>
<th>$i_{max}$ (A)</th>
<th>$\omega$ (rad/s)</th>
<th>$\theta_{on}$ ($^\circ$)</th>
<th>$\theta_{off}$ ($^\circ$)</th>
<th>$I_L$ (A)</th>
</tr>
</thead>
<tbody>
<tr>
<td>i</td>
<td>23.25</td>
<td>24</td>
<td>30</td>
<td>547</td>
<td>-7.2</td>
<td>8.8</td>
<td>19.59</td>
</tr>
<tr>
<td>ii</td>
<td>23.25</td>
<td>36</td>
<td>40</td>
<td>586</td>
<td>-7.8</td>
<td>9</td>
<td>27.8</td>
</tr>
<tr>
<td>iii</td>
<td>23.25</td>
<td>48</td>
<td>50</td>
<td>628</td>
<td>-8.4</td>
<td>9.2</td>
<td>39.76</td>
</tr>
</tbody>
</table>

The analytical results for the three cases, the shapes of the phase A voltage, phase A current, phase B current, and the load current obtained from analytical model are shown in Fig. 20. The shape of the phase current in all cases is flat-topped because the SRG is controlled using the optimal variable controls.

The experimental results in Fig. 21 show the waveforms of the phase A voltage, phase A current, Phase B current, and load current. The 3 cases of output power are: case i = 454.9 W, case ii = 967.5 W, and case iii = 1841.2 W.
(a) Case i: \( \theta_{on} = -7.2^\circ, \theta_{off} = 8.8^\circ \), and \( I_L = 19.59 \text{ A} \)

(b) Case ii: \( \theta_{on} = -7.8^\circ, \theta_{off} = 9^\circ \), and \( I_L = 27.8 \text{ A} \)

(c) Case iii: \( \theta_{on} = -8.4^\circ, \theta_{off} = 9.2^\circ \), and \( I_L = 39.76 \text{ A} \)

Fig. 20. Three cases for waveforms of the phase current, when the SRG is controlled with the control variables obtained from the analytical model.

(a) Case i: \( i_{\text{max}} = 30 \text{ A}, \theta_{on} = -7.2^\circ, \theta_{off} = 8.8^\circ, \) and \( I_L = 18.8 \text{ A}, \) and \( u = 24.2 \text{ V} \)

(b) Case ii: \( i_{\text{max}} = 40 \text{ A}, \theta_{on} = -7.8^\circ, \theta_{off} = 9^\circ, \) and \( I_L = 26.8 \text{ A}, \) and \( u = 36.1 \text{ V} \)

(c) Case iii: \( i_{\text{max}} = 50 \text{ A}, \theta_{on} = -8.4^\circ, \theta_{off} = 9.2^\circ, \) and \( I_L = 38.2 \text{ A}, \) and \( u = 48.2 \text{ V} \)

Fig. 21. Three cases of waveform of phase current, when the SRG is controlled using the control variables obtained by measurement.

The phase current shape in all cases is flat-topped and the maximum output power is produced because the SRG is controlled using the optimal control variables. These results confirm the validity of the proposed analytical model.

The values of the dc bus voltage, load current, and output power for all three cases are summarized in Table 7. The output power obtained from the analytical model is different from the measurements by an average of 3.49%.

<table>
<thead>
<tr>
<th>Case</th>
<th>Mathematical Model</th>
<th>Measurement</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>( u ) (V)</td>
<td>( I_L ) (A)</td>
</tr>
<tr>
<td>i</td>
<td>24</td>
<td>19.59</td>
</tr>
<tr>
<td>ii</td>
<td>36</td>
<td>27.8</td>
</tr>
<tr>
<td>iii</td>
<td>48</td>
<td>39.76</td>
</tr>
</tbody>
</table>
The value of the current obtained from the analytical model is more than the value of the current obtained from the measurement. Since the resistance of the phase windings in the analytical model is neglected. Considering the output power based on (6), the key factor used to calculate is the current. Consequently, the output power obtained from the measurement is slightly less than the output power obtained from the analytical model. The efficiency of the system depends on the system’s losses. The main losses of the system are copper loss and iron loss. The copper loss based on (7), the significant factor used to calculate is the current. The iron loss based on (9) depending on the flux linkage, the key factor used to calculate the flux linkage is the current. Therefore, the result obtained from the analytical model is different from the measurements.

VI. CONCLUSION
In this paper, the proposed inductance model applied from the flux linkage function is divided into three regions depending on the phase current and rotor position. It requires the geometrical parameters of an SRG at aligned and unaligned rotor positions. The parameters are easily quantified using the FEM. The characteristics of the inductance curve obtained using the proposed model compared with the FEM are closely matched. This result confirms the validity of the proposed model. The phase current model proposed in this paper is derived from the phase voltage equation in combination with the proposed inductance model. The shape of phase current obtained from the analytical model is also corresponding with the measurements. The optimal shape of phase current is investigated. Finally, a method to obtain the optimal control variables to maximize the output power for SRGs in single pulse mode operation is proposed. The optimal shape of the phase current is used to determine the optimal control variables. An 8/6 SRG experimental setup is used to verify the proposed method. Regarding to the results, the SRG can generate the maximum output power when the proposed optimal control variables are applied. The output power obtained from the analytical model is slightly different from the measurements. Therefore, the proposed method is accurate and reliable.

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Electromagnetic Coupling Analysis of Transient Excitations of Rectangular Cavity through Slot using TD-EFIE with Laguerre Polynomials as Temporal Basis Functions

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Abstract — This paper presents an electromagnetic coupling analysis of transient waves excited a rectangular cavity containing an interior scatterer coupled to an external scatterer through a slot. Based on the equivalence principle, time domain integral equations are established by enforcing the boundary conditions on the internal and external scatterers, the slot and the cavity walls. The method of moments is applied in space and time domains to solve the developed system of integral equations. The unknown coefficients of the electric and magnetic currents are approximated by the triangular piecewise functions associated with Dirac function as spatial basis. To obtain accurate and stable solutions, the Laguerre functions are used as temporal basis. Numerical results involving the coupling effects between the cavity components are investigated. The numerical results are found to be in good agreement with EM theory and literature.

Index Terms — Coupling effect, equivalence principle, Laguerre functions, MoM, piecewise triangular functions, slot, TD-EFIE, transient electric and magnetic currents.

I. INTRODUCTION

In electromagnetic compatibility, it is interesting to model the coupling through a slot-aperture backed by a rectangular cavity. When the cavity is excited by internal field source, it acts as an aperture antenna that radiates in external region. The problem of coupling between this radiation element and some components close to the cavity can influence the integrity of communication system. Another typical problem is the evaluation of the current induced by incident electromagnetic fields through an aperture because this current may damage some critical components in the system. Also the response of the cavity through the slot can damage the equipments at the side of cavity.

In order to tackle the two problems, many researchers have developed a lot of frequency domain methods for many cavity designs [1]-[5]. But, for the defense and security reasons, the transient response of these problems can be depicted easily in time domain.

The transient response of the cavity through an aperture is obtained primarily by Inverse Fourier Transforms (IFT) [6]. An exact solution using this method cannot be found. Reference [7] solves the transient problem by using the singularity expansion method (SEM). Recently, the integral equations are formulated and solved in time domain using a time stepping technique [8]-[11]. That later suffers from the late-time instabilities.

In this paper, we consider a communication system composed on a rectangular cavity backed slot containing an internal (probe) and external (thin wire antenna) scatterers. The latters are coupled through the slot.

The aim of this work is to develop a general system of Time Domain Electric Field Integral Equations (TD-EFIE) based on the equivalence principle and the appropriate boundary conditions. In fact, when the cavity is excited through the slot by internal and/or external transient electromagnetic waves, we can apply the developed integral equations system to predict the transient responses of the structure and to study the physical coupling effects between the component of the designed system.

In order to solve the proposed TD-EFIE by applying the Method of Moments (MoM) [12], we introduce a spatial and temporal testing procedures. The piecewise triangular functions [13] associated with Dirac functions are used as spatial basis functions. The present paper sets out to present an accurate technique to obtain stable solutions of the obtained TD-EFIE system using Laguerre polynomials as temporal basis functions [14]-[18]. The transient responses of the structure are compared to the results obtained by applying the B-Spline functions as a temporal basis [22]. In fact, an accurate comparison of the two time approaches (Laguerre scheme and B-Spline scheme) in terms of complexity, accuracy and stability was presented in [22].
In this work and based on the results stemming from this comparison, a coupling between the thin wire antenna and the probe through the slot is studied using Laguerre functions. The space distributions of the currents are presented.

The paper is divided in four sections. Section II presents the integral equations formulation in the time domain and the method of solving these equations. Numerical results are given in Section III. Section IV concludes the paper.

II. TD-EFIE FORMULATION

The geometry of the analyzed problem is depicted in Fig. 1, which consists of a cavity with a slot on its wall, located in the X-Y plan. A linear electric probe of the length \( l_p \) at the position \((x_p, y_p)\) is located inside the cavity at the distance \( D \) of the slot. Outside the cavity, we consider a thin wire antenna of the length \( l_A \), vertically placed at the distance \( d \) from the cavity and located in parallel to the slot at the position \((x_A, y_A)\).

To investigate the radiation characteristics of the structure, we assume that:
- The cavity walls, the antenna and the probe are considered as perfect conductors and very thin.
- The antenna is excited by \( \mathbf{E}_{\text{ant}} = E_{\text{ant}} \hat{y} \).
- The probe is excited by \( \mathbf{E}_{\text{probe}} = E_{\text{probe}} \hat{y} \).

Fig. 1. The geometry of the problem.

By involving the equivalence principle [19], an equivalent surface replaces the physical structure with equivalent magnetic current density radiating in free space over the slot. References [20]-[21] have demonstrated that the magnetic current density is moving an infinitesimal distance away from equivalent surface. Mathematical analysis of this step shows that the slot can be short-circuited. Therefore, two coupled-equivalent problems are established (Fig. 2). In both the external and internal equivalent problems, we retain the sources in the region of interest. In Fig. 3, we represent the different domains:

- \( D = \Omega_1 \cup \Omega_3 \cup \Omega_4 \) the external analyzed domain.
- \( D = \Omega_2 \cup \Omega_3 \cup \Omega_4 \) the internal analyzed domain.

Where \( \Omega_1 \) is the surface of the antenna; \( \Omega_2 \) is the surface of the probe; \( \Omega_3 \) is the equivalent domain replacing the slot; \( \Omega_4 \) is equivalent surface replacing the cavity walls. In fact, the domain \( \Omega_3 \) is the sub-domain infinitesimally close to, but not coincident with \( \Omega_4 \). Over it remains not null the equivalent magnetic current.

Fig. 2. The equivalent problem: (a) equivalent external problem, and (b) equivalent internal problem.

Fig. 3. The different domains: cavity walls, slot, antenna and probe.
The external problem contains the equivalent magnetic current density \( \vec{M} \), and the unknown electric current \( \vec{J}_1 \) (Fig. 2 (a)). The fields inside \( \Omega_4 \) are zero and outside are equal to the radiation produced by the incident field \( \vec{E}_o \) and scattered fields induced by \( \vec{M} \) and \( \vec{J}_1 \). These fields must satisfy the following boundary condition:

\[
\vec{M} = -\vec{n} \times \vec{E}_o, \quad (1)
\]

where

\[
\vec{E}_o = \vec{E}_{o1} + \vec{E}^{inc'}(\vec{M}) + \vec{E}^{sc}(\vec{J}_1). \quad (2)
\]

For the internal problem, as shown in Fig. 2 (b), the equivalent magnetic current density \( \vec{M} \), and the unknown electric current \( \vec{J}_2 \) are selected so that the fields outside equivalent surface \( \Omega_4 \) are zero and inside are equal to the radiation produced by \( \vec{E}_{o2} \) and the scattered fields \( \vec{E}^{inc'}(\vec{J}_2) \) and \( \vec{E}^{sc}(-\vec{M}) \). These fields must satisfy the boundary condition (1), where

\[
\vec{E}_2 = \vec{E}_{o2} + \vec{E}^{inc'}(\vec{J}_2) + \vec{E}^{sc}(-\vec{M}). \quad (3)
\]

An alternate formulation of the TD-EFIE is applied in this paper to describe unknown equivalent magnetic current sheet over the slot and the unknown electric currents at the probe and the antenna. This formulation is based on the equivalence principle in addition to enforce the following boundary conditions:

- The tangential magnetic fields are contentious through the slot aperture both inside and outside the cavity.
- The source is considered at the bottom of the probe inside the cavity.
- The source is considered at the center of the antenna outside the cavity.

Taking into account the interaction between the probe, the antenna and slot we obtain the following system:

\[
\begin{align*}
\left[ \vec{E}_{o1} + \vec{E}^{inc'}(\vec{J}_1) + \vec{E}^{sc}(\vec{M}) \right]_{\Omega_4} &= 0 \quad \text{over } \Omega_4, \\
\left[ \vec{E}_{o2} + \vec{E}^{inc'}(\vec{J}_2) + \vec{E}^{sc}(-\vec{M}) \right]_{\Omega_4} &= 0 \quad \text{over } \Omega_4, \\
\left[ \vec{E}^{inc'}(\vec{J}_1) + \vec{E}^{sc}(\vec{M}) \right]_{\Omega_3} &= \left[ \vec{E}^{inc'}(\vec{J}_2) + \vec{E}^{sc}(-\vec{M}) \right]_{\Omega_3} \quad \text{over } \Omega_3, \quad (4) \\
\left[ \vec{E}^{inc'}(\vec{J}_1) + \vec{E}^{sc}(\vec{M}) \right]_{\Omega_4} &= 0 \quad \text{over } \Omega_4, \\
\left[ \vec{E}^{inc'}(\vec{J}_2) + \vec{E}^{sc}(-\vec{M}) \right]_{\Omega_4} &= 0 \quad \text{over } \Omega_4,
\end{align*}
\]

where

\[
\vec{E}^{inc'}(\vec{M}) = -\frac{1}{\varepsilon} \nabla \times \vec{F}, \quad (5)
\]

\[
\vec{E}^{inc'}(\vec{J}_2) = -\frac{\mu}{Ct} \vec{A} - \nabla \phi. \quad (6)
\]

The magnetic vector and electric scalar potentials, \( \vec{A} \) and \( \phi \), are expressed on the antenna (i=1) or the probe (i=2) which are mathematically given by:

\[
\vec{A}(x, y, t) = \frac{\mu}{4\pi} \iint_{\Omega_1} \frac{\vec{J}(x', y', t - \frac{R}{C})}{R} \, ds', \quad (x, y) \in \Omega_j, \quad (7)
\]

\[
\phi(x, y, t) = \frac{1}{4\pi} \iint_{\Omega_2} \frac{q(x', y', t - \frac{R}{C})}{R} \, ds', \quad (x, y) \in \Omega_j, \quad (8)
\]

where \( \mu \) is the permeability and \( C \) is the velocity of propagation of the electromagnetic wave in free space.

The potential electric vector \( \vec{F} \) is given by the time retarded integral equation involving the magnetic current \( \vec{M} \). It is mathematically obtained by application of the concept of electromagnetic duality for Maxwell theories:

\[
\vec{F}(x, y, t) = \frac{\varepsilon}{4\pi} \iint_{\Omega_1} \frac{\vec{M}(x', y', t - \frac{R}{C})}{R} \, ds', \quad (x, y) \in \Omega_j, \quad (9)
\]

where \( \vec{M} \) is the equivalent density current vector along \( \Omega_3 \) and \( \varepsilon \) is permittivity of free space. The retarded time is expressed by \( t - \frac{R}{C} \). The distance \( R_s = \sqrt{(x-x')^2 + (y-y')^2} \) represents the interaction between the observation point \( (x, y) \in \Omega_j \) and the source point \( (x', y') \) as shown in Fig. 4.

Fig. 4. The different distances between the observation and source points for the internal and external equivalent problems.

For convenience of numerical computation [13], it is useful to derive the scalar and vector potentials from the sources terms by means of the Hertz vectors which does not have a time integral term. Hence, the electric current and the charge density can be expressed in terms of the single Hertz vector \( \vec{G} \) :

\[
\vec{J}_1(x, y, t) = \frac{d}{dt} \vec{G}_1(x, y, t), \quad (10)
\]

\[
q(x, y, t) = -\nabla \cdot \vec{G}_1(x, y, t). \quad (11)
\]

The magnetic current is also expressed by means of the Hertz vector \( \vec{E} \) :

\[
\vec{M}_1(x, y, t) = \frac{d}{dt} \vec{E}_1(x, y, t). \quad (12)
\]

By substituting (5)-(12) in (4), we obtain the following time domain integral equations system:
approximate the unknown Hertz vectors \( \vec{G}_i \), \( \vec{G}_3 \) and \( \vec{L} \). Indeed, the choice of the basis function strongly influences the properties of the MoM approximation of the unknowns. We define the spatial basis functions as the piecewise triangular functions [13] associated with Dirac function \( \tilde{f}_n = H^{(\alpha)}_n N^{(\alpha)}_{x_i} \), where,

\[
H^{(\alpha)}_n(h) = \begin{cases} 
  h - h_{i-1}, & h \in [h_{i-1}, h_i] \\
  h_{i+1} - h, & h \in [h_i, h_{i+1}] \\
  0, & \text{otherwise}
\end{cases}
\]

\[
N^{(\alpha)}_{x_i}(x) = \begin{cases} 
  1, & x = x_i \\
  0, & \text{otherwise}
\end{cases}
\]

The choice of the piecewise triangular function has the following advantages: it minimized the required computational and it leads to an accurate evaluation of the unknowns with few expansion functions.

The goal of the association between the Dirac function and the piecewise triangular function is metalized the domains \( \Omega_i \) only.

Now, we express the Hertz vectors \( \vec{G}_i \) and \( \vec{L} \) in terms of vector basis functions \( \tilde{f}_n \) and \( \tilde{h}_n = \vec{n} \times \tilde{f}_n \), respectively:

\[
\vec{G}_i(x, y, t) = \sum_{n=1}^{N_1} g_n(t) \tilde{f}_n(x, y), \quad (19)
\]

\[
\vec{f}_n(x, y) = H^{(\alpha)}_n(y) A^{(\alpha)}_{x_i}(x) \vec{y}, \quad (20)
\]

\[
\vec{L}(x, y, t) = \sum_{n=1}^{N_1} \tilde{l}_n(t) \tilde{h}_n(x, y), \quad (21)
\]

Note that the axis of the antenna, the probe and the slot are divided into \( N_1, N_2 \) and \( N_3 \) equal sub-domains of which the space step is \( \Delta y \).

In order to discretize \( \Omega_x \), we divide the cavity walls in \( N_4 \times N_4 \) equal sub-domains whose the lengths are \( \Delta x \) and \( \Delta y \) along \( x-y \) directions, respectively. The functions \( \tilde{f}_n^4 \) illustrated in Fig. 5 is given by:

\[
\tilde{f}_{n}^4(x, y) = H^{(\alpha)}_{\beta(x,y)}(y) A^{(\alpha)}_{x_{n_{y}}}^{(\beta)(x,y)}, \quad (22)
\]
We substitute (19)-(21) into (13)-(17), then we apply the spatial testing procedure, we obtain:

\[
\sum_{n=0}^{N} \left[ - \mu_{11} d\bar{g}_{11}^{2}(t-n\omega_{c}/\varepsilon) + b_{11}^{2} g_{1}(t-n\omega_{c}/\varepsilon) \right] dt + \sum_{n=0}^{N} c_{11} d\bar{l}_{1}(t-n\omega_{c}/\varepsilon) \bigg|_{t=0}^{t=t_{a}} = -E_{a}(t), \tag{23}
\]

\[
\sum_{n=1}^{N} \left[ - \mu_{12} d\bar{g}_{12}^{2}(t-n\omega_{c}/\varepsilon) + b_{12}^{2} g_{1}(t-n\omega_{c}/\varepsilon) \right] dt + \sum_{n=1}^{N} c_{12} d\bar{l}_{1}(t-n\omega_{c}/\varepsilon) \bigg|_{t=0}^{t=t_{a}} = 0, \tag{24}
\]

\[
\sum_{n=1}^{N} \left[ - \mu_{21} d\bar{g}_{21}^{2}(t-n\omega_{c}/\varepsilon) + b_{21}^{2} g_{1}(t-n\omega_{c}/\varepsilon) \right] dt + \sum_{n=1}^{N} c_{21} d\bar{l}_{1}(t-n\omega_{c}/\varepsilon) \bigg|_{t=0}^{t=t_{a}} = 0, \tag{25}
\]

\[
\sum_{n=1}^{N} \left[ - \mu_{22} d\bar{g}_{22}^{2}(t-n\omega_{c}/\varepsilon) + b_{22}^{2} g_{1}(t-n\omega_{c}/\varepsilon) \right] dt + \sum_{n=1}^{N} c_{22} d\bar{l}_{1}(t-n\omega_{c}/\varepsilon) \bigg|_{t=0}^{t=t_{a}} = 0. \tag{26}
\]

\[
\sum_{n=1}^{N} \left[ - \mu_{31} d\bar{g}_{31}^{2}(t-n\omega_{c}/\varepsilon) + b_{31}^{2} g_{1}(t-n\omega_{c}/\varepsilon) \right] dt + \sum_{n=1}^{N} c_{31} d\bar{l}_{1}(t-n\omega_{c}/\varepsilon) \bigg|_{t=0}^{t=t_{a}} = 0, \tag{27}
\]

\[
\sum_{n=1}^{N} \left[ - \mu_{32} d\bar{g}_{32}^{2}(t-n\omega_{c}/\varepsilon) + b_{32}^{2} g_{1}(t-n\omega_{c}/\varepsilon) \right] dt + \sum_{n=1}^{N} c_{32} d\bar{l}_{1}(t-n\omega_{c}/\varepsilon) \bigg|_{t=0}^{t=t_{a}} = 0. \tag{28}
\]

We assume that the unknown transient quantities \(g_{a}(t), l_{a}(t)\) does not change appreciably within the segment \(\Delta y\) so that \(\bar{E}_{a}(t) = \bar{E}_{a}(y), \bar{E}_{a}(y+\Delta y) = \bar{E}_{a}(y)\):

\[
\bar{E}_{a}(t) = \left( \bar{f}_{a}(x, y), \bar{E}_{a}(x, y) \right) \bigg|_{t=0}^{t=t_{a}} = \int_{\Omega} \bar{f}_{a}(x, y) \left[ \bar{E}_{a}(x, y, y, t) \right] ds, \tag{29}
\]

On the other hand, we consider the temporal procedure. In order to obtain stable and accurate solutions, a temporal basis functions derived from the Laguerre polynomials \[14\] is applied. The transient electric and magnetic coefficients introduced in (19) and (20) are expanded by:

\[
g_{a}(t) \approx \sum_{n=0}^{N} g_{a,n} \varphi_{a}(st), \tag{30}
\]

\[
l_{a}(t) \approx \sum_{n=0}^{N} l_{a,n} \varphi_{a}(st), \tag{31}
\]

where \(\{g_{a,n}, l_{a,n}\}\) are the unknown coefficients; \(\varphi_{a}(st), a = 0, \infty\) is the temporal basis functions derived from the Laguerre function \(\varphi_{a}(st) = e^{-s/2}L_{a}(st)\). So, the term \(L_{a}(st)\) represents the Laguerre polynomial of order “a” and “s” is a time scaling factor \[17\]. The mathematical properties of these functions, the first and the second derivatives are introduced in \[14\]-[15].

In order to apply the Galerkin’s Method \[12\] in time domain, we substitute (30)-(33) into (23)-(27) and we apply the temporal testing procedure with \(\varphi_{a}(st)\), we obtain the following system:

\[
\sum_{n=0}^{N} \left[ - \mu_{11} d\bar{g}_{11}^{2}(t-n\omega_{c}/\varepsilon) + b_{11}^{2} g_{1}(t-n\omega_{c}/\varepsilon) \right] \varphi_{a}(st) dt + \sum_{n=0}^{N} c_{11} d\bar{l}_{1}(t-n\omega_{c}/\varepsilon) \bigg|_{t=0}^{t=t_{a}} = -E_{a}(t), \tag{32}
\]
We note that, we can change the upper limit of the sum (30) and (31) form \( \infty \) to “b” based on the orthogonality condition detailed in [14]:

\[
I_{\nu}(s \frac{R_{\nu}}{C}) = \left\{ \varphi_{\nu}(st), \varphi_{\nu}(s(t - \frac{R_{\nu}}{C})) \right\}
\]

\[
= \int \varphi_{\nu}(st) \varphi_{\nu}(s(t - \frac{R_{\nu}}{C}))d(st)
\]

\[
= \left[ e^{-\frac{ak}{2c}} [L_{a,b}(s \frac{R_{a,b}}{C}) - L_{b,a}(s \frac{R_{a,b}}{C})] \right] a \leq b\]

The scalar product of the field \( E_{\nu}^{i} \) (28) by the Laguerre function of order “b” is given by:

\[
E_{\nu}^{i} = \int_{0}^{\infty} \varphi_{\nu}(st)E_{\nu}^{i}(t)d(st).
\]  

(33)

In the system (32), we move the terms including \( g \), and \( i_{a,b} \), which is known for \( a < b \) to the right-hand side. Rewriting the resulting equations in a simple form, we have:

\[
\sum_{n=1}^{N} A_{a,n}g_{a,n} + \sum_{n=1}^{N} b_{a,n} = E_{a} + S_{a},
\]

(34)

\[
\sum_{n=1}^{N} A_{a,n}g_{b,n} + \sum_{n=1}^{N} b_{b,n} = E_{b} + S_{b},
\]

(35)

\[
\sum_{n=1}^{N} A_{a,n}g_{a,b,n} + \sum_{n=1}^{N} b_{a,b,n} = K_{a},
\]

(36)

\[
\sum_{n=1}^{N} A_{a,n}g_{b,a,b,n} + \sum_{n=1}^{N} b_{b,a,b,n} = K_{b}.
\]

(37)

\[
\sum_{n=1}^{N} A_{a,n}g_{a,b,n} + \sum_{n=1}^{N} b_{a,b,n} = \]

(38)

The spatial matrices \( [A, B] \) and the retarded terms \( S_{a}, S_{b}, K_{a} \) and \( H_{b} \) are presented in appendix. 

Now, we write (34)-(38) in a matrix form:

\[
\begin{bmatrix}
A_{11} & 0 \\
A_{12} & B_{11} \\
A_{13} & B_{12} \\
A_{14} & B_{13} \\
0 & B_{14}
\end{bmatrix}
\begin{bmatrix}
g_{a} \\
g_{b}
\end{bmatrix}
= \begin{bmatrix}
E_{a} + S_{a} \\
E_{b} + S_{b} \\
K_{a} \\
K_{b}
\end{bmatrix}.
\]

(39)

It is important to note that the matrix A is not a function of the degree of the temporal testing function “b”. Therefore, we obtain the unknown coefficients by solving (39) as increasing the degree of temporal testing functions. Consequently, we solve the problem only in space for each degree of the Laguerre function. Finally, the transient currents are calculated by:

\[
J_{i}(t) = \frac{d}{dt} G_{i}(t) = \sum_{i=1}^{N} \left[ \frac{1}{2} g_{i} + \sum_{k=0}^{N} g_{i,k} \right] \varphi_{i}(st),
\]

(40)

\[
M_{i}(t) = \frac{d}{dt} L_{i}(t) = \sum_{i=1}^{N} \left[ \frac{1}{2} t_{i} + \sum_{k=0}^{N} l_{i,k} \right] \varphi_{i}(st),
\]

(41)

where A is the maximum order of the Laguerre function [16].

### III. NUMERICAL RESULTS

In this section, we present the different numerical results by applying the proposed formulation and we compared the latters with the results obtained by the B-Spline scheme, developed in [22]. Thus, we consider the structure shown in Fig. 6. Their related parameters are presented in Table 1. The space parameters \( N_{j} \), \( 1 \leq i \leq 4 \) are described in Table 2.

![Fig. 6. Structure of the problem.](Image)

**Table 1: Cavity parameters**

<table>
<thead>
<tr>
<th>Symbol</th>
<th>Quantity</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>a</td>
<td>Cavity width</td>
<td>0.346λ, m</td>
</tr>
<tr>
<td>b</td>
<td>Cavity length</td>
<td>0.692λ, m</td>
</tr>
<tr>
<td>( l_{x} )</td>
<td>Slot length</td>
<td>0.048λ, m</td>
</tr>
<tr>
<td>( (x_{0}, y_{0}) )</td>
<td>Center of slot</td>
<td>(0.346λ, 0.173λ)</td>
</tr>
<tr>
<td>( l_{p} )</td>
<td>Probe length</td>
<td>0.25λ, m</td>
</tr>
<tr>
<td>( (x_{p}, y_{p}) )</td>
<td>Probe center</td>
<td>(0.172λ, 0)</td>
</tr>
<tr>
<td>( l_{A} )</td>
<td>Antenna length</td>
<td>0.25λ, m</td>
</tr>
<tr>
<td>( (x_{A}, y_{A}) )</td>
<td>Position of antenna</td>
<td>(a+0. 172λ, 0)</td>
</tr>
<tr>
<td>e</td>
<td>Thickness</td>
<td>10⁻³</td>
</tr>
</tbody>
</table>

In this table, we used same parameters of the cavity presented in [2] at the resonance frequency of 1.9 GHz.

**Table 2: Space parameters**

<table>
<thead>
<tr>
<th>Symbol</th>
<th>Quantity</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>( \Delta x )</td>
<td>Space step</td>
<td>2.73 10⁻³ m</td>
</tr>
<tr>
<td>( \Delta y )</td>
<td>Space step</td>
<td>5.46 10⁻³ m</td>
</tr>
<tr>
<td>( N_{1} )</td>
<td>Number of antenna sub-domains</td>
<td>( l_{A}/\Delta y )</td>
</tr>
<tr>
<td>( N_{2} )</td>
<td>Number of probe sub-domains</td>
<td>( l_{P}/\Delta y )</td>
</tr>
<tr>
<td>( N_{3} )</td>
<td>Number of slot sub-domains</td>
<td>( l_{A}/\Delta y )</td>
</tr>
<tr>
<td>( N_{4} )</td>
<td>Number of cavity walls sub-domains</td>
<td>( (N_{A}, N_{S}) = (a/\Delta x, b/\Delta y) )</td>
</tr>
</tbody>
</table>
We assume that the probe (inside the cavity) is excited by the Gaussian pulse voltage generator \( V(t) \) at the bottom. The antenna is also excited by \( V(t) \) at its center. The source \( V(t) \) is expressed by:

\[
V(t) = V_0e^{-(t-t_0)/\tau},
\]

(42)

with parameters \( V_0 = 4V \), \( \tau = 10^8 s^{-1} \) and \( t_0 = 10^{-9} s \). We suppose that the antenna and the probe have same length \( (l_A = l_p) \) and same radius \( e \).

The value of the time scale factor \( s \) is \( 2.10^9 \) and the number of Laguerre functions \( A \) is fixed at 80. The values of \( s \) and \( A \) are sufficient to get accurate solutions. In fact, the terms \( s \) and \( A \) are the Laguerre function parameters.

In order to verify the results, we apply the B-Spline scheme developed in [22]. In fact, for the two schemes (Laguerre and B-Spline), we consider the case when the two antennas are excited by (42). The transient responses of the structure are depicted in Fig. 7 and Fig. 8. Thus, we clearly observe the good agreement between the results obtained by the two schemes.

Consequently, we apply the Laguerre scheme developed in this paper to study the coupling between the components of the designed communication system. This scheme is unconditionally stable. The space and the time steps are not related as the most temporal techniques (FDTD, TLM, MOT). We obtain stable and accurate solutions.

![Fig. 7. The transient response of the antenna: (a) the normalized electric current density at the instant \( t=1.33\times10^{-9} \) s, and (b) transient electric current density at the center of antenna.](image)

![Fig. 8. The transient response of the probe: (a) the normalized electric current density at the instant \( t=1.33\times10^{-9} \) s, and (b) transient electric current density at the top of the probe.](image)

### A. Electric and magnetic currents distributions

In order to describe the transient behaviors of the structure, three cases are considered:

- The antenna and the probe are both excited.
- Only the antenna is excited.
- Only the probe is excited.

The transient responses are determined at the center of antenna, at the bottom of probe and the center of slot.

Figure 9 shows the space distribution of electric current density of the antenna. It is obvious that the peak of the electric current occurred at the center point as expected for the case when the antenna is excited. But when the antenna is unexcited, the current vanish. The transient response plotted, in Fig. 10, at the center of the antenna confirms the latter interpretation.

The transient response of the probe is shown in Fig. 11 and Fig. 12; we should note that the peak of electric current occurred at the feed point and the amplitude vanish when the probe is unexcited. The miniature curve in Fig. 11 presents the coupling effect when the probe is unexcited and the antenna is excited.

The magnetic current density is presented in space and time when the antenna and the probe are both excited in Fig. 13.
B. Study of coupling through the slot depending on separate distances \(d\) and \(D\)

In order to study the Antenna-Slot coupling and Probe-Slot coupling, we consider four separate distances:

\[
\begin{align*}
d &= D = d_1 = 0.018 \times a, \\
d &= D = d_2 = 0.146 \times a, \\
d &= D = d_3 = 0.274 \times a, \\
d &= D = d_4 = 0.875 \times a.
\end{align*}
\]

Starting by the study of the coupling effect of antenna and cavity as function of different distances \(d\) (the value of \(D\) is fixed in Table 1).

Figure 14 represents the theoretical coupling between the antenna and the slot calculated via the matrix \(A_{ij}\). It is clearly that the coupling vanishes from \(d = 0.005\) m. The transient magnetic current plotted in Fig. 15 for different separated distances \(d\) confirms the theoretical results. We note that, if \(d > d_c\), the magnetic current does not change.

To study the coupling between the cavity and the probe, we vary the separate distances \(D\) and we fix the distance \(d\) (Table 1).

The coupling vanishes from \(d = 0.008\) m as presented in Fig. 16. This result is detailed in Fig. 17; but for the distance \(d_c\), the current amplitude is non-zero and a low variation is detected. This variation is due to the effect of the cavity walls which amplified the currents inside the cavity.

Also, lower values of the electric current at the probe and the magnetic current at the slot are observed when the cavity size becomes larger (Fig. 18).
C. Study of coupling through the slot depending on slot length

It is very important to study the coupling taking into account the length of the slot. Therefore, we vary the latter parameter and two cases can be defined. One hand, when only the probe is excited, the transient current density at the top of the antenna as shown in Fig. 19. The amplitude becomes more important when the length of the slot is increased. This result is confirmed by this shown in Fig. 10.

On the other hand, when only the antenna is excited, the transient current density at the top of probe for the shorter slot is lower than the longer ones, as depicted in Fig. 20.

**IV. CONCLUSIONS**

The transient behaviors of communication system composed on rectangular cavity containing an interior scatterer coupled to an external scatterer through a slot were studied. A 2-D numerical time domain formulation based on the combination of the equivalence principle and the MoM has been successfully developed and applied to the designed system.

The physical coupling effects between the components of the system through the slot have been provided depending on the separate distances and the lengths of the slot. Stable and accurate results have been found mainly with electric and magnetic currents.
responses.

This formulation can be extended to 3-D and applied for many complex structures.

**APPENDIX**

The spatial matrices $A_{mn}^\theta$ and $B_{mn}^\theta$ can be defined as:

$$
A_{mn}^\theta = \left[ -s^2 \mu \mu_{mn} \delta_{ij} \right] I_{s \theta} \left( R_{mn}^\theta \right),
$$

$$
B_{mn}^\theta = \pm \frac{s}{2} I_{s \theta} \left( R_{mn}^\theta \right),
$$

where

$$
I_{s \theta} \left( R_{mn}^\theta \right) = e^{-s R_{mn}^\theta}.
$$

The retarded terms takes the following forms:

$$
S_{11}^\theta = \frac{1}{\mu} \sum_{i=1}^{N_n} \sum_{j=1}^{N_m} \left[ \frac{1}{2} \delta_{ij} + \frac{1}{4} \sum_{k=1}^{N_n} \delta_{ij} \right] l_{i,j} \left( R_{mn}^\theta \right),
$$

$$
S_{21}^\theta = \frac{1}{\mu} \sum_{i=1}^{N_n} \sum_{j=1}^{N_m} \left[ \sum_{k=1}^{N_n} \delta_{ij} \right] l_{i,j} \left( R_{mn}^\theta \right),
$$

$$
S_{12}^\theta = \frac{1}{\mu} \sum_{i=1}^{N_n} \sum_{j=1}^{N_m} \left[ \frac{1}{2} \delta_{ij} \right] l_{i,j} \left( R_{mn}^\theta \right),
$$

$$
S_{22}^\theta = \frac{1}{\mu} \sum_{i=1}^{N_n} \sum_{j=1}^{N_m} \left[ \frac{1}{2} \delta_{ij} + \frac{1}{4} \sum_{k=1}^{N_n} \delta_{ij} \right] l_{i,j} \left( R_{mn}^\theta \right).
$$

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Ultra-Wideband Balanced Bandpass Filters Based on Transversal Signal-Interference Concepts

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Abstract — Two ultra-wideband balanced bandpass filters based on transversal signal-interference concepts are proposed in this paper. By employing the two structures: microstrip/slotline transition, and quarter-wavelength shorted coupler lines, 180° phase shift can be easily implemented for each design. In addition, due to the different transmission paths, wideband common mode rejection and differential mode transmission bandwidth can be achieved. Two balanced filters centered at 3 GHz are calculated, fabricated and measured successfully. For the differential mode, the 3-dB fractional bandwidths are 102% from 1.47 GHz to 4.53 GHz and 103% from 1.52 GHz to 4.61 GHz, respectively, and the return loss are both greater than 20 dB. For the common mode, signals are suppressed below -20 dB and -15 dB over the whole frequency band.

Index Terms — Balanced filter, differential/common mode, shorted stubs, Ultra-wideband.

I. INTRODUCTION

RF circuits and systems are becoming a complicated closer space with more functions. So, the electromagnetic interference between nodes at different dielectric layers and mutual link in communication systems to be eliminated is indispensable. Compared to single-ended technology, balanced circuits have advantages of wideband common mode rejection capability which is beneficial to immunity to the environmental noise and dynamic range.

As a high performance balanced circuit, only the differential mode signals pass through in the desired frequency band, while the common mode signals must be all-stop. In the past few years, different balanced filters for single-band, dual-band and wideband were proposed using different ways [1-8]. Chen et al. designed differential filter firstly with wideband common-mode suppression exploiting microstrip and CPW transition [1]-[3]. However, the differential-mode width increase difficulty and the insertion loss is a little big. Cascaded branch-line was used to design wideband balanced filters as well [4]-[5]. However, these filters implied larger circuit size, and their out-of-band common mode suppression is not so good. Another way to get wideband balanced filter is using T-shaped structures [6]. Their main advantage is the high selectivity of the passband, but the suppression of common mode is always unsatisfactory. As a very convenient balanced transmission line for balanced circuits, DSPSL can be used to design ideal phase shifter for balanced filters [7]-[8]. They show good performance for the differential mode and common mode, but the problem of heat dissipation remained unsolved. In [9], a low profile balanced tunable BPF using λ/2 resonator has been presented. The common mode suppression can be kept at a high level by adding a varactor. In [10], it is a balanced SIW BPF which is simplified though designing a 2-port filter with high performance. But [9] and [10] both have the defect of differential-mode bandwidth.

II. ANALYSIS OF PROPOSED BALANCED FILTERS

In this part, the proposed two balanced filters based on transversal signal-interference concepts are illustrated detailedy. Firstly, we use the differential mode and common mode equivalent circuits to analyze their performance. And then, in Part C, simulated results of the two balanced filters are presented.

A. Balanced filter analysis with microstrip/slotline transition

The ideal circuit of the balanced filter with two ideal 180° phase shifter is shown in Fig. 1 (a). The transformer works as a phase shifter. We see that it is a central symmetric structure of all components. Specifically, there are two shunted λ/4 shorted lines loaded at four arms each, and the characteristic impedances of the input ports and output ports microstrip lines are all Z_0 = 50 Ω. For the purpose of elaborate description, the equivalent circuits of the differential mode and common mode are
Fig. 1. (a) The ideal circuit of the balanced filter with microstrip/slotline transition, (b) equivalent circuit for the differential mode, and (c) equivalent circuit for the common mode.

When ports 1 and 1' are excited by differential mode signals, out-phase signals will be changed into in-phase signals after one kind phase of the signals pass through the phase converter. We can obtain \( \theta_{12} \left( f_0 \right) = 360^\circ \), \( \theta_{12} \left( f_0 \right) = 360^\circ \left( \theta = 90^\circ \right) \) at the center frequency \( f_0 \), then they are combined at the center point and propagate to output ports. The differential-mode circuit illustrated in Fig. 1 (b) shows that it is half of the circuit in Fig. 1 (a) due to the symmetrical structure. Besides, a virtual open appears at the joint of four arms. This means that the differential-mode circuit comes out a typical bandpass filter. The \( ABCD \) matrix of the transmission lines and the shorted lines are:

\[
M_{s1} = \begin{bmatrix} 1 & 0 \\ 1/JZ_{s1} \tan \theta & 1 \end{bmatrix},
\]

\[
M_{s2} = \begin{bmatrix} 1 & 0 \\ 1/JZ_{s2} \tan \theta & 1 \end{bmatrix},
\]

\[
M_1 = \begin{bmatrix} \cos \theta & jZ_1 \sin \theta \\ j \sin \theta / Z_1 & \cos \theta \end{bmatrix},
\]

\[
M_2 = \begin{bmatrix} \cos 2\theta & jZ_2 \sin 2\theta \\ j \sin 2\theta / Z_2 & \cos 2\theta \end{bmatrix},
\]

\[
M_{1/2} = \begin{bmatrix} \cos (\theta/2) & jZ_1 \sin(\theta/2) \\ j \sin(\theta/2) / Z_1 & \cos(\theta/2) \end{bmatrix},
\]

\[
M_1 = \begin{bmatrix} -1 & 0 \\ 0 & -1 \end{bmatrix}.
\]

From port 1 to port 2, the \( ABCD \) parameter matrix can be defined as \( M_0 \times M_1 \times M_2 \times M_3 \times M_{10} \times M_{12} \times M_{10} \times M_{12} \times M_1 \) derived from Equations (1) to (6). Then we can get the frequency response form \( ABCD \) to \( Y \)-parameter conversions. The poles, the insertion loss and bandwidth can be controlled by changing the transmission line impedances.

The simulated frequency responses of Fig. 1 (b) are shown in Fig. 2 (Simulated with ANSYS Designer v3.0). We can find that the transmission zero at both sides of the passband will not change with the characteristic impedance \( Z_{s1} \) and \( Z_{s2} \) But \( Z_{s1} \) increased can improve performance of \( |S_{dd11}| \) obviously in the passband. Without the two shorted lines of \( Z_{s1} \), transmission poles decrease to three. As to \( Z_{s2} \), it influences the bandwidth deeply, bigger \( Z_{s2} \) and bandwidth wider. But when it comes to the return loss in-band, the performance shows better first and then becomes worse with \( Z_{s2} \) increased. Although the bandwidth is much bigger, \( |S_{dd11}| \) will down to 10 dB.

Fig. 2. Simulated frequency responses of Fig. 1 (b): (a) \( |S_{cc11}| \) and \( |S_{cc21}| \), \( Z_1 = 50 \Omega, Z_2 = 70 \Omega, Z_{s2} = 70 \Omega \), and (b) \( |S_{cc11}| \) and \( |S_{cc21}| \), \( Z_1 = 50 \Omega, Z_2 = 70 \Omega, Z_{s1} = 120 \Omega \).

When ports 1 and 1' are excited by common mode signals, similarly, in-phase signals will be changed into out-phase signals after one kind phase of the signals pass through the phase converter. That means input signals will be cancelled out at the center point, due to \( \theta_{12} \left( f_0 \right) = 180^\circ \) and \( \theta_{12} \left( f_0 \right) = 360^\circ \left( \theta = 90^\circ \right) \) at the center frequency \( f_0 \). In this case, the common mode circuit in Fig. 1 (c) is generated and there is a virtual short appears at the joint of four arms. Because common mode signals canceled at the joint, the circuit presents a stopband filter over the whole frequency band. So common mode signals suppression in/out-of-band for the differential mode is easy to achieve.

Figures 3 (a)-(b) plot the simulated frequency responses of Fig. 1 (c). As the two plots show that \( |S_{cc11}| \) is almost steady over whole frequency band despite the characteristic impedance \( Z_{s1} \) and \( Z_{s2} \) change. But \( |S_{cc21}| \) increased follows \( Z_{s1} \) or \( Z_{s2} \) increased. The performance becomes worse. Moreover, without the two shorted lines of \( Z_{s1} \) or \( Z_{s2} \), \( |S_{cc11}| \) and \( |S_{cc21}| \) becomes worse than they existed.
B. Balanced filter analysis with λ/4 shorted coupled line

Figure 4 (a) shows ideal circuit of the balanced filter, which is a central symmetric structure of all components. Figures 4 (b) and (c) are the equivalent differential/common mode circuits to be analyzed accordingly. We use λ/4 shorted coupled line in this circuit and its equivalent circuit model gives in Fig. 5. The ABCD matrix can be written as follow:

\[
\begin{bmatrix} A & B \\ C & D \end{bmatrix} = \begin{bmatrix} 1 & 0 \\ Y_{11} + Y_{12} & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 \\ Y_{22} + Y_{12} & 1 \end{bmatrix} = \begin{bmatrix} -1 & 0 \\ 0 & -1 \end{bmatrix} \begin{bmatrix} Y_{11}/Y_{12} & 1/Y_{12} \\ (Y_{11}^2 - Y_{12}^2)/Y_{12} & Y_{11}/Y_{12} \end{bmatrix}, \quad (7)
\]

\[
\begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix} = \begin{bmatrix} \cos(180^\circ) & jZ \sin(180^\circ) \\ jY \sin(180^\circ) & \cos(180^\circ) \end{bmatrix}, \quad (8)
\]

\[
\begin{align*}
Y_{11} &= Y_{22} = -j(Y_{ee} + Y_{oo})/(2 \cdot \tan \theta), \\
Y_{12} &= -j(Y_{eo} - Y_{oe})/(2 \cdot \sin \theta). \quad (9)
\end{align*}
\]

From the upper Equations from (7) to (10), we can draw a conclusion that the two λ/4 shorted coupled lines are as 180° phase shifter and will not change with frequency. The characteristic impedances of the input ports and output ports microstrip lines are all Z_0 = 50 Ω.

As discussed in Part A, a virtual open appears at the joint of four arms for differential-mode circuit in Fig. 4 (b). Due to the signals combined at the center point, it comes out a typical bandpass filter when ports 1 and 1’ are excited by differential mode signals. Furthermore, a virtual short appears at the joint of four arms for common-mode circuit in Fig. 4 (c). Due to the signals canceled at the center point, it comes out a stopband filter when ports 1 and 1’ are excited by common mode signals.

Figures 6 (a) and (b) are the simulation results at different value of k. k is coupling coefficient of the two shorted coupler line. It is apparent that the differential mode and common mode performance show better with k increases. But there is a problem that big coupling coefficient is hard to realize in monolayer PCB microstrip line structure. In order to get tight coupling, we introduce a slot under the coupled lines [12]. For a simple coupled line, even-mode impedance is mainly depend on capacitance of metal strips to ground, while odd-mode impedance is depend on capacitance of metal strips and coupled lines to ground. So even-mode and odd-mode capacitance will decrease with the slot at the same time. In addition, placing a rectangular conductor inside the slot can make sure that odd-mode decrease in company with even-mode.

Fig. 3. Simulated frequency responses of Fig. 1 (c): (a) |S_{cc11}| & |S_{cc21}|, Z_1 = 50 Ω, Z_2 = 70 Ω, Z_2 = 70 Ω, and (b) |S_{cc11}| & |S_{cc21}|, Z_1 = 50 Ω, Z_2 = 70 Ω, Z_2 = 120 Ω.

Fig. 4. (a) The ideal circuit of the balanced filter with λ/4 shorted coupled line, (b) equivalent circuit for the differential mode, and (c) equivalent circuit for the common mode.

Fig. 5. Equivalent circuit model of the λ/4 shorted coupled line.

Fig. 6. Simulated frequency responses of Figs. 4 (b) and (c): (a) |S_{dd21}| & |S_{dd11}|, Z_1 = 55 Ω, Z_2 = 65 Ω, Z_1 = 120 Ω, Z_2 = 120 Ω, and (b) S_{cc21} & S_{cc11}, Z_1 = 55 Ω, Z_2 = 65 Ω, Z_1 = 120 Ω, Z_2 = 120 Ω.
C. Proposed to balanced bandpass filters

Based on the above discussions and the simulated results in Part A and B of Section II, the 3-dB bandwidths of the two balanced filters are chosen as 102% and 103%, and the final parameters in Figs. 1, 4 are: $Z_0 = 50 \Omega, Z_1 = 50 \Omega, Z_2 = 90 \Omega, Z_3 = 114 \Omega, Z_4 = 104 \Omega, Z_5 = 50 \Omega, Z_6 = 45 \Omega, Z_7 = 60 \Omega, Z_8 = 112 \Omega, Z_9 = 112 \Omega, Z_{10} = 200 \Omega, Z_{11} = 30 \Omega$. Figure 7 presents the geometries of the two balanced filters (78.94 mm $\times$ 56.83 mm, 66.2 mm $\times$ 37.28 mm) are: $l_1 = 16.85 \text{ mm}, l_2 = 17.08 \text{ mm}, l_3 = 9.83 \text{ mm}, s_1 = 16.85 \text{ mm}, s_2 = 18.42 \text{ mm}, g_1 = 0.3 \text{ mm}, l_g = 7 \text{ mm}, a = 9.5 \text{ mm}, b = 2 \text{ mm}, w_0 = 1.56 \text{ mm}, w_1 = 1.36 \text{ mm}, w_2 = 0.23 \text{ mm}, w_3 = 0.3 \text{ mm}, w_4 = 0.44 \text{ mm}, d_1 = 0.6 \text{ mm}, d_2 = 1 \text{ mm}, d_3 = 10 \text{ mm}; l_1 = 16.7 \text{ mm}, l_2 = 17.7 \text{ mm}, l_3 = 8.05 \text{ mm}, s = 0.15 \text{ mm}, s_1 = 16.7 \text{ mm}, s_2 = 16.7 \text{ mm}, q_1 = 7.95 \text{ mm}, q_2 = 14.7 \text{ mm}, w_0 = 1.36 \text{ mm}, w_1 = 2.1 \text{ mm}, w_2 = 0.24 \text{ mm}, w_3 = 1 \text{ mm}, w_4 = 0.26 \text{ mm}, g_1 = 1.6 \text{ mm}, g_2 = 0.3 \text{ mm}, d_1 = 0.6 \text{ mm}, d_2 = 0.8 \text{ mm}.

The simulated results and photographs of the two balanced filters with two different 180° phase shifter are shown in Figs. 8 and 9 (Simulated with ANSYS HFSS v.11.0). As to the balanced filter with microstrip/slotline transition, the in-band insertion loss of differential mode is less than 1 dB with 3-dB bandwidth approximately 102% from 1.47 GHz to 4.53 GHz. And insertion loss of common mode is greater than 23 dB (0-5.7 GHz, 1.9f_0), which manifested good wideband suppression. Furthermore, for the balanced filter with λ/4 shorted coupled line, 3-dB bandwidth is about 103% from 1.52 GHz to 4.61 GHz, meanwhile, the in-band insertion loss of differential mode is less than 1.3 dB. And insertion loss of common mode is greater than 15 dB (0-7.5 GHz, 2.5f_0).

III. RESULTS AND DISCUSSION

Figures 8 and 9 present the measured results and photographs of the two balanced filters. We can see from Fig. 8 (a), 3-dB bandwidth is approximately 100% (1.47-4.48 GHz), the passband return loss is greater than 12.5 dB and insertion loss is less than 1.1 dB of the differential mode; for the common mode in Fig. 8 (b), a broadband rejection (0-7.5 GHz, 2.5f_0) is achieved which up to 20 dB. The results of the second balanced filter are shown in Fig. 9, 3-dB bandwidth is approximately 100% (1.48-4.48 GHz), the passband return loss is greater than 11 dB and insertion loss is less than 1.3 dB of the differential mode; for the common mode in Fig. 9 (b), a broadband rejection (0-7.4 GHz, 2.46f_0) is achieved which up to 15 dB. Aforementioned two structures, the in-band performance of differential mode has a bit discrepancy is mainly caused by the fabrication inaccuracy and errors of measurement.
Fig. 9. Measured and simulated results of the balanced filter with \( \lambda/4 \) shorted coupled line. (a) Differential mode and (b) common mode.

IV. CONCLUSION

Two ultra-wideband balanced filters based on transversal signal-interference concepts are proposed in this paper. Wideband common mode signals suppression can be implemented conveniently for the filters with two structures, microstrip/slotline transition and quarter-wavelength shorted coupler lines. Then 180° phase shift realized therefore. It shows good performance for common mode suppression and little insertion loss of differential mode.

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Novel Pentagonal Dual-Mode Filters with Adjustable Transmission Zeros

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Abstract — Novel compact pentagonal dual-mode filters with open stub is presented. The field patterns of this type of resonators are investigated using full-wave electromagnetic simulations. The technique of utilizing capacitive and inductive source-load coupling to improve the performance of filters is explored. Advantages of using this type of filter are not only its compact size, but also its transmission zeros that can be independently controlled. Then, two dual-mode bandpass filters are designed, fabricated and tested to validate the design concept. Both simulated and measured results are presented.

Index Terms — Bandpass filter, dual-mode, transmission zero.

I. INTRODUCTION

Bandpass filter (BPF) is one of the most important components in microwave circuits. To meet the requirement of modern microwave communication systems, microwave BPFs with compact size and high performance are in demand. The dual-mode resonators are attractive because each resonator can be used as a doubly tuned circuit, and therefore, the number of resonators is reduced by half, resulting in a compact size. Wolff first demonstrated a microstrip dual-mode filter in 1972 [1]. Since then, dual-mode microstrip filters have been widely used in communications systems [2-3]. Among them, E-shaped microstrip resonators and filters have been originally reported in [4]. More recently, the E-shaped resonator was modeled as a dual-mode resonator in [5]. However, it was difficult to control the location of the transmission zeros. Circular dual-mode filter based on source-load coupling was proposed in [6]. By introducing a capacitive cross-coupling between the input and output ports so that additional zero can be generated, but transmission zeros cannot be independently controlled. The open stub dual-mode filter with adjustable transmission zeros by inductive source-load coupling (S-L coupling) was firstly proposed in [8]. Two novel bandpass filters with multiple transmission zeros using only four open/shorted stubs was proposed in [9]. The out-of-band transmission zeros can be adjusted easily by only changing the electrical length of the four open/shorted stub. But the resonator occupied a large circuit area, size reduction is becoming a major design consideration for practical applications.

In this paper, two compact bandpass filters with a pentagonal dual-mode resonator and S-L coupling are introduced. The reduction of the size is achieved by using a pentagonal dual-mode resonator. Furthermore, with a cross-coupling between the input and output feed lines, two tunable transmission zeros are obtained that can be controlled independently by the amount of the capacitive or inductive S-L coupling. The proposed bandpass filter shows a good stopband rejection because of the two tunable transmission zeros. Two exemplary filters verify the feasibility of the technique.

II. DUAL-MODE PENTAGONAL RESONATOR

Figure 1 shows the geometry of proposed dual-mode filters with the pentagonal open-loop resonator by open stub loaded. The filter comprises an improved pentagonal half-wavelength resonator and a hexagonal open loaded stub. The S-L coupling include two types, i.e., capacitive and inductive S-L coupling.

Fig. 1. Layout of the proposed pentagonal dual-mode bandpass filter: (a) capacitive S-L coupling, and (b) inductive S-L coupling.

Figure 2 shows a T-shaped resonator model with a half-wavelength resonator and a shunt open stub, where \( z_1 \) is the characteristic impedance of the half-wavelength resonator.

\[ z_1 = \frac{c}{2\pi f} \]

where \( c \) is the speed of light and \( f \) is the frequency of operation.
wavelength resonator with the electrical length \( \theta_1 = 90^\circ \), and \( z_2 \) the characteristic impedance of the open stub with electrical length \( \theta_2 \). \( C_1 \) is the coupling capacitance between resonator and feed line. As \( \theta \) increases, the odd resonant frequencies are fixed, while the even resonant frequencies and the transmission zero frequency keeps decreasing. When \( \theta_2 \) is slightly smaller/larger than \( 90^\circ \), the even resonant frequency is slightly beyond/below fundamental frequency of the half-wavelength resonator, respectively. The proposed resonator then becomes a dual-mode resonator with a transmission zero close to the resonant frequency [7]. If the S-L coupling is introduced, an additional transmission zero is presented. Then, the proposed filter will show two transmission zeros.

![Fig. 2. Schematic of a half-wavelength resonator with an open stub.](image)

For a compact size, the half-wavelength length line is bent to a pentagon, and the uniform shunt open stub is also replaced by a hexagonal stepped-impedance open stub. The open hexagonal stub at the centre of the half-wavelength resonator introduces another transmission pole near the fundamental frequency. This new pole and the fundamental frequency pole of the half-wavelength resonator make the proposed structure a dual-mode resonator. So this symmetric structure can support two modes, i.e., an even mode and odd mode.

The commercially available full-wave electromagnetic simulators (HFSS) were used to characterize the electric field patterns for a dual-mode open-loop resonator. HFSS uses the finite element method (FEM) to analyze the electromagnetic characteristics of 3D objects. The basic process of solving the problem by FEM includes three parts, which are the mesh discretization of the object, the solution of the simultaneous matrix equations related the mesh and the postprocessing calculation of the problem.

It can be seen that the whole structure is symmetrical with the center point, so the center point is modeled as the origin point and the mirror operation is applied. The physical excitation of the filter is by the coaxial line with the TEM wave. In order to use the wave-guide port in the simulation code, the port surface must cover more than ninety-five percent of the TEM field. It is assumed that the width of the excitation microstrip is \( w \) and the thickness of the dielectric layer is \( h \). The height of the wave port is generally set to 6~10h. When \( w > h \), the width of the wave port is set to about 10w; when \( w < h \), the width of the wave port is set to about 5w. Finally, the height and width of the wave port of are 10h and 10w in this paper.

According to the standard which is set up by user, HFSS simulation code uses adaptive mesh generation technology. The solution frequency of the meshing is generally set at the center frequency of the filter. After each new mesh subdivision, HFSS will compare the results of the S parameters with the old one. If the error is less than the set criterion, it is shown that the result is convergent and the adaptive process will end. The dimensions are optimized by a full-wave simulation to take all the discontinuities into consideration.

Figure 3 depicts the simulated electric field vector between the metal strip and ground plane at the resonance frequency. The electric field pattern of the odd mode is illustrated in Fig. 3 (a), where the maxima of the field are located along the left and right arms. The field distribution is similar to that of a half-wavelength single-mode resonator. As a consequence, the open stub does not affect the resonant frequency of the odd mode. Figure 3 (b) shows the electric field pattern of the even-mode, where the maxima of the field are moved to the open stub and both side of the arms. Moreover, it is observed from the direction of the electric field vector that the field is symmetric with respect to the symmetry axis. Hence, changing the dimension of the open stub makes the resonant frequency of the even mode shift.

![Fig. 3. Simulated electric field patterns for a dual-mode pentagonal open-loop resonator: (a) odd mode and (b) even mode.](image)
the resonant frequency of the even mode decreases from 3.58 to 3.45 GHz, while that of the odd mode hardly change.

![Graph](image1)

**Fig. 4.** Simulated resonance frequencies of the two modes against: (a) $l_2$, where $r_2 = 5$ mm, and (b) $r_2$, where $l_2 = 15$ mm.

### III. BANDPASS FILTERS USING DUAL-MODE PENTAGONAL RESONATOR

Figure 1 shows the layout of the pentagonal dual-mode open loop BPF. It consists of the capacitive S-L coupling and inductive S-L coupling filter. The gap between the resonator and coupling arms was selected in consideration of strong coupling and etching tolerance. The characteristic impedance of the input/output microstrip is taken as 50 ohm. The length of the S-L coupling line is $l_p$. The gap of the S-L coupling line is $s_2$.

As illustrated in [6], the open stub dual-model filter has an interesting property. There is an inherent finite-frequency transmission zero when the two modes split. If $f_{\text{even}} < f_0 < f_{\text{add}}$, the inherent transmission zero would be in the lower stopband. If $f_{\text{add}} < f_0 < f_{\text{even}}$, the inherent transmission zero would be in the upper stopband. $f_0$ is the center frequency.

For further improving the filters’ performance, S-L coupling is introduced to generate an additional zero. Using capacitive and inductive S-L coupling technique, the response with two adjustable zeros can be obtained for the proposed dual-mode filters. The locations of these two transmission zeros can be controlled by transforming the type and amount of the source-load cross-coupling. For the capacitive S-L coupling, an additional transmission zero shows in the upper stopband. For the inductive S-L coupling, an additional transmission zero shows in the lower stopband. Two exemplary filters verify the feasibility of the new technique. We take $f_{\text{add}} < f_0 < f_{\text{even}}$ for instance.

#### A. Dual-mode filter with capacitive S-L coupling

Filter A exhibits both the inherent transmission zero and the additional zero in the upper stopband. Here, the capacitive S-L coupling is introduced to generate the additional zero. The locations of the additional zero may be controlled by adjusting the values of the $s_2$ and $l_p$. As shown in Fig. 5, when $s_2$ decreases from 1.1 to 0.7 mm, the inner transmission zero almost doesn’t change, and the outer transmission zero moves toward the passband edge. As shown in Fig. 6, when $l_p$ increases from 1 to 2.4 mm, the inner transmission zero almost doesn’t change, while the outer transmission zero moves toward the passband edge. Therefore, the $s_2$ and $l_p$ can be selected to meet the required filter selectivity.

![Graph](image2)

**Fig. 5.** Simulated scattering parameters of the filter A for five values of $s_2$.

![Graph](image3)

**Fig. 6.** Simulated scattering parameters of the filter A for four values of $l_p$. 
The dimensions are optimized by a full-wave simulator to take all the discontinuities into consideration. The designed filter is fabricated on the substrate Rogers RO4003, which relative dielectric constant is 3.38 and the thickness is 0.508 mm. Figure 7 shows the photograph of the fabricated filter A. Both measured and simulated results are plotted in Fig. 8. As seen from the measured results, at the center frequency of 2.75 GHz, the 3 dB fractional bandwidth is about 8%. Two transmission zeros are realized at 3.2 and 4.75 GHz. The insertion loss is less than 2 dB in passband, and the minimum of insertion loss is 1.5 dB. The return loss is greater than 20 dB in passband. Simulation results almost agree with the measured results.

![Photograph of the fabricated filter A](image)

**Fig. 7.** Photograph of the fabricated filter A. Geometric parameters of the filters are $w = 1$, $r_2 = 5$, $l_2 = 11.55$, $l_p = 2$, $s_2 = 1$ and $s = 0.22$. All are in mm.

![Simulated and measured frequency responses of the filter A](image)

**Fig. 8.** Simulated and measured frequency responses of the filter A.

### B. Dual-mode filter with inductive S-L coupling

Filter B demonstrates a filter characteristic with the inherent finite frequency zero located at the upper side of the passband, while the additional zero at the lower side. This is because $f_{add} < f_0 < f_{even}$ for the proposed filter. As it has been noted, the inherent zero is at upper side of passband. The inductive S-L coupling is introduced to generate an additional zero at lower side of passband. As shown in Fig. 9 and Fig. 10, when $s_2$ decreases from 3.5 to 0.5 mm, the inherent transmission zero changed little, while the additional transmission zeros increases from 0.85 to 1.9 GHz. When $l_p$ increases from 1 to 7 mm, the inherent transmission zero also changed little, while the additional transmission zeros increases from 0.85 to 1.65 GHz. Thus, a sharper fall-off at both lower and upper passband edge may be achieved by adjusting the $s_2$ and $l_p$.

![Simulated scattering parameters of the filter B for four values of $s_2$](image)

**Fig. 9.** Simulated scattering parameters of the filter B for four values of $s_2$.

![Simulated scattering parameters of the filter B for four values of $l_p$](image)

**Fig. 10.** Simulated scattering parameters of the filter B for four values of $l_p$.

Figure 11 shows the photograph of the fabricated filter B. The simulated and measured frequency responses are shown in Fig. 12. The simulated results show that the filter B operated at 3 GHz and a 3 dB fractional bandwidth of 5.3%. Two transmission zeros are located at 1.55 GHz and 3.3 GHz respectively. The minimum insertion loss is about 0.6 dB, and the return loss is greater than 20 dB in passband. The measured minimum insertion loss is about 2 dB, and the return loss is greater than 16 dB in passband. The measured results meet the simulation results well.
Fig. 11. Photograph of the fabricated filter B. Geometric parameters of the filters are \( w = 1 \), \( r_1 = 8 \), \( l_2 = 11.55 \), \( l_3 = 8 \), \( l_4 = 4 \), \( s_2 = 1.5 \) and \( s = 0.18 \). All are in mm.

Fig. 12. Simulated and measured frequency responses of the filter B.

IV. CONCLUSION

The application of capacitive and inductive S-L coupling has been studied intensively in this paper. The novel pentagonal dual-mode filter loaded by hexagonal open stub is presented. By S-L coupling, the proposed filters exhibits two transmission zeros. It reveals that a quasi-elliptic response with two adjustable transmission zeros can be obtained easily. The proposed structure and design method is verified by two exemplary filters.

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Miniaturized Microstrip Suppressing Cell with Wide Stopband

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Abstract — In this paper, based on simple stepped impedance structures a methodology is followed to design a very compact size lowpass filter (LPF) as a suppressing cell. The proposed suppressing cell consists of stepped impedance ladder-type resonators, which provides a wide stopband by creating the transmission zeros to its frequency response. The proposed suppressing cell has clear advantages like low insertion loss in the passband and suitable roll-off. The designed LPF is fabricated on a microstrip layout and tested. The measured results highlight the efficiency of the filter and have a good agreement with the simulated results. With mentioned expression, the fabricated LPF can be a powerful block as a suppressing cell to implement in the distributed high frequency circuits.

Index Terms — Lowpass filter, microstrip, miniaturized, stepped impedance structure, suppressing cell.

I. INTRODUCTION

Design of the high frequency circuits are highly extended and demanded, in the modern technologies. Handset devices and wireless communications depend on the high frequency circuits and the suppression of the spurious signals is an important requirement in this field [1]. The position of the lowpass filters is undeniable in order to block and suppress the unwanted harmonics. From the introducing of microstrip technology until now, many microstrip structures and many methods have been presented to design the high efficient LPFs [1-14]. Planar structures are widely used for their simple fabrication and design in [1-13], which have sharp transition band and wide stopband, but these LPFs suffers from the big size, which makes them inappropriate as the suppressing cells to use in hybrid high frequency circuits, in addition to their almost low attenuation levels in their stopbands.

A miniaturized LPF with very sharp roll-off has been designed to eliminate the unwanted harmonics for a Wilkinson power divider in [14], which has a so narrow stopband bandwidth. In [15], using defected ground structures (DGS), an Elliptic-function LPF has been fabricated with sharp roll-off, this technique design not only is not easy to implement but also has disadvantages such as large circuit size and narrow stopband width. A lowpass filter has been designed in [16] with sharp roll-off as a miniaturized LPF, but it is not so compact and has a very narrow rejection bandwidth. In [17], the fabricated LPF has a compact structure, although the stopband region is not wide enough. A wide stopband LPF has been presented in [18], but this filter suffers from its high insertion loss in the passband and its large circuit size. In [19-24], the low value of maximum variation of the group delay in the passband has been introduced as an effective factor in LPF designs, which it tried to be improved in the proposed work. Also, the simple geometry and topology of the designed filter is a significant specification, which can be effective in fabrication and implementation. Therefore the design of a simple structure and high efficient LPF is the main objective of this paper.

In this paper, using stepped impedance microstrip stubs and ladder-type structures, a miniaturized LPF with good rejection bandwidth is designed. The LPF has - 3 dB cut-off frequency at 4 GHz. The rejection band is achieved from 5 to 23.3 GHz. A simple methodology is used to design this filter that follows in the next session. All the simulations are done using Agilent Advanced Design System (ADS) software, and all of the microstrip layouts are designed and fabricated on RT/Duriod5880 substrate with dielectric constant ($\varepsilon_r$) of 2.22, the thickness of 0.508 mm and the loss tangent of 0.0009.
II. SUPPRESSING CELL DESIGN

At the first step, Elliptic function resonator layout has been selected and expanded using high and low impedance lines, as shown in Fig. 1. The physical lengths of the low-impedance and high-impedance lines are calculated using below equations [10]:

\[ L_i = \frac{\theta_i Z_0}{2 \pi f_i} \]  \hspace{1cm} (1)

\[ C_i = \frac{\theta_i Z_0}{2 \pi f_i} \]  \hspace{1cm} (2)

\[ d_{L1} = \frac{\lambda_{G1}}{2 \pi} \sin^{-1}\left(\frac{2 \pi f_i L_1}{Z_0 L_1}\right) \]  \hspace{1cm} (3)

\[ d_{C1} = \frac{\lambda_{G1}}{2 \pi} \sin^{-1}\left(\frac{2 \pi f_i C_i Z_{oc1}}{Z_0 L_1}\right) \]  \hspace{1cm} (4)

where, \( Z_{oc1} \) and \( Z_{oc2} \) are corresponded to the impedance transmission lines with low and high impedance, respectively. \( g \) and \( go \) are the element values of each part of the prototype layout, \( \lambda_{G1} \) and \( \lambda_{G1} \) are the guided wavelengths of high and low impedance lines, respectively. With considering the Fig. 1, the ABCD matrix for the proposed resonator can be written as:

\[
\begin{bmatrix}
A & B \\
C & D
\end{bmatrix}
= \begin{bmatrix}
1 & \frac{Z_{oc2}}{2} \\
0 & 1
\end{bmatrix} \times
\begin{bmatrix}
1 & 0 \\
\frac{Z_{oc3}}{2} & 1
\end{bmatrix} \times
\begin{bmatrix}
Y_{oc1} + Y_{oc2} + Y_{oc1} + Y_{oc2} \\
Y_T + 1
\end{bmatrix}.
\]  \hspace{1cm} (5)

where, \( Y_{oc1}, Y_{oc2}, Y_{oc1}, \) and \( Y_{oc2} \) are the admittances of the high and low impedance transmission lines, which are indicted in Fig. 1. With calculation, the Equation (5) can be simplified in:

\[
\begin{bmatrix}
A & B \\
C & D
\end{bmatrix}
= \begin{bmatrix}
1 & \frac{Z_{oc3}}{2} \\
0 & 1
\end{bmatrix} \times
\begin{bmatrix}
1 & 0 \\
Y_T & 1
\end{bmatrix} = \begin{bmatrix}
1 + Z_{oc3} Y_T & Z_{oc3} \\
Z_{oc3} & 1
\end{bmatrix},
\]  \hspace{1cm} (6)

where, \( Y_T = Y_{oc1} + Y_{oc2} + Y_{oc1} + Y_{oc2} \).

The proposed resonator has a symmetric shape as illustrated in Fig. 1; so clearly, it is expected that the resonator must have a reciprocal response. Therefore, the determinant of the ABCD matrix must be equal to 1. The determinant of the ABCD matrix can be calculated from Equation 6 as:

\[
\Delta = \begin{bmatrix}
A & B \\
C & D
\end{bmatrix}
= 1 + Z_{oc3} Y_T - Z_{oc3} Y_T = 1. \hspace{1cm} (7)
\]

Then, the Equation (7) validates the achieved ABCD matrix of the proposed resonator. The simulated S-parameters of the proposed resonator are shown in Fig. 2. As considered, in the passband the resonator has a transmission pole at 2.87 GHz with attenuation level of -15.6 dB and in the stopband the resonator has a transmission zero at 8.5 GHz with attenuation level of -64.3 dB with -3 dB cut-off frequency at 5.6 GHz. The resonator has a smooth passband region; although it has a narrow rejection bandwidth and gradual response in the transition band.

For this structure, the LC model for the proposed resonator is extracted as shown in Fig. 3. In this model, C represents the overall capacitance of low impedance stubs (\( Z_{oc1}, Z_{oc2} \)) respect to ground; L1 represents the overall inductance of high impedance lines (\( Z_{oc1}, Z_{oc2} \)); L represents the inductance of high impedance line of \( Z_{oc3} \).

A comparison of the S21 parameters of this model and layout is shown in Fig. 4. The values of the lumped elements are illustrated in this figure, which are the conventional values of an Elliptic function 3 order resonator with -3 dB cut off frequency at 5.6 GHz. Transfer function for the calculation of the first transmission zero has been extracted using the proposed LC model as below:

\[
T(s) = \frac{(9.793 \times 10^3) s^2 + 2.789 \times 10^3}{s^3 + (3.39 \times 10^3) s^2 + (1.539 \times 10^3) s + 2.789 \times 10^3}. \hspace{1cm} (8)
\]

In this equation, the coefficients of the polynomials of the numerator and denominator depends on the capacitances and inductances values of the LC model.
and the location of transmission zero can be adjusted by changing these values. For example, as seen in Fig. 5, by increasing the lengths of $d_{L1}$ and $d_{L2}$ from 1 mm to 1.5 mm, due to increment of the capacitance of $C$, which is the total capacitance of $Z_{OL1}$ and $Z_{OL2}$, first transmission zero moves from 8.4 GHz to 7.2 GHz. Also, by decreasing the lengths of $d_{L1}$ and $d_{L2}$ from 1 mm to 0.5 mm, due to decrement of the capacitance of $C$, first transmission zero moves to 11.4 GHz.

To modify the frequency response, another resonator can be cascaded to the previous one with the same dimensions as shown in Fig. 6. These dimensions are as follows: $W = 1$ mm, $W_1 = 1.1$ mm, $d = 4.8$ mm, $d_2 = 0.5$ mm and $d_3 = 11$ mm. Using Equation (7), the ABCD matrix for the proposed cascaded resonator can be written as:

$$\begin{bmatrix} A & B \\ C & D \end{bmatrix} = \begin{bmatrix} 1 + Z_{oc3} Y_{TS} & Z_{oc3} \\ Y_{TS} & 1 \end{bmatrix} \times \begin{bmatrix} 1 + Z_{oc3} Y_{TS} & Z_{oc3} \\ Y_{TS} & 1 \end{bmatrix},$$

(9)

The simulated S12-parameter of the proposed cascaded resonator is shown in Fig. 7. As seen, by moving the existence transmission zero to the lower frequency at 6.8 GHz and creating a new transmission zero at 10.7 GHz; the -3 dB cut-off frequency moves to 5 GHz with sharper transition band. Also, wider rejection band can be obtained. It has a stopband bandwidth from 5.9 GHz to 13 GHz for the attenuation level of -20 dB. But, the rejection band is so narrow yet. To extend the stopband width enough, the proposed cascaded resonator can be improved by adding another resonator, symmetrically with same dimensions as the proposed filter, as shown in Fig. 8. Thus, the stopband bandwidth has been improved up to 157% of the previous rejection band. Also, two stubs are added to the feeding lines at input and output to match the proposed filter to 50 Ω coaxial line. The ABCD matrix for the proposed cascaded symmetric resonator can be written using symmetric rules for impedances and with considering Equation 9 as:

$$\begin{bmatrix} A & B \\ C & D \end{bmatrix} = \begin{bmatrix} 1 + 2Z_{oc3} Y_{TS} & 2Z_{oc3} \\ Y_{TS} & 1 \end{bmatrix} \times \begin{bmatrix} 1 + 2Z_{oc3} Y_{TS} & 2Z_{oc3} \\ Y_{TS} & 1 \end{bmatrix},$$

(10)

where, $Y_{TS} = \frac{1}{2}(Y_{oc1} + Y_{oc2}) + 2(Y_{oc1} + Y_{oc2})$.
The designed LPF has been fabricated on RT/Duriod5880 substrate with dielectric constant (\(\varepsilon_r\)) of 2.22, the thickness of 0.508 mm and the loss tangent of 0.0009 and is illustrated in Fig. 9. The mentioned dimensions in the Fig. 8 are determined using equations (1-10) and tuned using ADS as a tuning tool. These dimensions are as follows: W = 1 mm, W1 = 1.1 mm, W2 = 0.1 mm, W3 = 1.5 mm, d = 4.8 mm, d1 = 0.2 mm, d2 = 0.5 mm, d3 = 11 mm and d4 = 2 mm. The measurements are done using HP8757A network analyzer. The simulated and measured S-parameters of the fabricated LPF are shown in Fig. 10. As seen, the -3 dB cut-off frequency is placed at 4 GHz. The rejection band is extended from 5 to 23.3 GHz with corresponding attenuation level of 16.6 dB. Also, the return loss is about 0 dB in the more space of the rejection band. The maximum insertion loss in the 90% of the passband region is 0.1 dB, where the maximum return loss is 15.6 dB. The physical size of the fabricated circuit, which occupies very small area, is only 11 mm × 5.5 mm = 60.5 mm² (0.197\(\lambda_g\)×0.098\(\lambda_g\)), where \(\lambda_g\) is the guided wave length at -3 dB cut-off frequency. A comparison between the characteristics of the proposed LPF and some referred works are shown in Table 1. In this table, RL (dB) and IL (dB) are the maximum return loss and insertion loss in the passband region, respectively.

As can be seen in Table 1, the proposed LPF has the smallest size (60.3 mm) and the best insertion loss (0.1) among the referred filters. The good specifications of the rejection band and the small size are the important factors, which show that the proposed filter can be used in compact modern high frequency circuits as a suppressing cell in order to suppress the unwanted harmonics and interferences. The group delay of a microwave filter has a relationship to the insertion loss of a filter and design of a filter with flat group delay in the passband region is desirable [18-24]. As seen in Fig. 11, maximum variation of the measured group delay in the passband has not a significant variation and has a dispensable value and is only 0.23 ns. Table 2 shows a comparison between the maximum variation of the measured group delay in the passband for the proposed LPF and some referred works with reported group delay. As illustrated, the proposed filter has the best performance in the case of group delay.

### Table 1: Performance comparisons between the proposed LPF and some referred works

<table>
<thead>
<tr>
<th>Ref.</th>
<th>(f_c) (GHz)</th>
<th>SF</th>
<th>SBW/(f_c)</th>
<th>Size ((\text{mm}^2))</th>
<th>RL dB</th>
<th>IL (dB)</th>
</tr>
</thead>
<tbody>
<tr>
<td>2</td>
<td>5.45</td>
<td>2</td>
<td>5.7</td>
<td>221</td>
<td>15</td>
<td>0.12</td>
</tr>
<tr>
<td>3</td>
<td>2</td>
<td>1</td>
<td>8.4</td>
<td>482</td>
<td>10</td>
<td>1.00</td>
</tr>
<tr>
<td>4</td>
<td>1.78</td>
<td>2</td>
<td>2.8</td>
<td>643.7</td>
<td>~10</td>
<td>0.30</td>
</tr>
<tr>
<td>5</td>
<td>1.67</td>
<td>1</td>
<td>5.9</td>
<td>100</td>
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<td>0.50</td>
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<tr>
<td>6</td>
<td>1</td>
<td>2</td>
<td>4.1</td>
<td>638.4</td>
<td>20</td>
<td>0.40</td>
</tr>
<tr>
<td>12</td>
<td>1.18</td>
<td>1.5</td>
<td>5.9</td>
<td>174.2</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>13</td>
<td>1.5</td>
<td>2</td>
<td>12.1</td>
<td>364</td>
<td>20</td>
<td>0.13</td>
</tr>
<tr>
<td>14</td>
<td>3.6</td>
<td>2</td>
<td>1.9</td>
<td>59.8</td>
<td>~15</td>
<td>0.11</td>
</tr>
<tr>
<td>15</td>
<td>2.4</td>
<td>2</td>
<td>2.2</td>
<td>284</td>
<td>17</td>
<td>0.30</td>
</tr>
<tr>
<td>16</td>
<td>1.5</td>
<td>1.5</td>
<td>0.9</td>
<td>269</td>
<td>10</td>
<td>-</td>
</tr>
<tr>
<td>17</td>
<td>4.16</td>
<td>2</td>
<td>3</td>
<td>83.7</td>
<td>11</td>
<td>0.11</td>
</tr>
<tr>
<td>18</td>
<td>2.3</td>
<td>2</td>
<td>9.5</td>
<td>169.1</td>
<td>~10</td>
<td>1.80</td>
</tr>
</tbody>
</table>

**This work** | 4 | 1.6 | 4.6 | 60.5 | ~15 | 0.10 |
Fig. 11. The measured group delay in the passband for the proposed LPF.

Table 2: A comparison between the maximum variations of the measured group delay between some referred works with reported group delay.

<table>
<thead>
<tr>
<th>Ref</th>
<th>f0 (GHz)</th>
<th>Maximum Variation of the Group Delay in the Passband (ns)</th>
<th>Resonator Structure</th>
</tr>
</thead>
<tbody>
<tr>
<td>19</td>
<td>1.1</td>
<td>0.5</td>
<td>Tapered</td>
</tr>
<tr>
<td>20</td>
<td>1.74</td>
<td>0.5</td>
<td>Butterfly-shaped</td>
</tr>
<tr>
<td>21</td>
<td>1.55</td>
<td>0.5</td>
<td>Semi-circle stepped impedance</td>
</tr>
<tr>
<td>22</td>
<td>3.55</td>
<td>0.4</td>
<td>Spiral</td>
</tr>
<tr>
<td>23</td>
<td>2.37</td>
<td>0.44</td>
<td>Stub loaded semi-circle stepped impedance</td>
</tr>
<tr>
<td>24</td>
<td>4</td>
<td>0.27</td>
<td>T-Shaped, patch and stepped impedance</td>
</tr>
<tr>
<td>This work</td>
<td>4</td>
<td>0.23</td>
<td>Stepped impedance</td>
</tr>
</tbody>
</table>

**IV. CONCLUSION**

A miniaturized LPF has been proposed with wide rejection band as a small size and efficient suppressing cell. The proposed LPF rejects the spurious signals from 5 GHz to 23.3 GHz with attenuation level of 16.6 dB. The maximum variation of the measured group delay in the passband region is only 0.23 ns, which is the less value in comparison with some reported works. The measurement results clear the accuracy of the simulations.

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A Low-pass Filter with Sharp Transition and Wide Stop-band Designed based on New Metamaterial Transmission Line

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Abstract — A low-pass filter (LPF) with sharp and wide stop-band was designed based on a new metamaterial transmission line. Its characteristics were investigated, and found to have good low-pass performances with only one cell. The simulated results of the LPF are in good agreements with the measured results. Two measured transmission poles were observed at 1.35 GHz and 2.13 GHz. The measured 3 dB cut-off frequency is 2.38 GHz and the transmission attenuate to -20 dB at 2.62 GHz, thus, a 240 MHz width sharp transition was achieved. The measured attenuation of the stop-band is larger than 25 dB with two transmission zeros at 2.79 GHz and 5.19 GHz. A wide stop-band over 10 GHz was observed. Therefore, the proposed LPF is a good candidate for RF systems.

Index Terms — Low-pass filter, metamaterial transmission line, sharp transition, wide stop-band.

I. INTRODUCTION

Compact and high-performance low-pass filters (LPFs) are highly desired in many communication systems as they can be used to suppress undesired harmonics and spurious signals of the mixing products in RF front-ends [1-4]. These designs are based on several design units of defected ground structure (DGS) [1], stepped-impedance resonator (SIR) [2], lumped elements [3], and finite-ground microstrip line [4]. On the other hand, there has been a growing interest for the use of metamaterial transmission lines (TL) in the development of compact microwave components [5-6] and band-pass filters [7-8]. The concept of metamaterial has been successfully utilized to design complementary split ring resonator (CSRR) based band-pass filters (BPF) [9-11] and LPF [12]; however, these filters need multiple units to obtain sharp transition and wide stop-band, which is not suitable for compactness.

In our previous work [13], a novel metamaterial TL with right-hand (RH) property at low and high-frequency bands and left-hand (LH) property at middle-frequency band was analyzed. The TL exhibits low-pass function with/without notches. As discussed in [13], the high RH band can be designed as stop-band through using large grounded capacitance, and the middle LH band can be designed to be impedance mismatch to have a good low-pass response. It is seen that the LH band is the transition between the low-pass and high-stop bands. Thus, narrower LH band means sharper skirt. Besides LH cell is much smaller than the wavelength. Therefore, compact low-pass filter with sharp and wide stop-band can be designed based on such metamaterial TL.

In this work, the designing of LPF based on our previous studied metamaterial TL [13] is proposed. In [13], an implementation of the metamaterial TL was achieved by cross stubs-loaded square DGS and microstrip line-connected equilateral triangle stubs. It is found that the high RH band of the implementation is transmission forbidden due to the large grounded capacitance produced by the equilateral triangle stubs. Its attenuation is up to 30 dB even for a single cell. However, impedance mismatch happens at the upper part of the low RH band and the middle LH band matches well. Thus, a notch was observed for the low-pass response.

To solve this problem, characteristics of the upper part of the low RH band and the middle LH band must be disturbed. Therefore, dimension of the triangle stubs, which correspond to RH characteristics, were adjusted to impedance match the low RH band. Then, parallel
stubs connected to the cross stubs were utilized to enhanced LH property. The introduced parallel stubs increase the capacitance $C_{h1}$, the capacitance $C_{v2}$ and the inductance $L_{v2}$. Thus, LPF’s upper edge is greatly decreased, and the transition width between the pass-band and stop-band is improved. Therefore, the proposed design enables a steep attenuation transition from pass- to stop-band and wide stop-band with only one element, while a large number of components are needed in [1-4, 12]. Thereafter, a LPF was designed successfully. As the fabrication of the proposed LPF is based on printed circuit board (PCB) technique, its manufacture is very easy and extremely low cost, as well as facilitating for integrates into integrated circuit (IC).

II. METAMATERIAL LPF DESIGN AND ANALYSIS
A. Metamaterial LPF design

Configuration of the proposed LPF was shown in Fig. 1 (a). The LPF was fabricated on RT/Duroid 5880 substrate with relative permittivity $\varepsilon_r = 2.2$ and thickness 0.508 mm. Two triangle stubs with length of a side $L_1$ were connected to a 50 $\Omega$ microstrip line, and one of their peaks locate at the center of the microstrip line. On the other side of the substrate, a square defected pattern was etched in the ground plane with width $L$. Four cross stubs with width $W_1$ were placed in the square defected pattern. To enhance LH property, four parallel stubs were introduced by connecting to the four cross stubs, respectively. Then, the LH capacitance $C_{h1}$, and the capacitance $C_{v2}$ and inductance $L_{v2}$ are increased, which lead to the decreasing of LH frequency and the obtaining of a desired low-pass function. In simulations, the proposed LPF was excited by two wave ports at the ends of the microstrip line. The wave port is used to imitate infinite length of the microstrip line. Typical parameters of the LPF are $L = 16$ mm, $L_1 = 10$ mm, $W_0 = 1.52$ mm, $W_1 = 1.8$ mm, $W_2 = 2.15$ mm, $g = 0.2$ mm, and $d = 1.5$ mm. In many researches, LH properties were obtained by periodic structures, however, according CRLH theory, CRLH property can be achieved by single cell [5, 13]. Then, to obtained compact size, only one cell was used in this design.

The equivalent circuit model of the structure was derived and shown in Fig. 1 (b). The capacitances and inductances are related to certain parts of the LPF structure. The capacitance $C_{v2}$ is owing to the voltage gradients between the triangular stubs and the cross/parallel stubs. The inductance $L_{v2}$ is generated by the current flowing along the stubs, while the capacitance $C_{v3}$ is engendered by the voltage gradients between the triangular stubs and the ground plane. Then, the shunt resonant tank ($Y_2$) that forms by $C_{v2}$, $L_{v2}$ and $C_{v3}$ was obtained. The gap $g_0$ between the four parallel stubs are used to produce capacitance $C_{h1}$, while the etched pattern in the ground plane provides inductance $L_{h1}$. Then, series resonant tank ($Z_1$) was achieved. The reactance $Z_2$ ($L_{h3}$) is due to the current flowing along the microstrip line. Therefore, it is easily understand that the introduction of parallel stubs greatly increase the LH capacitance $C_{h1}$ and the capacitance $C_{v2}$. The equivalent circuit model is similar to the basic type 1 of the proposed metamaterial TL in [13]. We must point out that this circuit model is rough as some of the distributed effects may not be considered. However, it offers us a way to perceive insight into the structure.

As the S-parameters are utilized for $T$ type network parameters, effective permittivity and permeability extraction, then, simulated results of the proposed LPF were illustrated in Fig. 2 for further analysis. For comparison, results of a structure without parallel stubs (the inset of Fig. 2) were also demonstrated. It is found from the figure that the proposed LPF exhibits excellent low-pass function, wide stop-band and sharp transition. Its -3 dB upper edge is 2.39 GHz, and its transmission attenuation decrease to -20 dB at 2.65 GHz. Therefore, the transition width is 260 MHz. A wide stop-band with attenuation more than 20 dB was obtained over 10 GHz. The results of the structure without parallel stubs show a -3 dB upper edge at 3.15 GHz. Its transmission attenuation decreases to -20 dB at 3.70 GHz. The transition width is 550 MHz. Though the LPFs with and without parallel stubs have the same size, the one with parallel stubs is much superior in better low-pass impedance matching, lower cut-off frequency and sharper transition between pass-band and stop-band.
B. **T** type network parameters

The derived lumped-circuit model offers us a way to perceive insight into the structure. However, the calculating of the values of the capacitors and inductors is a burdensome and difficult task. Besides, the equivalent circuit model is a rough one with some distribute parameters not considered. Therefore, equivalent **T** type network would be a helpful substitution. Extract the parameters of the equivalent **T** type network is convenient thanks to the microwave network theory and fast developed computer technology. Importantly, **T** type network enable us conveniently and directly cognize the nature of the structure. As the structure is symmetrical, we assumed its **T** type network has horizontal branch \( Z_1 \) and vertical branch \( Z_2 \) as shown in Fig. 3. Typical **T** type network is derived by solving \( ABCD \) matrix according to \( S \) parameters of full wave simulation [14]. Then, we have:

\[
Z_1 = \frac{(A-1)/C}{Z_2} = 1/C. \tag{1}
\]

The reactance curves of \( Z_1 \) and \( Z_2 \) of the two LPFs with and without parallel stubs were calculated by formula (1) and illustrated in Fig. 4. Their corresponding \( S_{21} \) curves were also plotted in the figure for comparison. It is found that both the structures with/without parallel stubs exhibit LH property (\( \text{Imag}(Z_1) < 0 \) and \( \text{Imag}(Z_2) > 0 \)) at middle-frequency band and RH property (\( \text{Imag}(Z_1) > 0 \) and \( \text{Imag}(Z_2) < 0 \)) at high-frequency band. Imaginary \( Z_2 \) is close to zero for the low-frequency band. Balanced conditions are met at the transitions between the low-frequency RH band and middle-frequency LH band, while ENG bands [15] (epsilon-negative, \( \text{Imag}(Z_1) > 0 \) and \( \text{Imag}(Z_2) > 0 \)) were observed between the middle-frequency LH band and the high-frequency RH band. It is found that the introduction of parallel stubs moves the lower edge of LH band from 2.61 GHz to 2.07 GHz and the width of the LH band is reduced from 570 MHz to 380 MHz, consequently, LPF with more compact size and sharper transition is achieved.

**C. Effective permittivity and permeability**

As the dimension of the unit cell is much less than the operational wavelength, the structure can be characterized by quasi-TEM model and effective medium theory. Thereby, effective permittivity \( \varepsilon_{\text{eff}} \) and effective permeability \( \mu_{\text{eff}} \) will be extracted to further validate our conclusion. An improved Nicolson-Ross-Weir (NRW) approach was adopted for the effective constitutive parameters extraction [16]. First, we list the formula in the following:

\[
\mu_{\text{eff}} = \frac{2}{j k_0 d} \frac{1-(S_{21}-S_{11})}{1+(S_{21}-S_{11})}, \quad \varepsilon_{\text{eff}} = \mu_{\text{eff}} + \frac{2S_{11}}{k_0 d}, \tag{2}
\]

where \( k_0 \) is the wave number of free space, \( d \) is the length of the unit-cell, \( S_{11} \) and \( S_{21} \) are the scattering parameters. The transmission factor can be described as \( \tau = \exp(-jkd) \).

The extracted effective permittivity and permeability...
for the structures with/without parallel stubs were illustrated in Fig. 5, and their $S_{21}$ parameters were also presented for comparison. Their effective permittivity and permeability are positive for the low- and high-frequency RH bands, and negative for the middle-frequency LH band. The ENG regions have negative permittivity and positive permeability. The constitutive parameters have good agreement with the $T$ type network parameters, then, validity of these extract approaches on our structures were verified.

III. RESULTS AND DISCUSSION

We have investigated the LPF by means of equivalent circuit model, $T$ type network, and constitutive parameters in Section II. Then, the operational mechanism of the LPF was revealed. To better understand the LPF’s characteristics, and utilize it for further design in practical applications, more discussion and results were illustrated in this Section.

A. Parametric study

In Section II A, we found the derived equivalent circuit model elements have certain connection with certain parts of the LPF, therefore, the adjusting of certain parameters of the structure will tune their corresponding capacitances or inductances, consequently LPF performances. Therefore, the proposed LPF with parallel stubs were studied by sweeping its three key parameters ($g$, $d$, and $L_t$) as shown in Fig. 6. The simulated $S$-parameters for the three parameters were demonstrated in Figs. 6 (a), (b) and (c), respectively, while their extracted constitutive parameters were illustrated in Figs. 6 (d), (e) and (f) correspondingly. Note that, when one parameter is changed, others are fixed.

According to the capacitance calculation formulation $C = \frac{\varepsilon S}{d_0}$ of parallel-plate conductors with area $S$ and distance $d_0$, if the parameter $g$ is increased from 0.2 mm to 0.6 mm, it is equivalent to enlarge the distance $d_0$. Therefore, the coupling between the parallel stubs is minimized and LH capacitance $C_{h1}$ is decreased. Subsequently, the LH frequency increase as demonstrated in Fig. 6 (d). The ascending of the LH band results in increasing of the cut-off frequency as shown in Fig. 6 (a). While the parameter $d$ increased from 1.5 mm to 5.5 mm, the lengths of the parallel stubs are shortened. Thus, it is equivalent to reduce area $S$ of a parallel-plate, consequently, both the LH capacitance $C_{h1}$ and inductance $L_{v2}$ are decreased. Therefore, both the cut-off frequency and LH band is ascended as exhibited in Figs. 6 (b) and (e). It is found from Figs. 6 (a) and (b) that the decreasing of $C_{h1}$ (LH feature weakened) also leads to a more broadened transition. For example, when $g$ increased from 0.2 mm to 0.6 mm, the width of the transition for -3 dB to -20 dB increased from 260 MHz to 390 MHz, while $d$ increased from 1.5 mm to 5.5 mm, the width transition for -3 dB to -20 dB increased from 260 MHz to 480 MHz. While parameter $L_t$ is increased, the couplings between the triangular stubs and cross/parallel stubs/ground plane are increased. Therefore, both the capacitances $C_{v2}$ and $C_{v3}$ are enhanced. Then, the parameter $L_t$ mainly affects RH characteristics of the proposed LPF. Therefore, the high RH stop-band presents better attenuation and the middle LH band is almost unmoved with $L_t$ increasing as shown in Figs. 6 (c) and (f). However, enhanced RH feature results in difficulty for impedance matching of upper part of low RH band. At last, we can conclude from Fig. 6 that, LH enhancement lead to sharper transition and lower cut-off frequency, while RH enhancement means better high RH stop-band attenuation. However, by considering impedance matching of the pass-band (especially upper part of low RH band), trade-off among impedance matching, transition width and high stop-band must be considered by its LH and RH properties.

Fig. 5. Constitutive parameters and $S_{21}$ parameter: (a) the proposed LPF with parallel stubs, and (b) the LPF without parallel stubs.
B. Fabrication and measurement

To further validate the properties of the proposed LPF, it was fabricated as illustrated in Fig. 7. Based on printed circuit board (PCB) technique, the manufacture of the proposed LPF is very easy and extremely low cost, as well as facilitating for integrates into integrated circuit (IC). Two SMA connectors were soldered at the ends of the microstrip fed line for measurement as presented in the figure. Note that, the measurements were performed by an Agilent E5071C ENA series network analyzer with the highest measurable frequency at 8.5 GHz. The simulated and measured S-parameters of the proposed LPF are demonstrated in Fig. 8 with good agreements between them observed. The measured 3 dB cut-off frequency is 2.38 GHz with two transmission poles at 1.35 GHz and 2.13 GHz, and two transmission zeros at 2.79 GHz and 5.19 GHz. The transmission attenuation decreased to -20 dB at 2.62 GHz. Therefore, the measured transition width is 240 MHz. Besides, the measured attenuation of the stop-band is larger than 25 dB.

Fig. 7. Photograph of the proposed LPF.

Fig. 8. Simulated and measured S-parameters.

IV. CONCLUSION

This paper provides a new approach for LPF designing by applying metamaterial TL technology. Equivalent circuit model of the proposed LPF was derived, and comparison of LPFs with/without parallel stubs was conducted by S-parameters, T type network parameters and constitutive parameters. Parametric study was performed to reveal the relations between LH/RH characteristics and LPF performances (S parameters). Research found the effects of LH and RH features on the cut-off frequency, transition width and high-stop band, which in turn could be used to guide the designing of LPF. The proposed LPF has sharp transition and wide stop-band that turn out to be good candidate for RF front-end circuits.

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High Selectivity Balanced Filters Based on Transversal Signal-interaction Concepts

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Abstract — Two novel high selectivity balanced filters based on transversal signal-interaction concepts with wideband common mode suppression are proposed in this paper. Four and six transmission zeros near each passband are realized to improve the selectivity for the differential mode. In addition, the common mode can be suppressed with insertion loss greater than 15 dB over a wide frequency band. Two prototypes (εr = 2.65, h = 0.5 mm, tan δ = 0.003) with 3-dB fractional bandwidths of 31.3% and 32% for the differential mode with upper stopband greater than 18 dB are designed and fabricated. Good agreements can be observed between measured results and theoretical expectations.

Index Terms — Balanced filter, differential/common mode, open/shorted coupled lines, transmission zeros, transversal signal-interaction concepts.

I. INTRODUCTION
Balanced circuits have recently attracted special attention in communication systems for their higher immunity to the environmental noises, better dynamic range, and lower electromagnetic interference [1]. Many different balanced filters with selective filtering of differential mode (DM) signal and suppression for common-mode (CM) response are illustrated in [2]-[12]. In [2], the multi-stage branch-line topologies on a single-layer microstrip line were utilized to design a class of wideband balanced filters, but these filters have disadvantages of large overall circuit area. Some differential ultra-wideband (UWB) balanced bandpass filters based on double-sided parallel-strip line (DSPSL) are illustrated in [3], however, the upper stopbands for the differential mode are a little narrow. In [4], wideband differential filters employing the transversal signal-interaction concept are used to improve the common mode suppression, as well as the simple design theory with large insertion loss. In [5], the low-loss balanced filter with wideband common mode suppression using microstrip-slotline coupling are realized, but the numbers of transmission zeros near the differential mode passband are difficult to increase. In addition, the T-shaped resonator in [6] and ring resonator in [7]-[8] were applied to design balanced filters. In [9], the common-mode suppression of the balanced bandpass filter can be kept at a high level by adding a varactor to the center of the resonator. In [10], a balanced SIW filter using source-load coupling is proposed. To further improve the selectivity of the balanced filters, coupled lines and quarter/half-wavelength open/shorted stubs have been widely used [11-12].

In this paper, two novel balanced filter circuits based on transversal signal-interaction concepts with multiple transmission zeros for the differential/common mode are proposed. Four and six transmission zeros near the differential mode passband can be easily realized using the transversal signal interference concept, and five and seven transmission zeros can be also used to realize wideband common mode suppression. Two prototypes of the balanced filters operating at 3.0 GHz are constructed on the dielectric substrate with εr = 2.65, h = 0.5 mm, and tan δ = 0.003.

II. ANALYSIS OF PROPOSED BALANCED FILTERS
In this section, two balanced filters based on transversal signal-interaction concepts are analyzed in detail. The differential mode and common mode circuit are used to analyze the transmission characteristics of the two balanced filters in Part A and Part B, the simulated results of the two balanced filters are given in Part C.

A. Balanced filter analysis (Structure I)
Figure 1 (a) shows the ideal circuit of the proposed balanced filter structure with four transmission zeros. Four open/shorted coupled lines (even/odd-mode characteristic impedance Zeo2 and Zeo2, electrical length θ) with two quarter-wavelength transmission lines (characteristic impedance Z1, electrical length θ) on each side are shunted connected in the input/output ports 1, 1', 2, 2'. Four open coupled lines (even/odd-mode characteristic impedance Zel and Zoel, electrical length θ) are located in the center of the equivalent circuit with two shorted stubs (characteristic impedance Zi, electrical length θ)
length $\theta$ and an open stub (characteristic impedance $Z_0$, electrical length $2\theta$). The characteristic impedances of the microstrip lines at the input/output ports are $Z_0 = 50 \, \Omega$.

When the differential mode signals are excited from ports 1 and 1' in Fig. 1 (a), a virtual short appears along

$$Y_{o1-DM} = -j \frac{2\cos \theta \left[ 2Z_{oe1}Z_{oe2} \cos^2 \theta - Z_e (Z_{oe1} + Z_{oe2}) \sin \theta \right]}{2Z_{oe1}Z_{oe2}(Z_{oe1} + Z_{oe2}) \sin \theta \cos^2 \theta + Z_e \sin \theta \left[ (Z_{oe1} + Z_{oe2})^2 \cos^2 \theta - (Z_{oe1} - Z_{oe2})^2 \right]}$$

$$+ j \frac{-(Y_{oe1} + Y_{oe2})Z_e \cos^2 \theta + (Y_{oe1} - Y_{oe2})Z_e + 2(Y_{oe1} + Y_{oe2}) \sin^2 \theta}{(Y_{oe1} + Y_{oe2})Z_e \sin \theta \cos \theta - (Y_{oe1} - Y_{oe2})^2 Z_e \tan \theta + (Y_{oe1} + Y_{oe2}) Z_e \sin 2\theta}$$

$$Y_{o1-DM} = \frac{j[4Z_e \tan^2 \theta - 2(Z_{oe1} + Z_{oe2})]}{(Z_{oe1} + Z_{oe2}) (2Z_e + Z_{oe1} + Z_{oe2}) \tan \theta}$$

$$+ \frac{j[Z_e (2Z_{oe2} + Z_{oe1}) + 2Z_{oe2}Z_{oe1}]}{2Z_e + Z_{oe1} + Z_{oe2} \cot \theta - Z_e^2 (Z_{oe1} + Z_{oe2}) \tan \theta}$$

$$Y_{o2-DM} = -j \frac{2Z_{oe1}Z_{oe2}(Z_{oe1} + Z_{oe2}) \sin \theta \cos^2 \theta + Z_e \sin \theta \left[ (Z_{oe1} + Z_{oe2})^2 \cos^2 \theta - (Z_{oe1} - Z_{oe2})^2 \right]}{2Z_{oe1}Z_{oe2}(Z_{oe1} + Z_{oe2}) \sin \theta \cos^2 \theta + Z_e \sin \theta \left[ (Z_{oe1} + Z_{oe2})^2 \cos^2 \theta - (Z_{oe1} - Z_{oe2})^2 \right]}$$

$$+ j \frac{2Z_{oe2}Z_{oe1} \cot \theta - Z_e^2 (Z_{oe1} + Z_{oe2}) \tan \theta'}{2Z_{oe1}Z_{oe2} \cot \theta - Z_e^2 (Z_{oe1} + Z_{oe2}) \tan \theta}$$

$$Y_{o2-DM} = Y_{o1-DM}.$$  \hfill (4)

When $Y_{o1-DM} = 0$, the resonator frequencies in the passband for the odd-mode for the differential mode can be calculated as:

$$\theta_{11} = \arccos \left( \frac{-B \pm \sqrt{B^2 - 4AC}}{2A} \right), \quad \theta_{12} = \pi - \arccos \left( \frac{-B \pm \sqrt{B^2 - 4AC}}{2A} \right),$$

$$A = a_1^2 a_2 Z_e (a_1 Z_2 + 2) + 2aa_1^2 Z_e (Z_2 Z_e + Z_{eo2} Z_{eo1}) + 4aa_1 Z_e (Z_2 Z_e + Z_{eo2} Z_{eo1}) + 4aa_1 Z_e (a_1 Z_2 + 2),$$

$$B = -\{a_1^2 Z_e (b_1^2 Z_2 + 2a_1 Z_2) + 2aa_2 a_1 Z_2 Z_e Z_2^2 + 2a_2 Z_2 Z_e Z_{eo1} Z_{eo2} + 2a_1 b_2 Z_2 Z_2 Z_e^2 + 2a_1 b_2 Z_2 Z_e Z_{eo1} Z_{eo2} + a_1^2 b_2 Z_2 Z_2 Z_e^2 Z_{eo1} Z_{eo2}\},$$

$$C = 2a_1 b_2 Z_2 Z_e Z_2^2 + b_1^2 b_2^2 Z_e Z_2 + 2a_1 b_2 Z_2 Z_e,$$

$$a_1 = Z_{eo1} + Z_{eo2}, \quad b_1 = Z_{eo1} Z_{eo2},$$

$$a_2 = Y_{oe2} + Y_{oe1}, \quad b_2 = Y_{oe2} - Y_{oe1}. \hfill (5)$$

Figures 3 (a)-(d) plot the odd-mode resonant frequencies versus $\theta$ and the simulated results of the circuit in Figs. 1 (b)-(c). Due to the superposition of signals for Paths 1 and 2, four transmission zeros ($f_{z1}$,
$f_{o1}, f_{o2}, f_{o3}$ can be easily achieved for the proposed balanced filter [13]. The 3-dB bandwidth of the differential mode is mainly determined by the odd-modes $f_{o1}$ and $f_{o2}$. In addition, the odd-modes $f_{o1}$ and $f_{o2}$ move towards $f_0$ as $Z_1$ increases, and the 3-dB bandwidth of the differential mode increases as the coupling coefficient $k_1 = (Z_{oo}+Z_{oo1})/ (Z_{oo}+Z_{oo1})$ increases. The unwanted even/odd modes for the common mode can be suppressed less than -30 dB by the five transmission zeros. Four transmission zeros $(f_{o1}, f_{o2}, f_{o3}, f_{o4})$ move away from $f_0$ as the characteristic impedance $Z_1$ increases. In this way, the out-of-band harmonic suppression of the differential mode can be adjusted by the characteristic impedance $Z_2$ without changing the bandwidth of the differential mode. Next, to further improve the selectivity and the common mode suppression of the balanced filter with four transmission zeros, another high selectivity balanced filter structure with six transmission zeros close to the differential mode passband, and wideband common mode suppression will be presented.

![Diagram](image1)

**B. Balanced filter analysis (Structure II)**

Figure 4 (a) shows the ideal circuit of the balanced filter structure with two half-wavelength open stubs ($Z_1$, $2\theta$), instead of two quarter-wavelength short stubs ($Z_1$, $\theta$), and the other part is the same as the balanced filter of Structure I.

As discussed in Part A, when the differential/common mode are excited from ports 1 and 1' in Fig. 4 (a), a virtual short/open appears along the symmetric lines. The even/odd-mode equivalent circuits for the differential/common mode are shown Figs. 5 (a)-(b), and the input admittance for the differential mode of Fig. 5 (a) can be illustrated as:

$$Y_{o1-DM} = j \frac{Z_0(Z_{oo1} + Z_{oo2})\sin 2\theta \cos 2\theta + 4Z_{oo1}Z_{oo2}\sin 2\theta \cos^2 \theta}{Z_0 \cos 2\theta \left( (Z_{oo1} - Z_{oo2})^2 - (Z_{oo1} + Z_{oo2})^2 \cos^2 \theta \right) + Z_0 Z_{oo1}(Z_{oo1} + Z_{oo2})\sin^2 2\theta}$$

$$+ j \frac{- (Z_{oo1} + Z_{oo2}) Z_0 \cos \theta + (Z_{oo1} - Z_{oo2}) Z_0 \sin \theta - (Z_{oo2} - Z_{oo1}) Z_0 \tan \theta + (Z_{oo1} + Z_{oo2}) Z_0 \sin 2\theta}{(Z_{oo1} + Z_{oo2})^2 \tan \theta \tan 2\theta - 2Z_{oo1}(Z_{oo1} + Z_{oo2}) \tan \theta}$$

$$Y_{o2-DM} = -j \frac{Z_0 Z_{oo1} \tan 2\theta + 4Z_0 \tan \theta}{(Z_{oo1} + Z_{oo2})^2 \tan \theta \tan 2\theta - 2Z_0 (Z_{oo1} + Z_{oo2}) \tan \theta}$$

$$+ j \frac{Z_0 (Z_{oo1} + Z_{oo2}) Z_0 \tan \theta}{2Z_0 Z_{oo1} Z_{oo2} \cot \theta - Z_0 (Z_{oo1} + Z_{oo2}) \tan \theta}$$

$$[\text{Fig. 3. (a) Analysis of resonator frequencies versus } \theta, \text{ (b) phase of } |S_{dd21}|/(Z_0=50 \Omega, Z_1=50 \Omega, Z_2=100 \Omega, Z_{oo1}=160 \Omega, Z_{oo2}=90 \Omega, f_{o1}=3.0 \text{ GHz}, f_{o2}=3.0 \text{ GHz}, f_{o3} \theta = 90^\circ).]$$

**Fig. 4 (a) shows the ideal circuit of the balanced filter structure.**

**Fig. 5 (a) and (b) show the even/odd-mode equivalent circuits for the differential and common mode, respectively.**

**Fig. 6 (a) and (b) show the input admittance for the differential and common mode, respectively.**

**Fig. 7 (a) and (b) show the 3-dB bandwidth of the differential and common mode, respectively.**
When \( Y_{\text{eo,DM}} = 0 \), the resonator frequencies in the passband for the even-modes for the differential mode can be calculated as:

\[
\begin{align*}
\theta_{\text{o1}} &= \arctan \frac{-B + \sqrt{B^2 - 4AC}}{2A}, \\
\theta_{\text{o2}} &= \pi - \arctan \frac{-B + \sqrt{B^2 - 4AC}}{2A},
\end{align*}
\]

\[A = -4Z_1Z_3^2(Z_{\text{oe1}} + Z_{\text{oo1}}),\]

\[B = (Z_{\text{oe1}} + Z_{\text{oo1}})^2[2Z_{\text{oe1}}Z_{\text{oo1}} + Z_2(Z_{\text{oe1}} + Z_{\text{oo1}})] + 2(Z_{\text{oe1}} + Z_{\text{oo1}})(Z_{\text{oe2}} + Z_{\text{oo2}})(Z_1Z_3^2 + Z_2^2)\]

\[+ 4Z_1Z_3Z_2(Z_{\text{oe1}} + Z_{\text{oo1}}) + 8Z_1Z_3Z_2Z_{\text{oe2}}Z_{\text{oo2}},\]

\[C = -4Z_1Z_3Z_2Z_{\text{oe2}}(Z_{\text{oe1}} + Z_{\text{oo1}}).\]

Figures 6 (a)-(c) show the even/odd-mode resonant frequencies for the differential mode circuit versus \( \theta \) and the simulated results of the circuit in Figs. 4 (b)-(c). Due to the superposition of signals for Paths 1 and 2, six transmission zeros \( (f_{\text{e1}}, f_{\text{e2}}, f_{\text{e3}}, f_{\text{e4}}, f_{\text{e5}}, f_{\text{e6}}) \) can be easily achieved for the proposed balanced filter [13]. Compared with the balanced filter of Structure I, two additional transmission zeros \( (f_{\text{e5}}, f_{\text{e6}}) \) are located at 0.5\( f_0 \) and 1.5\( f_0 \), which can be used to further improve the differential mode passband selectivity and the common mode suppression level from 0 to 2\( f_0 \). In addition, the locations of two transmission zeros \( (f_{\text{e5}}, f_{\text{e6}}) \) do not change with all parameters of the circuit, and the locations of another four transmission zeros \( (f_{\text{e1}}, f_{\text{e2}}, f_{\text{e3}}, f_{\text{e4}}) \) do not change with \( Z_{\text{oe1}}, Z_{\text{oo1}}, Z_1 \) and \( Z_2 \). The 3-dB bandwidth of the differential mode is mainly determined by the even-modes \( f_{\text{e1}} \) and \( f_{\text{e2}} \), and the 3-dB bandwidth of the differential mode decreases and differential/common mode suppression becomes better with the decrease of \( k_1 \) \( (k_1 = (Z_{\text{oe1}} - Z_{\text{oo1}})/(Z_{\text{oe1}} + Z_{\text{oo1}})) \). The common mode can be suppressed less than -30 dB by the seven transmission zeros. Compared with the balanced filter of Structure I, the selectivity and common mode suppression of the balanced filter of Structure II has been further improved.

Fig. 4 (a) The ideal circuit of the balanced filter, (b) equivalent circuit for the differential mode, and (c) equivalent circuit for the common mode. (Structure II.)

Fig. 5 (a) Even/odd-mode equivalent circuit of differential mode circuit, and (b) even/odd-mode equivalent circuit of common mode circuit. (Structure II.)
C. Proposed two balanced bandpass filters

Referring to the discussions and the simulated results in Part A and B, the final parameters for the filters of Figs. 1, 4 are listed as below: $Z_0 = 50 \, \Omega$, $Z_i = 50 \, \Omega$, $Z_2 = 100 \, \Omega$, $Z_{oe1} = 160 \, \Omega$, $Z_{oe2} = 150 \, \Omega$, $Z_{oo1} = 110 \, \Omega$, $Z_{oo2} = 150 \, \Omega$, $Z_{oe1} = 90 \, \Omega$, $Z_{oe2} = 90 \, \Omega$, $Z_{oo1} = 90 \, \Omega$, $Z_{oo2} = 110 \, \Omega$. The structure parameters for two balanced filters (52.28 mm × 35.85 mm, 55.48 mm × 36.26 mm) shown in Figs. 7 (a)-(b) are: $l_1 = 15.31 \, mm$, $l_2 = 17.5 \, mm$, $l_3 = 6.66 \, mm$, $l_4 = 10.42 \, mm$, $l_5 = 15.46 \, mm$, $l_6 = 16.06 \, mm$, $l_7 = 10.6 \, mm$, $l_8 = 12.55 \, mm$, $l_9 = 11.65 \, mm$, $l_{10} = 4.2 \, mm$, $l_{11} = 9.42 \, mm$, $l_{12} = 4 \, mm$, $w_0 = w_2 = w_5 = 1.34 \, mm$, $w_1 = 0.2 \, mm$, $w_2 = 0.35 \, mm$, $w_3 = 0.3 \, mm$, $s_1 = s_2 = 0.2 \, mm$, $t_1 = t_2 = 1.89 \, mm$, $d = 0.7 \, mm$, $l_1 = 15.79 \, mm$, $l_2 = 17.5 \, mm$, $l_3 = 4.55 \, mm$, $l_4 = 10.42 \, mm$, $l_5 = 4 \, mm$, $l_6 = 18 \, mm$, $l_7 = 16.96 \, mm$, $l_8 = 17.56 \, mm$, $l_9 = 12.3 \, mm$, $l_{10} = 8.75 \, mm$, $l_{11} = 13.76 \, mm$, $l_{12} = 4.3 \, mm$, $l_{13} = 9.32 \, mm$, $l_{14} = 4 \, mm$, $w_0 = w_1 = 1.34 \, mm$, $w_1 = 0.2 \, mm$, $w_5 = 0.76 \, mm$, $w_3 = 0.45 \, mm$, $w_4 = 0.3 \, mm$, $s_1 = s_2 = 0.2 \, mm$, $t_1 = t_2 = 1.89 \, mm$, $d = 0.7 \, mm$.

Figures 8-9 illustrate the simulated results of the two balanced filters with four/six transmission zeros (Simulated with ANSYS HFSS v.13.0). For the differential mode of balanced filter of Structure I, four simulated transmission zeros are located at 1.52, 2.23, 3.88 and 4.64 GHz, the in-band insertion loss is less than 0.5 dB with 3-dB bandwidth approximately 30.7% (2.54-3.46 GHz); for the common mode, the insertion loss is greater than 17 dB from 0 GHz to 8.5 GHz, indicating good wideband rejection. Moreover, for the differential mode of the balanced filter of Structure II, five simulated transmission zeros are located at 1.44, 2.29, 3.63, 4.1 and 4.87 GHz, the 3-dB bandwidth is 29% (2.44-3.31 GHz) with return loss greater than 13 dB (2.56-3.34 GHz); for the common mode, over 15-dB common mode suppression is achieved from 0 GHz to 8.92 GHz.

Fig. 6. (a) Analysis of resonator frequencies versus $\theta$ for the differential mode. (b) $|S_{dd21}|$ & $|S_{cc21}|$ versus $Z_l$ ($Z_0 = 50 \, \Omega$, $Z_i = 50 \, \Omega$, $Z_2 = 70 \, \Omega$, $Z_{oe1} = 160 \, \Omega$, $Z_{oe2} = 90 \, \Omega$, $Z_{oo1} = 110 \, \Omega$, $Z_{oo2} = 150 \, \Omega$, $Z_{oe1} = 90 \, \Omega$, $Z_{oe2} = 90 \, \Omega$, $Z_{oo1} = 90 \, \Omega$, $Z_{oo2} = 110 \, \Omega$). (c) $\Delta f_{3\text{dB}}$, $|S_{dd21}|$, $|S_{cc21}|$ versus $Z_{oe1}$, $Z_{oo1}$. (Structure II).

III. MEASURED RESULTS AND DISCUSSIONS

For comparisons, the measured $S$-parameters of the two balanced filters are also illustrated in Figs. 8-9. For the differential mode of the balanced filter of Structure I, the 3-dB bandwidth is 31.3% (2.57-3.51 GHz) with return loss greater than 13 dB, four measured transmission zeros are located at 1.03, 2.32, 3.9 and 4.63 GHz, the insertion loss in the passband is less than 0.6 dB and greater than 20 dB from 3.75 to 8.9 GHz ($2.97f_0$); for the common mode, the insertion loss is greater than 15 dB from 0 GHz to 8.8 GHz ($2.93f_0$). For the differential mode of the balanced filter of Structure II, six measured transmission zeros are located at 1.21, 1.61, 2.26, 3.72, 4.14 and 4.96 GHz with 3-dB bandwidth of 32% (2.48-3.44 GHz), the insertion loss is less than 0.7 dB with return loss greater than 10 dB from 2.4 GHz and 3.53 GHz, an upper stopband with insertion loss greater than 20 dB is realized from 3.59 to 8.94 GHz ($2.98f_0$); in addition, the insertion loss for the common mode is greater than 15 dB from 0 to 9 GHz ($3f_0$). The slight frequency discrepancies of measured passbands for the differential mode are mainly caused by the imperfect soldering skill of the shorted stubs and folded transmission line of the two balanced filters.

To further illustrate the characteristics of the two balanced filters, Table 1 illustrates the comparisons of measured results for several balanced filter structures. Compared with other balanced filters [3]-[12], more transmission zeros near the passband are obtained for the two balanced filter structures, and the upper stopbands for the differential mode of the two balanced filters stretch up to 2.97$f_0$ ($|S_{dd21}|$<20 dB) and 2.98$f_0$ ($|S_{cc21}|$<20 dB). The insertion losses are lower with 0.6 dB and 0.7 dB for the proposed balanced filters.
Table 1: Comparisons of measured results for some balanced filters

| Filter Structures | | FBW | Stopband | $|S_{dd11}|$, dB |
|-------------------|----------------|------|----------|----------------|
| Ref. [3]          | 0 (3.0 GHz)   | 110% | $<-20, 2.2 f_0$ | $<-15, (0-8)$ |
| Ref. [6] - II     | 2 (6.85 GHz)  | 70.7%| $<-20, 2.77 f_0$ | $<-13.5, (0-19.5)$ |
| Ref. [8]          | 4 (2.4 GHz)   | 17%  | $<-15, 2.75 f_0$ | $<-18.8, (0-6.5)$ |
| Ref. [11] - I     | 3 (5.0 GHz)   | 67.6%| $<-15, 2.7 f_0$  | $<-15, (1.9-8.3)$ |
| Ref. [11] - II    | 5 (5.0 GHz)   | 37.8%| $<-15, 2.8 f_0$  | $<-20, (1.2-9.3)$ |
| Ref. [12]         | 3 (3.0 GHz)   | 82%  | $<-15, 2.1 f_0$  | $<-10, (1.65-3.95)$ |
| Structure I       | 4 (3.0 GHz)   | 31.3%| $<-20, 2.97 f_0$ | $<-15, (0-8.8)$  |
| Structure II      | 6 (3.0 GHz)   | 32.0%| $<-20, 2.98 f_0$ | $<-15, (0-9)$    |

IV. CONCLUSION

In this paper, two novel balanced filters based on transversal signal-interaction concepts with multiple transmission zeros are proposed. Four and six transmission zeros close to the differential mode passband can be easily achieved using the transversal signal-interaction concepts. In addition, wideband common mode suppression can easily realized by the multiple transmission zeros. High selectivity and wideband common mode suppression are obtained for the two balanced filters, indicating good candidates for microwave wireless applications.

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