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Convergence Analysis in Deterministic 3D Ray Launching Radio Channel Estimation in Complex Environments

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Abstract — In this paper, a convergence analysis to obtain the optimal calculation parameters in an in-house 3D ray launching algorithm to model the radio wave propagation channel in complex indoor environments is presented. Results show that these parameters lead to accurate estimations with reduced computational time. In addition, simulation results of an indoor complex scenario in terms of received power and power delay profile are presented, showing significant influence of multipath propagation in an indoor radio channel. The adequate election of simulation parameters given by convergence conditions, can aid in optimizing required computational time.

Index Terms—Convergence analysis, 3D ray launching, radio channel simulation, radio wave propagation.

I. INTRODUCTION

With the growing demand for wireless communication services in the last two decades, a radioplanning tasks for these systems is compulsory. Thus, a thorough analysis of indoor propagation channel characteristics represents a fundamental step toward the design and the implementation for an efficient setup of an indoor wireless network.

Traditionally, empirical methods were used (such as COST-231, Walfish-Bertoni, Okumura-Hata, etc.) for initial coverage estimation [1-3]. These empirical models can give rapid results but cannot take into account site-specific features, exhibiting limited accuracy. As an alternative, deterministic methods have been proposed [4-10] based on numerical approaches involving either solution of Maxwell’s equations using full-wave simulation techniques, such as method of moment (MoM) and finite difference time domain (FDTD) [11]; or, using geometrical approximations such as Ray Launching (RL) [12] and Ray Tracing (RT) [13]. During the 1990s, RL and RT were both classified as ray tracing methods; although, more recently the differences between both methods have been distinguished. RL is based on the fact that a transmitter launches thousands of test rays in a solid angle and the true path is determined by searching for the rays arriving at the receiver; whereas, RT estimates the principal radio wave propagation regions and rays are traced in such regions. These methods are precise but are time-consuming due to inherent computational complexity. Their combination with Uniform Theory of Diffraction (UTD) [14] is frequently applied to radio coverage prediction [15-18]. RT and RL models potentially represent the most accurate and versatile methods for urban and indoor multipath propagation characterization. Nevertheless, the computational time in conventional ray launching defined by the Shooting-and-Bouncing-Ray (SBR) method [19], can be very large depending on the required accuracy of the results.

A prerequisite to the modeling of complex environments with standard computer equipment,
is an outstanding reduction of memory efforts with an accurate approach. In [20], Weinmann presents a new efficient approach in order to assess the simulation of scattered fields from arbitrary metallic objects with a ray tracing algorithm, which combines the principles of Physical Optics (PO) and the Physical Theory of Diffraction (PTD). This paper demonstrates that both CPU time and accurate results depend strongly on the number of rays N and the accuracy of the geometric model. The treatment of multiple interactions in a GO-PO approach has been studied in [21], showing that on one hand the multiple PO approach indeed is more accurate. On the other hand, the simulation efforts increase exponentially with the order of the reflection, which makes multiple PO not applicable to more than double reflections. In [22], a description of a novel implementation of diffracted rays according to the Uniform Theory of Diffraction (UTD) concept into a SBR code is presented, describing a convergence analysis with respect to the number of diffracted rays and in terms of multiple diffractions. In [23], a Geometrical Ray Implementation for Mobile Propagation Modeling (GRIMM) is presented, which splits the 3-D ray construction problem into two successive two-dimensional (2-D) stages, without loss of generality and with a great gain in time and simplicity. This permits to take into account reflections and diffractions in any order to meet convergence. In [24], a convergence analysis of a refined ray-tracing algorithm is carried out with respect to the propagation time and the number of bounces.

It is proved that an increase of the number of rays and angular resolution in a SBR approach achieves satisfying results but leads to the significant drawback of higher computational time [20]. Therefore, it is relevant to study the convergence of the results when using ray tracing methods.

In this work, an analysis of the convergence of an in-house developed 3D ray launching algorithm has been presented with respect to the computational time, the angular resolution of the shooting rays and the number of reflections considered. This code has been employed as an effective tool to analyze the effect of radio wave propagation in different complex environments [25-28]. Results show the optimal parameters to achieve the commitment between accuracy and computational time in radio wave simulations of complex indoor environments.

This paper is organized as follows: the in-house developed 3D ray launching code is analytically described in sections II and III. Section IV presents the considered scenario for the analysis and the convergence analysis of the algorithm, whereas simulation results of the algorithm are discussed in section V. Finally, conclusions are given in section VI.

II. THE RAY-LAUNCHING TECHNIQUE

As it is illustrated in Fig. 1, ray launching techniques are based on identifying a single point on the wave front of the radiated wave, with a ray that propagates along the space following a combination of optic and electromagnetic theories.

![Wave front propagation with rays associated with single wave front points.](image)

Each ray propagates in the space as a single optic ray. The electric field $E$ created by an antenna with a radiated power $P_{rad}$, with a directivity $D_d(\theta, \Phi)$ and polarization ratio $(X_\perp, X_\parallel)$ at distance $d$ in the free space is calculated by [29]:

$$E_i^\perp = \frac{P_{rad}D_d(\theta, \Phi)\eta_0}{2\pi d} e^{-j\beta_\perp r} X_\perp L_\perp,$$  \hspace{1cm} (1)

$$E_i^\parallel = \frac{P_{rad}D_d(\theta, \Phi)\eta_0}{2\pi d} e^{-j\beta_\parallel r} X_\parallel L_\parallel,$$  \hspace{1cm} (2)
where \( \beta_0 = 2\pi f_0 \sqrt{\epsilon_0 \mu_0} \), \( \epsilon_0 = 8.854 \times 10^{-12} \), \( \mu_0 = 4\pi \times 10^{-7} \) and \( \eta_0 = 120\pi \). \( L^\perp \) are the path loss coefficients for each polarization.

When this ray finds an object in its path, two new rays are created: a reflected ray and a transmitted ray. These rays have new angles provided by Snell's law \([30]\). Once the parameters of transmission \( T \) and reflection \( R \) are calculated and the angle of incidence \( \Psi_i \) and \( \Psi_t \), the new angles \((\theta_r, \Phi_r)\) of the reflected wave and \((\theta_t, \Phi_t)\) of the transmitted wave can be calculated. A general case where a ray impinges with an obstacle with \((\theta_i, \Phi_i)\) angles, is represented in Fig. 2. Taking into account all the possible angles of incidence, the new angles for the reflected and transmitted wave are calculated, as it is shown for a general case in Fig. 2.

The diffracted field is calculated by \([31]\):
\[
E_{UTD} = e_0 \frac{e^{-jk s_1}}{s_1} D^\perp \frac{s_1}{s_2(s_1 + s_2)} e^{-jk s_2},
\]
where \( s_1, s_2 \) are the distances represented in Fig. 3, from the source to the edge and from the edge to the receiver point. \( D^\perp \) are the diffraction coefficients in (4) given by \([31-33]\) as:
\[
D^\perp = e^{-i\left(\frac{\pi}{\eta_0} n_2 \frac{1}{s_2} - \frac{\pi}{\eta_0} n_1 \frac{1}{s_1}\right)} \left( \cot \left( \frac{\pi}{\eta_0} n_1 \frac{1}{s_1} \right) F \left( k \frac{a^*}{2} (\Phi_2 - \Phi_1) \right) \right) + \cot \left( \frac{\pi}{\eta_0} n_2 \frac{1}{s_2} \right) F \left( k \frac{a}{2} (\Phi_2 - \Phi_1) \right) + R_0^\perp \cot \left( \frac{\pi}{\eta_0} n_1 \frac{1}{s_1} \right) F \left( k \frac{a^*}{2} (\Phi_2 + \Phi_1) \right) + R_0^\perp \cot \left( \frac{\pi}{\eta_0} n_2 \frac{1}{s_2} \right) F \left( k \frac{a}{2} (\Phi_2 + \Phi_1) \right),
\]
where \( n\pi \) is the wedge angle, \( F, L \) and \( a^\pm \) are defined in \([31]\), and \( R_{0,n} \) are the reflection coefficients for the appropriate polarization for the 0 face or n face, respectively. The \( \Phi_2 \) and \( \Phi_1 \) angles in (4) would refer to the angles in Fig. 3.

The received power is calculated at each point taking into account the losses of propagation through a medium \((\epsilon, \mu, \sigma)\) at a distance \( d \), with the attenuation constant \( \alpha \) (Np/m) and the phase constant \( \beta \) (rad/m). The received power is calculated with the sum of incident electric vector fields in an interval of time \( \Delta t \) inside each cuboid of the defined mesh. Based on the above theory, the main characteristic of the ray-launching technique is that it provides the impulse response of the channel \( h(t, f_c, \Delta f, d) \) for each transmitter, at a given carrier frequency, \( f_c \), at a given bandwidth \((f_c \pm \Delta f)\), where the materials have a similar response and at a given position. With this information, a stationary channel can be wholly characterized.

**III. THE 3D RAY-LAUNCHING ALGORITHM**

The developed in-house 3D ray-launching algorithm has three phases:
- Phase I: Scenario creation
- Phase II: 3D ray-launching simulation
- Phase III: Analysis results
In phase I, the 3D scenario is created by considering several objects, walls, transmitters, receivers and the whole components of the environment. It is important to emphasize that a grid is defined in the space to save the parameters of each ray. Accordingly, the environment is divided into a number of cuboids of a fixed size. When a ray enters a specific hexahedron, its parameters are saved in a matrix. Parameters such as frequency of operation, radiation patterns of the antennas, number of multipath reflections, separation angle between rays and cuboid dimension can be modified in the algorithm. In order to achieve adequate values, it is highly important to take into account, not only the parameters which result in more accurate results, but also the computational time, which can be substantial for a complex environment.

Fig. 4. Principle of operation of the in-house developed 3D ray launching code.

In phase II, rays are launched and they propagate along the space interacting with the obstacles, causing physical phenomena such as reflection, refraction and diffraction, as shown in Fig. 4. The parameters of these rays are stored as they enter to each hexahedral until the ray has a certain number of reflections or it has exceeded the pre-propagation time set. The algorithm operates in an iterative manner, considering a ray and its reflections and storing the created ray for processing later the diffraction contribution. Figure 5 shows the different steps of this phase of the algorithm.

Fig. 5. Functional diagram of the 3D ray launching algorithm.
In phase III, by using the parameters stored in phase II and considering the predefined transmitter’s characteristics, such as radiated power, the emitting antenna’s directivity $(\theta, \Phi)$, wave polarization and the carrier frequency, it is possible to easily derive the relevant parameters for channel modeling.

**IV. CONVERGENCE ANALYSIS**

In principle, it can be stated that simulation results tend to be more accurate whether more rays are launched and more reflections are considered. However, computational time has to be taken into account in order to obtain accurate results with an acceptable time span.

In this work, the influence of considering a different number of reflections and launching rays has been done. For that purpose, two different angular resolutions of launching rays have been considered and analyzed, versus the number of reflections. Afterwards, once the optimal number of reflections has been obtained, the assessment of the most accurate angular resolution of launching rays has been done taking into account the computational time.

The scenario under consideration is a complex environment, which corresponds to several rooms of the iRadio Laboratory of the University of Calgary. A schematic view of the scenario is represented in Fig. 6.

**A. Convergence versus the number of reflections**

First, a convergence analysis of the algorithm versus the number of reflections that the ray could impact with the obstacles and walls has been...
obtained. Different cases have been considered varying the number of reflections. These cases correspond to different angular resolutions when the rays are launched, as shown in Table 2.

Table 2: Different cases considered for simulation versus the number of reflections

<table>
<thead>
<tr>
<th>Case</th>
<th>N_{launching rays}</th>
<th>\Delta \Phi=\Delta \theta</th>
</tr>
</thead>
<tbody>
<tr>
<td>Case 0</td>
<td>16200</td>
<td>2°</td>
</tr>
<tr>
<td>Case 1</td>
<td>64800</td>
<td>1°</td>
</tr>
<tr>
<td>Case 2</td>
<td>259200</td>
<td>0.5°</td>
</tr>
</tbody>
</table>

Table 3: Simulation parameters

<table>
<thead>
<tr>
<th>Ray Launching Parameters</th>
<th></th>
</tr>
</thead>
<tbody>
<tr>
<td>Frequency</td>
<td>2.4 GHz</td>
</tr>
<tr>
<td>Transmitter power</td>
<td>20 dBm</td>
</tr>
<tr>
<td>Cuboids resolution</td>
<td>12x12x12 cm</td>
</tr>
<tr>
<td>Transmitter point</td>
<td>(X=6.24 m, Y=16.94 m and Z=0.96 m)</td>
</tr>
</tbody>
</table>

Case 1 and case 2 have been analyzed versus the number of reflections and simulation, parameters for both cases are shown in Table 3.

Figures 9, 10, 11 and 12 depict simulation results for case 1. The mean value and the standard deviation of power (dBm) for different X and Z and along the Y-axis are presented. These linear radials correspond to the red lines in Figs. 8 (a) and (b). The mean power values exhibits large variability, due to the influence of the morphology and topology of the scenario. Therefore, the way in which the elements within the indoor scenario are configured and the different material properties play a fundamental role in the overall performance of the network.

The standard deviation of power decreases as the number of reflections increases and converges approximately for six reflections in all cases, which implies that the maximum efficiency of the algorithm will be reached when six reflections are considered.

Figure 10 shows the linear regression lines of the received power for different locations in the X axis, along the Y-axis and for different heights. It is demonstrated that when the number of reflections considered in the algorithm has been increased, the algorithm tends to converge and it exhibits the same behavior from a threshold value for the number of reflections, corresponding to six reflections. In all cases, the trend is to decrease over distance and it is observed that with more reflections the slope of the lines is smaller.

Figure 11 shows the mean and the standard deviation of power (dBm) for different Y and Z locations and along the X-axis (green lines in Fig. 8 (a)). It is observed that mean power converges as the number of reflections increases. Besides, the standard deviation of the received power decreases and also converges at approximately six reflections.

Figure 12 shows the linear regression lines of the received power for different values in Y axis and heights along the X-axis. It is observed that for smaller values in the Y axis, the slope of the lines is smaller as the number of reflections increases. From Y=6 m, figures of smaller heights show more variability due to the obstacles within the ray path. Lines converge as the number of reflections is increased. From Y=12 m, power increases along the X-axis due to the proximity of the transmitter antenna.
Fig. 9. Mean and standard deviation of power for different X and Z versus the number of reflections with $N_{\text{launching rays}}=64800$: (a) X=1 m, (b) X=3 m, (c) X=5 m and (d) X=7 m.

Fig. 10. Linear regression lines of the received power for different X, along the Y-axis and for different heights with $N_{\text{launching rays}}=64800$: (a) X=3 m and (b) X=7 m.
Fig. 11. Mean and standard deviation of power for different Y and Z versus the number of reflections with $N_{\text{launching\ rays}} = 64800$: (a) $Y=4$ m, (b) $Y=8$ m (c) $Y=12$ m and (d) $Y=16$ m.

Fig. 12. Linear regression lines of the received power for different Y, along the X-axis and for different heights with $N_{\text{launching\ rays}} = 64800$: (a) $Y=6$ m and (b) $Y=12$ m.
In order to validate previous results, which conclude that six reflections must be considered in complex indoor scenarios, the same analysis has been done with a lower angular resolution (case 2 in Table 2). This implies that the number of shooting rays is increased to test if the number of rays has an influence in the algorithm convergence. Simulation parameters are shown in Table 3.

Figure 13 represents the mean and the standard deviation of power (dBm) for different X and Z, along the Y-axis, with $\Delta \Phi = \pi/360$ and $\Delta \theta = \pi/360$. As in the previous case, large variability in the mean of the received power is observed, due to the influence of the morphology and topology of the scenario and multipath fading, which is a relevant phenomenon in this type of indoor scenarios. The standard deviation of the power does not converge as clearly as in the previous case. However, Figs. 13 (b) and (d) depict convergence of the algorithm.

The linear regression lines of the received power for different X, along the Y-axis and for different heights with angular resolution of launching rays of 0.5º, are presented in Fig. 14. It is observed that in all cases the received power decreases with the distance and the slope of the lines are smaller as the number of reflections increases. In case 2, the algorithm converges with one reflection less considered because of the rise of launching rays. The slope of the regression lines in Figs. 14 and 16 are for every height, smaller than case 1; therefore, the number of considered reflections to achieve the convergence is smaller.

However, the computational time is also deeply important. Simulations have been performed in an Intel Xeon CPU X5650 @ 2.67 GHz and 2.66 GHz. In case 2, the computational complexity is increased overall. Figure 17 shows the simulation time of the considered scenario depending on the number of reflections and the number of launching rays. Computational time with angular resolution of 0.5º (case 2) is hugely greater than case 1, with angular resolution of 1º for each number of reflections considered. It can be seen that the optimal parameters to achieve the most accurate results with an acceptable computational time, is to consider six reflections and angular resolution of launching rays of 1º.

Fig. 13. Mean and standard deviation of power for different X and Z versus the number of reflections with $N_{\text{launching rays}}=259200$: (a) X=1 m, (b) X=3 m, (c) X=5 m and (d) X=7 m.
Fig. 14. Linear regression lines of the received power for different X, along the Y-axis and for different heights with $N_{\text{launching rays}}=259200$: (a) $X=3$ m and (b) $X=7$ m.

Fig. 15. Mean and standard deviation of power for different Y and Z versus the number of reflections with $N_{\text{launching rays}}=259200$: (a) Y=4 m, (b) Y=8 m, (c) Y=12 m and (d) Y=16 m.
Fig. 16. Linear regression lines of the received power for different $Y$, along the X-axis and for different heights with $N_{\text{launching rays}} = 259200$: (a) $Y=6$ m and (b) $Y=12$ m.

Fig. 17. Comparison of computational time versus different number of launching rays and different number of reflections considered in the algorithm.

B. Convergence versus the number of launching rays

In order to validate the results obtained in the previous section, an analysis of the convergence of the algorithm versus the number of launching rays has been performed considering six reflections. Figures 18 and 19 show the mean and the standard deviation of power for different locations in the X, Y axes and heights, in comparison with the three cases shown in Table 2. It can be seen that mean value of received power increases as the number of shooting rays is increased, which is in agreement with previous results in the sense that the algorithm is more accurate with more shooting rays. Alternatively, the standard deviation of power decreases sharply for case 0 to case 1, and converges in all cases in case 1, which corresponds to angular resolutions of 1°. Accordingly, these results validate the previous statements taking into account the high amount of CPU-time required to analyze the large amount of rays of case 3, which is shown in Fig. 20.
Fig. 18. Mean and standard deviation of power in room 2 versus different cases of shooting rays for different X.

Fig. 19. Mean and standard deviation of power in room 2 versus different cases of shooting rays for different Y.
Fig. 20. Comparison of computational time versus different number of launching rays.

V. SIMULATIONS RESULTS

In order to gain insight in the effect of parameter variation in wireless channel estimation, several test cases have been simulated. Figure 21 shows the estimated received power for two different heights in the indoor scenario of the iRadio Laboratory (depicted in Fig. 7). Simulations have been performed with the parameters shown in Table 3, with an angular resolution of one degree and six reflections.

Fig. 21. Bi-dimensional planes of estimated received power: (a) height 0.95 m and (b) height 1.908 m.

To illustrate the relevance of the multipath effect, the power delay profile has been predicted along for \( x=3 \) m, \( y=12 \) m and \( z=1.908 \) m, as shown in Fig. 22. It can be seen that there are a large number of echoes in the scenario, due to multipath propagation, which is the most important phenomena in this type of indoor scenarios.

Fig. 22. Power-delay profile for: \( x=3 \) m, \( y=12 \) m and \( z=1.908 \) m, in the considered scenario.
VI. CONCLUSION

In this work, the convergence analysis to obtain the optimal parameters to introduce in a 3D in-house implemented ray launching code have been presented. Results show that adequate election of parameters such as ray angular resolution, number of reflections and cuboid resolution, lead to accurate results with adequate computational time as a consequence of algorithm convergence. In addition, simulation results of an indoor complex scenario in terms of received power and power delay profile are presented, showing the significant influence of multipath propagation in an indoor radio channel. These results can aid in the correct development of radioplanning tasks with optimal computational time.

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REFERENCES


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The Error Cross-Section Method for Quantifying the Error in Electromagnetic Scattering Problems

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Abstract — The Error Cross-Section (ECS) is introduced to quantify the error associated with the numerical solution of electromagnetic scattering problems. The ECS accounts for different approximations and inaccuracies in the object discretization and numerical computations. The ECS definition is based on the power conservation principle and is visualized by comparing it to the radar cross-section of a thin wire for two-dimensional (2-D) problems or a small sphere for three-dimensional (3-D) problems. The proposed ECS method is independent of the adopted numerical technique and therefore can be used to give confidence in the obtained solution using several methods, such as the Method of Moments (MoM) and the Finite-Difference Frequency-Domain (FDFD) method. Application of the ECS to the optimization of certain parameters for some numerical formulations, such as the Combined-Field Integral Equation (CFIE) is also presented.

Index Terms — CFIE, numerical error, radar cross-section.

I. INTRODUCTION

Numerical treatment of Maxwell’s equations has steadily advanced for decades and a variety of computational methods have been devised to solve electromagnetic problems, especially problems involving scattering from arbitrarily shaped objects [1]-[4]. Such problems require using geometrical discretization methods to model the objects in a manner amenable to computers, followed by approximations of the equations associated with the used formulation and finally adopting a numerical routine to evaluate such approximate forms.

As yet, there have been quite a few works on the quantification of the error associated with the above procedure in computational electromagnetics [5]-[8]. Commercial software packages do not provide a confidence level to the user in the accuracy of the produced results. Verifying the boundary conditions (frequently used to guarantee that the solution satisfies them) typically uses the same operator equation, which has been approximated and thus suffers from complications related to singularities, mesh inaccuracy etc. and is specific to the adopted method.

This work proposes an error quantification approach, which is not only independent of the adopted numerical method but also visualizes the error by the so-called Error Cross-Section (ECS). In section II, the numerical error in solving a general electromagnetic scattering problem is discussed. In the same section, the definition of the ECS is presented and its relation to the Radar Cross-Section (RCS) of a thin conducting wire is investigated. Also, the correlation between the proposed error measure and the actual error is studied. In section III, the ECS is computed for various scattering problems using different numerical methods, such as the MoM and the FDFD, with an application to the optimization of the mixing factors in the MoM combined-field formulation. Conclusions and discussions are given in section IV.

II. PROBLEM FORMULATION

A. Residual error in scattering problems

Figure 1 shows an arbitrary object illuminated by a uniform plane wave. The total fields in the region enclosing the scatterer are given by:
\[ \mathbf{E}^t = \mathbf{E}^i + \mathbf{E}^s, \quad (1) \]
\[ \mathbf{H}^t = \mathbf{H}^i + \mathbf{H}^s. \quad (2) \]

Typically, the scattered fields are determined using a numerical technique for arbitrarily shaped objects. The inaccuracies associated with the adopted technique result in an error in the total fields, i.e.:
\[ \delta \mathbf{E}^t = \mathbf{E}^t_{\text{num}} - \mathbf{E}^t_{\text{exact}}, \quad (3) \]
\[ \delta \mathbf{H}^t = \mathbf{H}^t_{\text{num}} - \mathbf{H}^t_{\text{exact}}. \quad (4) \]

Estimation of the residual error \( \delta \mathbf{E}^t \) and \( \delta \mathbf{H}^t \) requires obtaining the exact solution \( \mathbf{E}^t_{\text{exact}} \) and \( \mathbf{H}^t_{\text{exact}} \), which is usually unknown for noncanonical problems. Conventionally, numerical methods adopt certain convergence criteria to the solution by increasing the number of unknowns till the difference between the current and the previous solutions becomes acceptable. This, however, neither gives an indication about the error in the current solution with respect to the exact one, nor provides a physical insight into the quantified error.

In this work, the goal is to define a new quantity, conveniently referred to as the Error Cross-Section (ECS), which is correlated with the actual residual error in a way that is independent of the adopted numerical technique. Unlike most error estimates [5]-[8], only few have a physical meaning like the Sobolev norm [9]. The ECS has this advantage and can be used to visualize the quantified error by comparing its definition to the RCS. Furthermore, the highest solution accuracy that can be achieved on a specific machine can be deduced by finding the lower limit of the ECS.

**Fig. 1. Geometry of a general scattering problem.**

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**B. Definition of the ECS**

In electromagnetic scattering problems, the power conservation principle [10] requires that the total exiting power \( W^t \) must vanish if the media bounded by \( S \) were lossless, i.e.:
\[ W^t = \frac{1}{2} \int_S \text{Re} \left\{ \mathbf{E}^t_{\text{exact}} \times \mathbf{H}^t_{\text{exact}} \right\} \cdot \hat{n} dS = 0, \quad (5) \]

where \( \hat{n} \) is the outward normal to the surface \( S \). Ideally, (5) should be satisfied; however, due to the errors in the scattered fields computation, the integral in (5) yields a residual value, viz.
\[ W_{\text{res}}^t = \frac{1}{2} \int_S \text{Re} \left\{ \mathbf{E}^t_{\text{num}} \times \mathbf{H}^t_{\text{num}} \right\} \cdot \hat{n} dS \neq 0. \quad (6) \]

Thus, the ECS is defined as follows:
\[ \text{ECS} = \frac{W_{\text{res}}^t}{P_{\text{in}}}, \quad (7) \]

where \( P_{\text{in}} = \frac{1}{2} \int_C \text{Re} \left\{ \mathbf{E}^t_{\text{num}} \times \mathbf{H}^t_{\text{num}} \right\} \) is the incident power density.

For 2-D problems, the Error Width (EW) is used instead of the more general term ECS and the integral in (6) is performed on a contour \( C \) rather than on a surface \( S \). This is similar to using the scattering width in 2-D problems, instead of the radar cross-section used in 3-D problems.

A fundamental lower limit to the EW is attributed to the error in the numerical evaluation of (6) in the absence of the scatterer, i.e.:
\[ \text{EW}_{\text{min}} = \frac{W_{\text{res}}^t}{P_{\text{in}}}, \quad (8) \]

where \( W_{\text{res}}^t = \frac{1}{2} \oint_C \text{Re} \left\{ \mathbf{E}^t_{\text{num}} \times \mathbf{H}^t_{\text{num}} \right\} \cdot dl \neq 0 \).

Although there could be different combinations of \( \mathbf{E}^t_{\text{num}} \) and \( \mathbf{H}^t_{\text{num}} \), which satisfy the power conservation principle, only one of them is correct in light of the uniqueness theorem [10]. Therefore, before finding the ECS, the boundary conditions should be verified to guarantee that the solution under consideration is actually the correct one. It is important to underline that the goal behind using any error estimate is not to decide whether the solution is correct or not, but to find out how accurate a correct solution is and to establish a confidence level in it. In light of this, error estimates can be employed for many purposes, i.e.: to minimize the computational effort using a specific numerical method by determining the optimum number of unknowns and to find the highest obtainable accuracy. Also, they can be used to compare the accuracy of different methods for a...
given problem and to provide a physical meaning for the error.

**C. Physical meaning of the ECS**

To visualize the quantified error and to have a feeling about how much this error is for a specific problem, the definition of the ECS is compared to the RCS or the scattering width (SW) in case of 2-D problems, which is defined as [10]:

$$
\sigma_{2D} = \lim_{\rho \to \infty} 2\pi \rho \frac{|E|^2}{|\delta E'|^2}.
$$

(9)

This can give an estimate of the error associated with the solution as a residual field $\delta E'$ and $\delta H'$, due to scattering from a fictitious thin wire, as compared to the original problem of scattering from the actual object. Figure 2 shows the SW of a thin conducting wire of radius $a_w$. The ordinate of Fig. 2 will be used to access the EW of the solution to determine the radius of the corresponding thin wire.

**Fig. 2.** Scattering width of a conducting wire with radius $a_w$ normalized to the wavelength $\lambda$. The arrow shows how the figure is used to visualize the error width.

**D. Correlation between the EW and the actual residual error**

The correlation between the EW and the actual residual error $|\delta E'|$ for the problem of TM$^\parallel$ plane wave scattering from a 2-D circular PEC cylinder having a radius $a = \lambda/2$, is studied (see Fig. 3). In this example, the EFIE formulation of the moment method is adopted and the effect of varying the number of basis functions per wavelength ($N_b$) is investigated. Invoking (6), the total exiting power is computed on a circular contour of radius $b = a + \lambda/4$. The exact solution $E'_{\text{exact}}$ for this problem can be found analytically using [10]:

$$
E'_{\text{exact}} = -E_0 \sum_{n=-\infty}^{\infty} \frac{j_n(k_0 a)}{H_n^{(2)}(k_0 a)} H_n^{(2)}(k_0 b) e^{j\phi},
$$

(10)

where $j_n$ and $H_n^{(2)}$ are the Bessel function of the first kind and the Hankel function of the second kind, respectively and $k_0$ is the free-space wavenumber. The residual error $|\delta E'|$ is computed by averaging its values on the circular contour C. Results as shown in Fig. 4, indicate that the EW and $|\delta E'|$ have the same asymptotic convergence rate.

**Fig. 3.** TM$^\parallel$ plane wave scattering from a 2-D circular (approximated as an octagon) PEC cylinder.

**Fig. 4.** Correlation between the EW and the actual residual error $|\delta E'|$ for the MoM solution of the problem shown in Fig. 3.
III. RESULTS

A. Method of moments

The proposed method is applied to selected numerical methods such as the MoM and the FDFD techniques. First, the example shown in Fig. 5 is considered to compare the EW for the problem of plane wave scattering from a 2-D PEC cylinder with equilateral triangular cross-section using different formulations. The scatterer is enclosed by a cylinder of radius $a = \lambda/2$. Figure 6 (a) shows the computed EW for the case of electric (EFIE) and magnetic field integral equation (MFIE) for TM$^z$ illumination. It can be inferred from Fig. 6, that for this problem the EFIE has superior performance compared to the MFIE and that triangular basis functions give lower EW as compared to the pulses. Figure 6 (b) also shows that the EW is almost independent of $b$, for $b < b_{\text{critical}}$. This critical value depends on the accuracy of the numerical routine used to evaluate the integrals. Increasing this, accuracy results in a higher $b_{\text{critical}}$ and vice versa. To explain that, the scattered field intensity is noticed to be proportional to $1/b$ in 2-D problems and therefore the error in calculating $E^s$ and $H^s$ is also proportional to $1/b$. At the same time, when $b$ increases, the integration contour $C$ is enlarged with $2\pi(\Delta b)$ and the associated error increases with the same proportionality, provided that the accuracy is high enough. Therefore, the reduction in the error when calculating the scattered fields compensates the increase in the error due to the numerical evaluation of the integrals. This is true up to a critical value, $b_{\text{critical}}$ after which the error in evaluating the integrals, i.e.: $\text{EW}^{\min}$ is not linear anymore with $b$, as shown in Fig. 6 (c).

Fig. 5. TM$^z$ plane wave scattering from a 2-D equilateral triangular PEC cylinder.

Fig. 6. (a) EW versus $N_{\lambda}$, (b) EW versus $b$ and (c) $\text{EW}^{\min}$ for the MoM solution of the problem shown in Fig. 5.
An interesting observation regarding the MoM solution is that when the electrical size of the problem greatly increases, the number of unknowns increases likewise. There is a critical value for the number of unknowns at which the MoM matrix becomes ill-conditioned and the error in finding its inverse affects the overall accuracy. This critical value can be determined using the EW. To manifest this phenomenon, the example of Fig. 3 is considered again but the radius of the PEC cylinder is now made electrically huge, i.e.: \( a = 5\lambda \gg \lambda \). Using the MoM-EFIE formulation, \( N_A \text{critical} \) and the corresponding \( EW_{\text{critical}} \) are shown in Fig. 7. Based on the EW/SW analogy and referring to Fig. 2, the thin wire radius corresponding to the error at \( N_A \text{critical} \) is 1000.

![Fig. 7. The error width of the MoM solution using EFIE formulation for the problem shown in Fig. 3.](image)

**B. Finite-difference frequency-domain**

Considering another numerical method, the error in the FDFD solution of plane wave scattering from a 2-D rectangular dielectric cylinder with a side length \( l \) and a dielectric constant \( \varepsilon \) is investigated. Due to the rectangular grid employed by the FDFD in defining the geometry and field points, it is more convenient for the integration contour to be rectangular with a side length \( L \), as shown in Fig. 8. For \( l = 2\lambda, L = 4\lambda \) and \( \varepsilon = 4 \), the EW is depicted in Fig. 9.

![Fig. 8. Plane wave scattering from a 2-D rectangular PEC cylinder.](image)

![Fig. 9. The error width of the FDFD solution for the problem shown in Fig. 8 under TM\(^z\) illumination.](image)

**C. Optimization of the mixing factors in the MoM-CFIE**

An interesting application to the proposed error estimate is to determine the best choice of the mixing factors used in the CFIE, commonly adopted in the MoM solution to remedy the internal resonance problem. The CFIE formulation is typically obtained as a weighted sum of the EFIE and MFIE with orthogonal weights [11], i.e.:

\[
\text{CFIE} = \frac{\beta}{\eta_0} \text{EFIE} + \alpha \text{MFIE},
\]

(11)

where CFIE, EFIE and MFIE are either the matrix of unknowns or the excitation vector and \( \eta_0 \) is the intrinsic impedance of free-space. In [12], the choice of the factors \( \alpha \) and \( \beta \) was random. The EW...
concept can be employed to study the effect of this choice on the solution accuracy.

This is done by determining the combination that results in a minimum EW at each number of unknowns. In light of this, it was found that having a 90° phase difference between the mixing factors and keeping \( \alpha \) constant for the given problem, the value of \( \beta \) is a very sensitive function of \( N_\lambda \), i.e.:

\[
\text{CFIE} = \frac{\beta(N_\lambda)}{\eta_0} \text{EFIE} + j\alpha \text{MFIE}. \quad (12)
\]

The example in Fig. 5 is investigated again using the proposed formula in (12) and pulse basis functions. The variation of the EW with \( \beta \) for different values of \( N_\lambda \), is shown in Fig. 10. Results manifest that for each \( N_\lambda \) there exists a certain value of \( \beta \), which results in a minimum EW; hence, leads to a significant improvement in the solution accuracy compared to the EFIE or MFIE formulations.

![Fig. 10. MoM-CFIE solution using the formula in (12) for the scatterer shown in Fig. 5. The error width is plotted versus \( \beta \) coefficient for different values of \( N_\lambda \).](image)

IV. CONCLUSION

A general method based on power conservation and independent of the adopted technique is proposed to quantify the error. The definition of the ECS is introduced to compare the solution accuracy for different numerical techniques. For 2-D problems, the EW is used instead of the more general term ECS. The proposed method is applied to the MoM and FDFD solutions of plane wave scattering from 2-D objects. This approach can also be applied to 3-D objects in a straightforward manner. A comparison between the ECS and the RCS of a thin wire for 2-D problems or small sphere for 3-D problems is introduced to visualize the amount of error. An interesting application for the proposed ECS method is finding the critical value for the matrix size in MoM solution after which the accuracy starts to degrade. Another application is to estimate the accuracy of different formulations, as illustrated with the CFIE and to optimize the choice of the mixing factors. Moreover, incorporating the proposed method with the results obtained using commercial software packages as a post-processing step, is on-going with the goal of providing a unified benchmark for the error of these packages.

REFERENCES


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FPO-Based Shooting and Bouncing Ray Method for Wide-Band RCS Prediction

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Abstract — The fast physical optics (FPO) method for calculating the monostatic radar cross section (RCS) of an object over a range of frequencies is extended to the shooting and bouncing rays (SBR) method where the multi-bounce phenomena of the launched rays is considered. The FPO method is an improved version of the physical optics (PO) method, which is efficient when calculating the monostatic RCS over a wide range of frequencies or/and angles. The SBR method, based on the combination of geometrical optics (GO) and PO methods, can reach a higher accuracy compared with the PO method. However, due to different length of ray tube paths, it is difficult to implement the phase compensation and phase retrieval in the SBR as that in the FPO method. In this paper, a coordinate transformation is introduced in the integral equation, which transforms the original ray tubes model into a new one. The FPO technique can then be taken on the revised model without difficulty. The validity and efficiency of the proposed method are validated though a couple of numerical experiments.

Index Terms - Coordinate transformation, electromagnetic scattering, fast physical optics (FPO), shooting and bouncing rays (SBR), and wide-band RCS.

I. INTRODUCTION

The radar cross section (RCS) is an important characterization of the electromagnetic scattering properties of large objects and it is common practice to use this data in the fields of war monitoring, radar imaging simulation, and target identification. Techniques for computing RCS can be categorized into high-frequency techniques and low-frequency techniques [1]. The low-frequency techniques include the method of moments (MoM) [2], the finite element method (FEM) [3] and other MoM-type methods [4-10]. The high-frequency techniques include the geometrical optics (GO) and physical optics (PO) [11] methods, and more complicated methods, such as ILDCs [12], GO/UTD-PO/PTD [13], and SBR [14]. The high-frequency techniques, usually based on the ray-tracing and edge diffraction, are fast and accuracy acceptable when used to electrically large objects [15]. For example, an aircraft at L wave band will up to 130 wavelengths, which will result a large number of unknowns if 0.1 wavelength mesh size is used in the MoM method. This large number of unknowns puts forward high demanding in computer performance. However, the high-frequency techniques express the scattering field in an analytical approach and avoid solving the matrix equations as that in the low-frequency techniques, which reduces the memory and CPU time cost significantly.

A lot of researchers have paid attention to the calculation of wide-band RCS of an object. The asymptotic waveform evaluation (AWE) [16-18] technique and model-based parameter estimation (MBPE) [19] have been developed to decrease the computation burden associated with repeated point-by-point calculations [18]. However, the techniques mentioned above are based on low-frequency techniques, and the objects analyzed are not electrically large. Recently, a fast physical optics (FPO) [20-25] has been developed to fast evaluate wide-band or/and wide-angle monostatic RCS. The algorithm is based on the observation that the scattering pattern of a finite scatterer is an...
essentially band limited function of the aspect angles and frequencies [20, 26]. In the FPO method, the rapid oscillation of the integrand is cancelled by a phase compensation process, the RCS response over a range of frequencies or angles can then be obtained by interpolating the phase-compensated field at much sparser sampling grids. The backscattering echo of a 2-dimensional rectangular cylinder [20] and double-bounce scattering phenomena involving two surfaces [21] have been analyzed by the FPO. However, the method is rather complicated and limited to double-bounce scattering phenomena.

The shooting and bouncing rays (SBR) [14], proposed by H. Ling etc., is a robust and an accurate method in analyzing electrically large objects. In the SBR approach rays representing the incident fields in a GO manner are used to determine the resulting equivalent surface currents and finally the resulting field contributions at the given observation points are derived by PO integration [27]. Besides the first-order scattered fields, the SBR provides more accurate results by including the scattered fields arising from multiple bounces [29]. In [29], a GPU-based SBR that implemented on the graphics processing unit (GPU) is proposed to reduce the computation time. The ray tracing is modified to evaluate the exit point and field quickly. The electromagnetic computing is integrated into the process of central ray tracing, including the evaluation of the reflected and scattered field. The main contribution of this paper is utilizing the GPU to accelerate both ray tracing and electromagnetic computing of the SBR, which is a parallelization of the SBR method under a certain frequency and incident angle. The wide-band and wide-angle monostatic RCS can then be obtained by repeating this process at different frequencies and angles. Whereas the method proposed in [20] is focused on getting wide-band or/and wide-angle monostatic RCS with interpolation technique. According to this method, only a few monostatic RCS need to be calculated directly, the RCS at required frequencies and angles can then be obtained by interpolating these calculated ones.

In this paper, the FPO method for calculating the monostatic RCS of an object over a range of frequencies is extended to the SBR method. Unlike the method in [29], the method proposed in this paper tries to get wide-band RCS by interpolating values calculated at sampling frequencies. The RCS at these sampling frequencies can be obtained by either a CPU-based or a GPU-based SBR method as that in [29]. However, due to different length of ray tube paths, it is difficult to implement the phase compensation and phase retrieval in the SBR as that in the FPO method. To conquer this problem, a coordinate transformation is introduced in the integral equation, which transforms the original ray tubes model into a new one. The FPO technique can then be taken on the revised model. The remainder of this paper is organized as follows. In section II, theory and formulations are discussed. Numerical results are presented and discussed in section III. Section IV concludes this paper. The time factor $e^{j\omega t}$ is assumed and suppressed throughout this paper.

II. THEORY AND FORMULATION

Consider a perfectly electrically conductor (PEC) shown in Fig. 1, a plane wave is incident on this object ($\overline{\mathbf{E}}_{in}$ and $\overline{\mathbf{H}}_{in}$ denote the electric and magnetic field of the incident wave, respectively). $R_0$ is half of the diagonal line of the cube that contains the object. To facilitate analysis, the object is meshed with small patches (for example, triangular patches), the number of meshed triangles of which is dependent on the frequency and the size of the object. In section A below, the theory of the FPO is briefly discussed. The SBR and its improvement are discussed in section B.

![Fig. 1. Scattering model of an object.](image-url)
where $\hat{n}$ is a unit outward normal vector at the point $r$ , $s$ is the unit vector in the direction of observation. $k$ is the wavenumber, $\eta$ is the intrinsic wave impedance, $N_t$ is the number of triangle patches in the illuminative area, $R$ is the distance between the observation point and the object, and $j = \sqrt{-1}$. In the high frequency region, $k$ is a large number, and function $p$ is a slowly varying function regarding frequencies. In this paper, the integral technique proposed by Gordon [30] is adopted to calculate the integral in equation (1).

When used to wide-band scattering problems (for example, the required frequency band is $[f_{\text{min}}, f_{\text{max}}]$ with frequency interval $\Delta f$), one has to repeat the calculation of equation (1) at a sequence of frequency samples, which is time-consuming when used to electrically large object. According to the Nyquist sampling theorem, the sampling interval of the frequency should satisfy $\Delta f < c / (4 R_o)$, where $c$ is the velocity of light. Consequently, the number of frequency samples is

$$N_f = \Omega_f \frac{4 R_o (f_{\text{max}} - f_{\text{min}})}{c},$$

where $\Omega_f$ is the oversampling ratio satisfying $\Omega_f > 1$. It is observed that the number of frequency samples is proportional to the size of the object $R_o$. One possible way to reduce the $N_f$ is by dividing the object into several non-overlapped groups as shown in Fig. 2. The $N_t^i$ of the $i$-th group is obtained by substituting $R_o$ with $R_o^i$ in equation (2). Since $R_o^i < R_o$, the sampling points $N_f^i < N_f$. The back-scattered field of each group at the required frequencies can then be interpolated by these sparser sampling grids. However, the scattered field obtained by direct interpolation is not accurate. To improve the interpolation accuracy, phase compensation is applied to the back-scattered field prior to interpolation, and a phase retrieval algorithm is used after the interpolation stage. The phase compensated field of the $n$-th group can be written as,

$$\overline{E}_s(f) = 2 j k \eta e^{j \mathbf{k} \mathbf{R}} \sum \frac{N_t}{4 \pi R} \int \hat{s} \times (\hat{n} \times \overline{H}_m) e^{j \mathbf{k} \hat{s} \cdot \hat{r}} d \hat{r},$$

$$\overline{E}_s(f) = 2 j k \eta e^{j \mathbf{k} \mathbf{R}} \sum \frac{N_t}{4 \pi R} \int \hat{s} \times (\hat{n} \times \overline{H}_m) e^{j \mathbf{k} \hat{s} \cdot \hat{r}} d \hat{r},$$

$$\overline{E}_s(f) = 2 j k \eta e^{j \mathbf{k} \mathbf{R}} \sum \frac{N_t}{4 \pi R} \int \hat{s} \times (\hat{n} \times \overline{H}_m) e^{j \mathbf{k} \hat{s} \cdot \hat{r}} d \hat{r},$$

where $N_t^i$ is the number of triangle patches in the illuminative area of the $n$-th group, $r_i$ is the center coordinate of the $n$-th group. The rapid oscillation of the integrand is cancelled by the phase compensation process, which makes the interpolation process more accurate. After interpolation, the back-scattered field of each group at the required frequency point can be retrieved by phase restoration, i.e.,

$$\overline{E}_s(f) = \overline{E}_s(f) e^{j \mathbf{k} \overline{r}_i \cdot \hat{r}_i}.$$  

The total back-scattered field $\overline{E}_s(f)$ can then be obtained by aggregation of the back-scattered field of all groups and the monostatic RCS can then be written as,

$$RCS(f) = \lim_{R \to \infty} \frac{4 \pi R^2}{|E_s(f)|^2}.$$  

Fig. 2. Schematic drawing of partitioning of the object.

**B. SBR and its improvement**

The SBR method involves two steps: ray tube tracing and electromagnetic computing at the exit point. First, the incident plane wave is modeled as a dense grid of ray tubes at a virtual aperture, which are shot toward the object. When used to wide-band RCS prediction, the number of ray tubes is set according to the highest frequency to be analyzed. Each ray tube is recursively traced to obtain the exit position and its scattered field. Figure 3 shows a possible path of a ray tracing. It
should be noted that the ray tracing process is time-consuming when the object is meshed with large number of triangles and when the number of ray tubes is large. In [28], the angular Z-buffer (AZB), the volumetric space partitioning (SVP) and the depth-limited search method are combined to accelerate the ray tracing process, whereas a stackless kd-tree traversal algorithm is adopted to evaluate the exit position and field quickly in [29].

In this paper, the octree [31-32] technique is used to recursively subdividing the box into eight children to decrease the number of intersection tests. Second, the PO integral is preformed to obtain the scattered field of this ray tube based on the pre-calculated exit positions and field [29]. A stackless kd-tree traversal algorithm is adopted to accelerate the ray tracing process, whereas a FPO technique can not be used into SBR directly. Comparison between equations (6) and (9) shows that the coordinate transformation extracts the exponential term in the square brackets and combines with the exponential term outside the square brackets. As shown in Fig. 3, the original integral of equation (6) at the exit point (Point C) can be replaced by the integral of equation (9) at point D. In other words, the new model (or integral domain) has taken into account the effect of path of each ray tube, whereas the original integral is taken at the exit point. However, the coordinate transformation does not affect the fields and the medium through the transformation because equation (9) is consistent with equation (6) in essence and there is only a change in variables. To make this clearer, we consider a dihedral corner reflector as shown in Fig. 4 (a), the incident and observation angles are \( \theta_i = 45^\circ \), \( \varphi_i = 0^\circ \), \( \theta_s = 45^\circ \), \( \varphi_s = 0^\circ \). After ray tracing and coordinate transformation, the integral of equation (9) is taking on a new model as shown in Fig. 4 (b). Then the new model can be regrouped without difficulty and take the FPO technique as in section II A to fast get the wide-band RCS prediction.

Let \( \vec{r} = \vec{r} - \vec{r}_i \) and substitute it into equation (8) we finally get,

\[
\vec{E}_i(f) = \frac{jk\eta e^{jk\vec{r}_i} e^{ik\hat{s}\cdot\hat{r}_i}}{4\pi R} \sum_{i=1}^{N_i} \hat{s} \times (\hat{s} \times (\hat{n} \times \nabla \vec{H}_i)) e^{jk\hat{s}\cdot\hat{r}_i} ds.
\]

(9)

![Fig. 3. Schematic drawing of the path of a ray tube.](image)
Points A, B, and C are intersection points during the ray tracing. Point D is obtained by moving point C in the reverse direction of \( \hat{s} \) such that the distance between C and D (remarked as \( r_s \)) satisfies
\[
r_s = r_{i1} + r_{i2} + r_{i3}.
\]

Fig. 4. Schematic drawing of scattering model of a dihedral corner reflector; (a) the dihedral corner reflector and (b) the new model after ray tracing and coordinate transformation.

**III. NUMERICAL RESULTS**

In this section, the approach and its efficiency are validated through a couple of numerical experiments. The program is implemented on a personal computer with Intel Dual-core CPU. The CPU and memory sizes are 2.99 GHz and 3.24 GB, respectively. In all examples below, the 4-point Lagrange interpolation technique is used in the FPO with evenly distributed sampling points. As mentioned above, the octree grouping technique is used to accelerate the ray tracing process in the SBR and regrouping is needed on the new model after the ray tracing process. Let \( l_1 \) denote the finest group size of the former while \( l_2 \) denote the group size of the latter step.

Firstly, we consider a trihedral corner reflector with a side length of 1 meter, the geometry of the trihedral corner reflector is shown in Fig. 5. The monostatic RCS on the \( \theta = 60^\circ \) plane is calculated. The virtual aperture is also meshed with 0.01\( \lambda \) as that in [29]. The results of the HH-polarization at 3 GHz are shown in Fig. 6, which show a good agreement between the method proposed in this paper and the method in [29]. The total computation time of the proposed method in this paper is about 281 seconds, whereas the GPU-based method in [29] only needs 8.73 seconds. The reason is that the method introduced in this paper mainly deals with wide-band RCS prediction. The result may become unacceptable when used to wide-angle problems because of the drastic change in the path of each ray tube at different angles. So the proposed method in Fig. 6 degenerates to the conventional SBR indeed. However, as mentioned in the introduction, the GPU-based SBR method in [29] can be combined together with the method proposed in this paper to accelerate the computation at sampling frequencies when dealing with wide-band RCS problems.

Fig. 5. The geometry of a trihedral corner reflector.

Secondly, we validate accuracy of the proposed method. As shown in Fig. 7, we consider a trihedral corner reflector with a side length of 10 meters. The number of ray tubes is set to be 68,730. The incident angles are \( \theta_i = 45^\circ \), \( \varphi_i = 45^\circ \), respectively. \( l_1 = l_2 = 0.4m \). The frequency range of interest is 0.5 GHz – 1.5 GHz, with frequency interval \( \Delta f = 20 \) MHz (resulting 51 frequencies points). By using equation (2) with \( \Omega_f = 2 \) and
\( R_o = \sqrt{3}l_2 / 2 \), only 10 frequencies points are needed to be calculated. Figure 8 is the monostatic RCS for VV polarization versus the frequency. The result simulated by the commercial software CST [33] is also shown in the figure, which shows a good agreement among these methods.

\[
\text{relative error} = \frac{\sum_{i=1}^{\text{frequency points}} |\text{RCS}_{\omega i} - \text{RCS}_{\omega i}^{\text{SBR}}|}{\sum_{i=1}^{\text{frequency points}} |\text{RCS}_{\omega i}^{\text{SBR}}|}, \tag{10}
\]

where the RCS calculated by the SBR method is used as the correct solution. It can be seen from the table that the proposed method can reduce the CPU time significantly with reasonable oversampling rates, and the relative error decreases with the oversampling rate increases.

<table>
<thead>
<tr>
<th>( \Omega_i )</th>
<th>No. of frequency points</th>
<th>CPU time (Sec.)</th>
<th>relative error</th>
</tr>
</thead>
<tbody>
<tr>
<td>SBR</td>
<td>701</td>
<td>433</td>
<td>5.6%</td>
</tr>
<tr>
<td>Proposed method</td>
<td>( \Omega_i = 2 )</td>
<td>34</td>
<td>18</td>
</tr>
<tr>
<td></td>
<td>( \Omega_i = 3 )</td>
<td>50</td>
<td>26</td>
</tr>
<tr>
<td></td>
<td>( \Omega_i = 4 )</td>
<td>68</td>
<td>35</td>
</tr>
</tbody>
</table>

Finally, we consider a tank model as shown in Fig. 10. The size of the tank is 8.11m x 2.59m x 1.82m. 690,454 ray tubes is used to get an accurate result over the C, X, and AN, FAN, DING, CHEN: FPO-BASED SHOOTING AND BOUNCING RAY METHOD FOR WIDE-BAND RCS.
Ku wave band (i.e., 4 GHz – 18 GHz). The frequency interval of interest is $\Delta f = 10$ MHz. The incident angles are $\theta_i = 60^0$, $\varphi_i = 0^0$. $l_1 = l_2 = 0.1m$ is adopted as that in the third example. Figure 11 shows the monostatic RCS versus the frequency with different oversampling rates of the proposed method, which shows a good agreement with the conventional SBR method. Table II lists the CPU time and error values of the proposed method, from which we can see that the proposed method can reduce the CPU time with a factor of 26 – 44 without losing accuracy compared with the conventional SBR. In terms of relative error, the error values in Table I are higher than those in Table II. The reason is that the monostatic RCS in Fig. 9 is much smaller than that in Fig. 11. The error values can be reduced by a double precision or higher order interpolation technique.

### Table II: CPU time cost and relative error for different oversampling rates.

<table>
<thead>
<tr>
<th>$\Omega_f$</th>
<th>No. of frequency points</th>
<th>CPU time (Sec.)</th>
<th>relative error</th>
</tr>
</thead>
<tbody>
<tr>
<td>SBR</td>
<td>---</td>
<td>1401</td>
<td>711</td>
</tr>
<tr>
<td>Proposed method</td>
<td>$\Omega_f = 2$</td>
<td>33</td>
<td>16</td>
</tr>
<tr>
<td></td>
<td>$\Omega_f = 3$</td>
<td>49</td>
<td>22</td>
</tr>
<tr>
<td></td>
<td>$\Omega_f = 4$</td>
<td>65</td>
<td>27</td>
</tr>
</tbody>
</table>

IV. CONCLUSION

In this paper, the FPO method for calculating monostatic RCS of an object over a range of frequencies is extended to the SBR method where the multi-bounce phenomena of the launched rays are considered. A coordinate transformation is used to cancel the influence of different length of ray tubes on phase compensation and phase retrieval, which transforms the original integral model into a new one. The FPO technique can then be taken on the revised model without difficulty. The proposed method and its efficiency are validated by numerical experiments, which show that the proposed method can reduce the CPU time significantly. The proposed method is especially suited for generation of synthetic data for radar imaging simulation.

ACKNOWLEDGMENT

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A Miniaturization Band-Pass Filter with Ultra-Narrow Multi-Notch-Band Characteristic for Ultra-Wideband Communication Applications

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Abstract — In this paper, we present a novel approach for designing compact Ultra-Wideband (UWB) band-pass filters with desired multi-notch-band characteristics. The multi-notch-band features are realized by using a ring-stub multi-mode resonator, while the equivalent model is also obtained by using the odd/even excitation resonance condition. The fabricated prototype of the band-pass multi-notch-band filter demonstrates a good behavior as expected. Simulated and experimental results show that the proposed filter with a compact size of 25×10 mm 2, has an impedance bandwidth between 3.0 GHz and 10.6 GHz; while the multiple notch bands are achieved and their center frequencies are located at 3.9 GHz, 5.25 GHz, 5.9 GHz, 6.8 GHz and 8.0 GHz. The proposed filter is suitable for being integrated in UWB radio systems to efficiently enhance the interference immunity from unexpected signals, such as Worldwide Interoperability for Microwave Access (WiMAX), Wireless Local-Area Network (WLAN), 6.8 GHz Radio Frequency Identification (RFID) communication and X-band satellite communications.

Index Terms - Band-pass filter, multi-notch band characteristics, notch band filter and UWB filter.

I. INTRODUCTION

Since the Federal Communications Commission (FCC) released the frequency band from 3.1 GHz to 10.6 GHz for commercial communication applications in February 2002, the Ultra-Wideband (UWB) radio system has been receiving more attention in both academic and industrial fields [1]. An UWB Band-Pass Filter (BPF) is one of the key passive components to realize an UWB radio system; which has attracted more attention recently. A great number of methods have been proposed in order to design BPFs with large Fractional Bandwidths (FBW) and good performance [2-6]. There have existed the typical structures, including low and high-pass filter configurations [2], Coplanar Waveguide (CPW) geometries [3], right/left-handed structures [4], microstrip cascaded fork-form resonator [5] and defected ground structure [6]. However, there are some drawbacks in these designs, such as out-of-band performance and complicated structure.

On the other hand, there are some existing narrow-band systems which have been used in the communications for a long time. Furthermore, the UWB frequency band overlaps with these existing narrow communication systems, such as WiMAX in 3.4 GHz to 3.69 GHz band, WLAN in 5.2 GHz and 5.8 GHz bands and 6.8 GHz RFID band. Those narrowband radio signals might interfere with UWB systems and vice versa. To mitigate the potential interference, design of compact UWB BPFs with notched band characteristics is one of the most challenging topics. Consequently, a number of methods have been proposed and investigated to design UWB BPFs with notched bands [7-22], such as embedded Complementary Split Ring Resonators (CSRR) [7], Defected
Ground Structures (DGS) [8,11], mismatch transmission lines [9], short-circuited stubs [10]; which can effectively reject unexpected radio signals. Nevertheless, some of these previously proposed band-notched UWB filters are still large in size [7], incompatible with Monolithic Microwave Integrated Circuits (MMIC) [8], complicated in structures [9] and sophisticated in band rejection characteristic design [7-11].

In this paper, we present a novel approach to designing compact Ultra-Wideband (UWB) band-pass filter with a good multi-notch-band characteristic. The proposed multi-notch-band characteristic is realized by using Ring-Stub Multi-Mode Resonator (RSMMR). The resonance condition of RSMMR is obtained by means of odd and even mode theorem [14]. In comparison with the existing UWB notch-band filter in [7-13], the proposed notched bands can be easily operated simultaneously. Performance of the proposed filter is validated by fabricating the proposed multi-notch-band pass UWB filter on a RT/Duroid 6006 substrate, that has a relative dielectric constant of 6.15 and a thickness of 0.635 mm.

II. FILTER DESIGN

In this section, the design procedure and synthesis method of the proposed compact UWB BPF with multi-notch bands are described on the basis of loading stub technique. Generally speaking, the proposed filter essentially exploits Multi-Mode Resonator (MMR) structures to realize the sharp and adjustable multi-notch-band.

The schematic diagram of the proposed UWB BPF with multi-notch-band characteristics is shown in Fig. 1. The proposed UWB BPF is a modification structure in [22], in which the multi-mode resonator structures are coupled to the middle ring resonator section, to achieve the required multi-notch bands. The equivalent transmission line network of the proposed filter is illustrated in Fig. 2. The interdigital coupled line is equivalent to a combination of the two signal transmission lines on two sides and a J-inverter susceptance. The multi-mode resonator coupled to the middle section of the ring resonator can be analyzed by a shunt series resonant branch, as shown in Fig. 2. Furthermore, in Fig. 1, the multi-mode resonator structure can be equivalent to an LC resonator network whose parameters are $L_i$ and $C_i$ with $1 \leq i \leq 5$; which are shown in Fig. 2. Therefore, each resonator can be regarded as an LC resonator, which is related to the corresponding notch band. Here, the $L$ and $C$ can be obtained from resonator theory.

Two capacitive-ended interdigital coupled into the I/O lines possess a wide-stop-band performance [22]. In this paper, we use the middle ring resonator to realize the multi-notch bands. The design idea and procedure of the multi-notch bands are described in Fig. 3. First, we improve the prototype filter proposed in [22] to cast a ring structure denoted as a simplified prototype filter. Then, a transition filter-1 is designed based on the simplified prototype filter by inserting a general stub into the ring resonator to produce a notch-band characteristic. As for filter-1, the inserted general stub is acted as a resonator, which is designed to provide a notch band. As a result, there is no coupling current at the output of filter-1. Therefore, the filter-1 can effectively reject the unwanted signal at corresponding frequency band. To make it suitable for dual-notch-band applications, another general stub is employed and inserted into the ring resonator based on the transition filter-1 to design a dual-notch-band filter, which is referred to as transition filter-2. The transition filter-2 can be treated as a ring resonator with two general stubs inserted into the simplified prototype filter. To render the proposed filter more useful in practice, a ring resonator with two stubs is employed to construct a tri-notch filter, referred to as transition filter-3. By considering the design procedure of the notch band filters aforementioned, an UWB band-pass filter with multi-notch bands is proposed by integrating three stubs, a ring resonator and a Stepped Impedance Resonator (SIR) into the simplified prototype filter; which is denoted as the final filter, as shown in Fig. 3. To give an insight into the performance of the proposed filter more useful in practice, a ring resonator with two stubs is employed to construct a tri-notch filter, referred to as transition filter-3. By considering the design procedure of the notch band filters aforementioned, an UWB band-pass filter with multi-notch bands is proposed by integrating three stubs, a ring resonator and a Stepped Impedance Resonator (SIR) into the simplified prototype filter; which is denoted as the final filter, as shown in Fig. 3. To give an insight into the performance of the proposed multi-notch-band filter, the proposed transition filters together with stubs and ring resonator are analyzed. Next, we investigate the characteristics of the single notch band to analyze the proposed transition filter-1. The geometry of the proposed transition filter-1 is illustrated in Fig. 4 and the equivalent circuits of the ring resonator are shown in Fig. 5.
Fig. 1. Configuration and dimensions of the proposed filter.

Fig. 2. Equivalent circuit network of the proposed filter.
methods. Even-mode and odd-mode equivalent circuits for the ring resonator fed by the interdigital coupled lines in Fig. 4, are shown in Figs. 5 (a) and (b). On the basis of the odd-mode equivalent circuit, the odd-mode forms a pass band only, while the even-mode is designed for not only the pass-band but also the notch-band. In this paper, it is desirable to design an UWB BPF with a sharp and tunable notch-band. Thus, we will describe how to realize a notch band using the stub technique mentioned above. As the even-mode can be designed for both the pass-band and notch-band applications, we will only analyze the resonance condition about the even-mode. By considering the analysis procedure in [14, 23], the even-mode resonance condition proposed herein can be achieved and expressed by the equation below for \( Y_{in} = 0 \). For the even-modes:

\[
Y_{in} = \tan \theta_1 + \frac{(K)^{-1} \tan \theta_4 + \frac{\tan \theta_1 + R}{1 - \tan \theta_1^R}}{1 - \tan \theta_1^R} = 0, \quad (1)
\]

where,

\[
R = (K)^{-1} \tan \theta_1 + \frac{\left[ \frac{1}{2} \left( K' \right)^{-1} \tan \theta_4 + K'^{-1} \right]}{1 - \frac{1}{2} \left( K' \right)^{-1} \tan \theta_4 + \frac{1}{2} \left( K' \right)^{-1} \tan \theta_4 \tan \theta_1}. \quad (2)
\]

In the equations above, the \( K' \), \( K'' \), \( K''' \) and \( K'''' \) can be expressed as:

\[
K = \frac{Z_2}{Z_1}, \quad K' = \frac{Z_2}{Z_1}, \quad K'' = \frac{Z_2}{Z_1}, \quad K''' = \frac{Z_2}{Z_1}. \quad (3)
\]

The simulation parameters are listed as follows (unit: mm): \( L1v=5.12, L1h=5.92, L2=4.25, L3=0.25, L4=5.39, L5=2.0, W1=0.46, W2=5.2, W3=1.3, W4=0.11 \) and \( W5=1.15 \). In the next step, we use these parameters to simplify the even-mode resonance condition. Based on the parameters above, we can get \( \theta_6 \approx 70^\circ \) and then the length of the stub is close to \( 7 \lambda_{notch}/36 \), where \( \lambda_{notch} \) corresponding to the center frequency of the notch can be expressed as:

\[
\lambda_{notch} = \frac{C}{f_{notch} \varepsilon_{eff}}, \quad (4)
\]

where, \( f_{notch} \) is the center frequency of the notch band, \( \varepsilon_{eff} \) is the effective dielectric constant and
C is the speed of light in free space. The effective dielectric constant $\varepsilon_{\text{eff}}$ can be given using the equation (5) below:

$$
\varepsilon_{\text{eff}} = \frac{\varepsilon_r + 1}{2} + \frac{\varepsilon_r - 1}{2} \left[ (1 + 12 \frac{h}{w})^{1/2} + 0.04(1 - \frac{w}{h})^2 \right]. \quad (5)
$$

To understand the performance of the proposed transition filter-1, the transmission coefficients obtained by using IE3D are plotted in Fig. 6.

![Fig. 6. Transmitted coefficient of the filter-1 for different L6 values.](image)

Figure 6 shows the transmission characteristic of the proposed transition filter-1. The notched band moves toward the lower frequency when increasing the length of L6 and L6=4.25 mm is selected for the optimum design. In this case, we can design the notch band to use at 6.8 GHz band. To further study the nature of the performance of this filter, the current density distribution on the proposed transition filter-1at 6 GHz in the pass-band, is shown in Fig. 7, meanwhile the notch band at 6.8 GHz is obtained.

![Fig. 7. Current density distribution on the proposed transition filter-1.](image)

It can be seen from Fig. 7, that the resonator neither resonates nor affects on the overall performance at 6 GHz in the pass-band, while the maximum current density distribution is on the stub that produces the 6.8 GHz notch band. Figure 7 (b) clearly shows that the resonator at the notched frequency is acting as a short circuit and no coupling exists on the stub at the output port. As mentioned above, the notch band can be easily controlled by adjusting the stub dimensions. In particular, the length of the stub decides the center frequency of the notch-band and the frequency bandwidth of the notch-band is determined by the stub width. Next, we will discuss the transition filter-2.

The theoretical design and synthesis of a compact UWB BPF with dual-notch-band features generated by using two inserted stubs, is shown in Fig. 8. Technically, the proposed filter design essentially exploits the open stub structures to realize dual sharp and adjustable notch-band characteristics. The corresponding lumped equivalent circuits and the equivalent circuit network of the proposed transition filter-2, are shown in Fig. 9.

![Fig. 8. Geometry of proposed transition filter-2.](image)

![Fig. 9. Equivalent circuit network of proposed transition filter-2.](image)

The middle ring resonator can be analyzed in terms of even and odd modes. Even-one, even-two and odd-mode equivalent circuits for the ring
resonator fed by the interdigital-coupled lines in Fig. 8, are shown in Figs. 10 (a), (b) and (c), respectively. From the odd-mode equivalent circuits, we can see that the odd-mode only designs for the pass band. On the other hand, the even-mode designs not only for the pass-band but also for the notch-band. In this paper, we design an UWB pass-band filter with two sharp and adjustable notch-bands. We only analyze the resonance condition for these even-modes and the even-mode resonance condition can be achieved for \( Y_{m} = 0 \).

\[
R = (K^{m})^{-1} \tan \theta + \frac{1}{2} \frac{(K')^{-1} \tan \theta + K^{-1}}{1 - \frac{1}{2} (K')^{-1} \tan \theta + K^{-1}} \]  \tag{7}

In the equation above, \( K, K', K^{m} \) and \( K'' \) can be expressed as:

\[
K = \frac{Z_{1}}{Z_{1}}, \quad K' = \frac{Z_{1}}{Z_{1}}, \quad K^{m} = \frac{Z_{1}}{Z_{1}} \quad \text{or} \quad \frac{Z_{1}}{Z_{1}}.
\]

In this filter, the design parameters are listed below (unit: mm): \( L_{1v}=5.12, L_{1h}=5.92, L_{2}=4.25, L_{3}=0.25, L_{4}=5.39, W_{1}=0.46, W_{2}=5.2, W_{3}=1.3, W_{4}=0.11 \) and \( W_{5}=1.15 \). Next, we use these parameters to simplify the even-mode resonance condition. Based on these parameters, we can get \( \theta_{5(7)} \approx 70^\circ \) and then the length of the stub is close to \( 7\lambda_{notch}/36 \). To understand the performance of the notch bands, the transmitted coefficients of the transition filter-2 are plotted in Fig. 11.

Figure 11 (a) shows the transmission property of the mentioned transition filter-2 with different
L5. By increasing L5, the notched band moves toward the lower frequency. The higher notch band can be tuned from 7 GHz to 9.2 GHz. Figure 11 (b) shows the transmission property of the transition filter-2 for different L7. It is found that the notched band moves toward the lower frequency with the increment of L7 and the notch band can be tuned from 4 GHz to 7 GHz. In this paper, L5=3.35 mm and L7=6.25 mm are selected to investigate the transition filter-2. In this case, the two notch bands are located at 4.9 GHz and 7.8 GHz. To investigate the operation property of the transition filter-2, the current density distribution on the proposed middle ring resonator in both the pass-band and notch-band are illustrated in Fig. 12.

![Fig. 12. Current density distribution on the transition filter-2.](a) 7.8 GHz (b) 4.9 GHz (c) 6 GHz

It can be seen from Fig. 12 that the current density distributions are on the transition filter-2 at 6 GHz in the pass-band. Thus, the resonator neither resonates nor affects the overall performance. At the notched frequencies 4.9 GHz and 7.8 GHz, the resonator has a focus of current density on the stubs. At the input port, the current is high while the current at the output port is low. Thus, the signal is prevented by the notch characteristics. Figures 12 (a) and (b) clearly show that the resonator at the notched frequency is acting as a short circuit and no coupling exists over the stub at the output port. As mentioned above, the frequency of the notched band can be easily controlled by adjusting the stub dimensions. In particular, the stub length decides the center frequency of the notch band and the stub width decides the bandwidth of the notch-band. For the geometry shown in Fig. 13, the proposed UWB filter with tri-notch-band characteristics, referred to as transition filter-3, is investigated in [14]. The transition filter-3 is composed of two interdigital hairpin resonator units, middle ring-stub multimode resonator and two 50-Ω SIR fed structures. Several folded stubs are inserted to the middle ring resonator section to achieve the tri-notch-bands.

![Fig. 13. Geometry of the proposed transition filter type three.](image)

Based on the analysis and the simulated results mentioned above, an UWB filter with multi-notch-band characteristic is proposed, as shown in Fig. 1. To analyze this filter in detail, we decompose the filter to 5 resonators. Firstly, we embedded the resonator-1 and resonator-2 to the simplified prototype filter to create the dual-notch-band characteristic. The first notch-band and the second notch-band are realized by using the resonator-1 and the resonator-2, respectively. The notch characteristics of this filter are obtained by using IE3D, as shown in Fig. 14 (a). We can see from Fig. 14 (a), that with increase of the length of L8, the center frequency of the second notch band moves to the lower frequency, while the center frequency of the first notch band changes slightly. Thus, the second notch band can be tuned by adjusting the length of L8.

Secondly, the resonator-1, resonator-2 and resonator-3 are inserted into the simplified
prototype filter to create tri-notch-band characteristic. At the beginning of this design, a general stub resonator is used for designing resonator-3. The notch characteristics of this filter are shown in Fig. 14 (b). With increasing the length of $L_{10} + L_{11}$, the center frequency of the third notch band moves to the lower frequency. In addition, the center frequency of the first notch band also shifts to the lower side. For getting a method to adjust to the third notch-band that has no effects on the first and the second notch bands, we use a Stepped Impedance Resonator (SIR) stub instead of the general stub resonator. The SIR-stub has another key parameter to adjust the notch-band. The characteristics of the SIR-stub are shown in Fig. 14 (c). With the increase of the width of $W_{11}$, the center frequency of the third notch band moves to the lower frequency and the other two notch-bands remain unchanged. These three notch-bands can be effectively adjusted.

Thirdly, we insert another stub to the designed tri-notch-band filter using resonator 1, resonator 2, resonator 3 and resonator 4, to create four-notch-band characteristics. The fourth notch-band is generated by the resonator-4. The notch band characteristic is simulated by using IE3D and the simulated results are shown in Fig. 14 (d). With the increase of the length of $L_9$, the central frequency of the fourth notch band moves towards the lower direction. At the same time, the central frequencies of the other notch bands are changed slightly. So we can control the fourth notch band by adjusting the dimension of $L_9$.

Finally, another resonator is inserted to the four-notch-band filter to create the final filter, which is obtained by using the resonator-5. The notch characteristics of the constructed filter are simulated by using IE3D and the simulated results are shown in Fig. 14 (e). With the increase of the length of $L_{12}$, the central frequency of the fifth notch band moves towards the lower frequency, while the central frequencies of the other notch bands keep nearly constant. Thus, we can control the fifth notch band by adjusting the dimension of $L_{12}$. According to the analytical and the simulated results above, the designed multi-notch-band characteristics can be simultaneously obtained by choosing the proper dimensions of the middle ring-stub multi-mode resonator, the stubs and the SIR.
To understand the filter property further, the current density distribution on the proposed final filter is investigated at several frequencies, as shown in Fig. 15; in which Figs. 15 (a) and (b) show the current density distribution on the multi-notch-band UWB filter at 4.5 GHz and 9 GHz in the pass-band. It can be seen that the current density distributions on the input and output port are significant, while the current density distribution on the multi-mode resonator is smaller at 4.5 GHz and 9 GHz in the pass-band; implying that the signal can be transformed from the input port to the output port in the pass-band and the multi-mode resonator has no effects on the pass band. Figures 15 (c)-(g) show the current density distributions on the proposed filter at the five notch bands. It is observed that the current density distributions on the input port are strong, while they are smaller on the output port and the current density distribution on the multi-mode resonator is very large at the notch-band frequency; implying that the signal cannot be transformed from the input port to the output port on the notch-band. These signals are rejected by the multi-mode resonator. Thus, the five notched bands can be obtained at 3.9 GHz, 5.25 GHz, 5.9 GHz, 6.8 GHz and 8 GHz. Figures 15 (h) and (i) illustrate the current density distributions on the proposed multi-notch-band UWB filter on the stop band. It can be seen that the current density distributions at 2.5 GHz and 12 GHz are mainly concentrated on the input port. The current density distributions are smaller on the multi-mode resonator in the stop-band, which means that the signals cannot be transformed from the input port to the output port in the stop-band. These signals are reflected by the interdigital coupled lines.

III. RESULTS AND DISCUSSIONS

To evaluate the performance of the multi-notch band UWB filter, the parameters of the designed filter are optimized numerically through IE3D. Optimal parameters of the five-notched band UWB filter are listed in Table 1. To verify the effectiveness of the proposed filter, the proposed filter is fabricated, which is shown in
Fig. 16. The filtering performance was measured by using Anristu 37347D vector network analyzer. Figure 17 demonstrates the frequency responses of the proposed filter. The measured results agree well with the simulated results. The discrepancies between the simulated and measured results may be caused by the fabrication errors.

Table 1: Dimensions of the proposed tri-notch band UWB filter

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Size (mm)</th>
<th>Parameter</th>
<th>Size (mm)</th>
</tr>
</thead>
<tbody>
<tr>
<td>L1v</td>
<td>5.12</td>
<td>W1</td>
<td>0.46</td>
</tr>
<tr>
<td>L1h</td>
<td>5.92</td>
<td>W2</td>
<td>4.4</td>
</tr>
<tr>
<td>L2</td>
<td>4.25</td>
<td>W3</td>
<td>0.11</td>
</tr>
<tr>
<td>L3</td>
<td>5.4</td>
<td>W4</td>
<td>0.11</td>
</tr>
<tr>
<td>L4</td>
<td>5.4</td>
<td>W5</td>
<td>1.5</td>
</tr>
<tr>
<td>L5</td>
<td>2</td>
<td>W6</td>
<td>0.4</td>
</tr>
<tr>
<td>L6</td>
<td>0.25</td>
<td>W7</td>
<td>0.4</td>
</tr>
<tr>
<td>L7</td>
<td>13.8</td>
<td>W8</td>
<td>0.4</td>
</tr>
<tr>
<td>L8</td>
<td>6.5</td>
<td>W9</td>
<td>0.1</td>
</tr>
<tr>
<td>L9</td>
<td>4.3</td>
<td>W10</td>
<td>0.4</td>
</tr>
<tr>
<td>L10</td>
<td>1</td>
<td>W11</td>
<td>0.2</td>
</tr>
<tr>
<td>L11</td>
<td>3.1</td>
<td>W12</td>
<td>0.1</td>
</tr>
<tr>
<td>L12</td>
<td>4.8</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

We can see that the fabricated filter has a measured pass-band from 3.1 GHz to 10.7 GHz with five notch bands, while the center frequencies of the notched bands are located at 3.86 GHz, 5.2 GHz, 5.9 GHz, 6.82 GHz and 7.95 GHz. The group delays shown in Fig. 18 are 0.2 ns and 0.6 ns at the six pass-bands, respectively. It is worth noting that the ring-stub multi-mode resonator can generate five notched bands at the desired frequencies with no significant influence on the wide pass-band performance. Moreover, the proposed UWB BPF has a good five-notch-band characteristic for implementing the functions of UWB radio system.

Fig. 17. Comparison between the simulated and measured results of the fabricated filter.

IV. CONCLUSIONS

In this paper, a compact UWB band-pass filter with five ultra-narrow notch-band characteristic has been proposed and has been verified experimentally and numerically. The design procedures are described in details and investigated by using IE3D. By inserting a ring-stub multi-mode resonator with various stubs into an original UWB BPF, the multi-notch-band functions are obtained to reject undesired signals, such as WiMAX (3.5 GHz band), WLAN (5.2 GHz and 5.8 GHz bands), 6.8 GHz RFID and X-band (7.25 GHz to 8.0 GHz) for satellite communication applications. Simulated and measured results have demonstrated that the ring-stub multi-mode resonator can give five narrow notched bands at the undesired radio signals with no significant influence on the wide pass-band performance. The proposed filter is promising for use in UWB systems due to its simple structure, compact size and excellent performance.
ACKNOWLEDGMENT

This work was partially supported by National Defense "973" Basic Research Development Program of China (No. 6131380101). This paper is also supported by Pre-Research Fund of the 12th Five-Year Plan (No.4010403020102) and Fundamental Research Funds for the Central Universities (HEUCFT1304). The authors are also thankful to Hebei VSTE Science and Technology Co., Ltd., for providing the measuring facility.

REFERENCES

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A Hybrid MoM-PO Method Combining ACA Technique for Electromagnetic Scattering from Target above a Rough Surface

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Abstract — In this paper, an efficient hybrid method of moments (MoM)-physical optics (PO) method combining adaptive cross approximation (ACA) technique is applied to calculate the electromagnetic scattering from three-dimensional (3-D) target and rough surface composite model. The current on the rough surface is obtained through the PO approximation, while the current on the target surface is obtained through the MoM. Furthermore, an ACA technique is used to accelerate the coupling interaction between the target and the rough surface. Numerical results demonstrate that the memory and time cost can be substantially reduced without losing precision by applying the hybrid method, and which can be used to analyze large scale target/rough surface scattering problems.

Index Terms — Adaptive cross approximation (ACA), electromagnetic scattering, and MoM-PO, rough surface.

I. INTRODUCTION

The electromagnetic scattering calculation of target and rough surface composite model has been applied in the fields of radar surveillance, microwave remote sensing, target recognition and target tracking extensively [1-6]. The solutions of the composite scattering problems are complicated but practical.

Some numerical methods have been developed for three-dimensional (3-D) target/rough surface scattering problems, e.g., the finite-difference time-domain (FDTD) algorithm [7-8], a hybrid Kirchhoff approximation (KA)-method of moments (MoM) algorithm [9], multilevel UV method [10-11], the MoM using higher order basis functions [12], the hybrid MoM-physical optics (PO) method [13], most of which are based on the MoM.

The conventional MoM yields a dense complex linear system, which is a serious handicap especially for electrically large scattering problems. Some hybrid methods such as MoM-PO [13-14] are applied to reduce the computation time and memory requirement substantially, while the results are in reasonable agreement with those based on an application of the MoM alone.

In [15], an adaptive cross approximation (ACA) algorithm is used to accelerate MoM computations of electromagnetic compatibility (EMC) problems. It takes advantage of the rank-deficient character of the coupling matrix blocks representing well-separated MoM interactions [16-18]. The ACA algorithm has several important advantages over the multilevel fast multipole algorithm (MLFMA) [19-25]. The beauty of the ACA algorithm is its purely algebraic characteristic. Thus, the development and implementation of ACA algorithm do not depend on the complete knowledge of the integral.
equation kernel, basis functions or the integral equation formulation itself. Moreover, due to its algebraic characteristic, ACA can be modular and very easily integrated into various MoM codes.

In this paper, an efficient hybrid MoM-PO method combining ACA technique is applied to calculate the electromagnetic scattering from 3-D target and rough surface composite model. Both the target and rough surface are assumed to be perfect electric conductor (PEC). Numerical results demonstrate that the memory and time cost can be substantially reduced without losing precision by applying the hybrid method, and which can be used to analyze large scale target/rough surface scattering problems.

II. FORMULATIONS

A. MoM-PO formulation

According to Fig. 1, the surface of the scattering body is split into a MoM-region and a PO-region, which correspond to a target and a rough surface, respectively. In principle, this subdivision can be performed in an arbitrary manner. We divide the scattering body in the manner aiming at making a tradeoff between solution accuracy and efficiency.

![Fig. 1. Composite scattering model of target above a rough surface.](image)

The surface currents of MoM-region and PO-region can be expanded by RWG basis function, written as,

\[ \mathbf{J}^{\text{MoM}} = \sum_{n=1}^{N_{\text{MoM}}} \alpha_n \mathbf{f}_n \]  
\[ \mathbf{J}^{\text{PO}} = \sum_{k=1}^{N_{\text{PO}}} \beta_k \mathbf{f}_k \],

where \( N_{\text{MoM}} \) and \( N_{\text{PO}} \) denote the number of unknowns in MoM-region and PO-region respectively, \( \alpha_n \) and \( \beta_k \) are expansion coefficients of \( \mathbf{f}_n \) and \( \mathbf{f}_k \), both of which are RWG basis functions [26].

In the hybrid MoM-PO method, the relationship between the current in PO-region, the incident field, the current in MoM-region could be expressed as,

\[ \mathbf{J}^{\text{PO}}(\mathbf{r}) = 2\mathbf{n} \times \mathbf{H}^{\text{inc}}(\mathbf{r}) + \sum_{n=1}^{N_{\text{MoM}}} 2\alpha_n \mathbf{n} \times L^H \mathbf{f}_n \]  \hspace{1cm} (3)

where \( \mathbf{H}^{\text{inc}}(\mathbf{r}) \) denotes the incident magnetic field, \( L^H \) is the magnetic field integral operator and \( L^H \mathbf{f}_n = \nabla \times \int_S \mathbf{f}_n(\mathbf{r}') \cdot g(\mathbf{r}, \mathbf{r}')dS' \), here \( g(\mathbf{r}, \mathbf{r}') = e^{-j|\mathbf{r} - \mathbf{r}'|/4\pi|\mathbf{r} - \mathbf{r}'|} \), the Green’s function of free space, \( \mathbf{r}' \) and \( \mathbf{r} \) denote the locations of source and observation point, respectively, \( \mathbf{n} \) denotes the unit outward normal vector of the conductor surface.

In order to get the expansion coefficient \( \beta_k \), the two unit vectors \( \hat{\mathbf{t}}_k^\pm \) are introduced in the middle of the \( k \)th edge. The \( \hat{\mathbf{t}}_k^\pm \) are respectively lying in the plane of the triangles \( T_k^\pm \) defined by the \( k \)th edge, and perpendicular to the \( k \)th edge. As shown in Fig. 2, \( \mathbf{f}_k(\mathbf{r}_k) \cdot \hat{\mathbf{t}}_k^\pm = 1 \) is valid when the point \( \mathbf{r}_k \) is in the middle of the \( k \)th edge.

![Fig. 2. The \( k \)th edge with two adjacent triangles \( T_k^+ \) and \( T_k^- \).](image)

Multiplying both sides of equation (2) with \( \frac{1}{2}(\hat{\mathbf{t}}_k^+ + \hat{\mathbf{t}}_k^-) \) and inserting equation (3) in the resulting equation lead to,

\[ \beta_k = \tau_k + \sum_{n=1}^{N_{\text{MoM}}} \alpha_n \tau_{n,k} \]  \hspace{1cm} (4)
where \( \tau_k = \left( \hat{t}_k^{+} + \hat{t}_k^{-} \right) \cdot (\hat{n} \times \mathbf{H}^{inc}(\mathbf{r})) \) and \( \tau_{n,k} = \left( \hat{t}_k^{+} + \hat{t}_k^{-} \right) \cdot (\hat{n} \times L^E f_n) \).

For the MoM-region, the electric field integral equation (EFIE) could be written as,

\[
(L^E J^{\text{MoM}})_\text{tan} + (L^E J^{\text{PO}})_\text{tan} = -E^{inc}_\text{tan} \tag{5}
\]

where \( L^E \) is electric field integral operator and

\[
L^E J = jk_0 \eta_0 \int_S \left( \mathbf{I} + \frac{\nabla \nabla}{k_0^2} \right) \mathbf{g} (\mathbf{r}, \mathbf{r}') \cdot \mathbf{J} dS', \text{ here } k_0
\]

and \( \eta_0 \) are the wave number and the wave impedance of free space, \( E^{inc} \) is the incident electric field.

Finally, inserting equations (1), (2), and (4) into equation (5) results in,

\[
\sum_{n=1}^{N^{\text{MoM}}} \alpha_n \left[ L^E f_n + \sum_{k=1}^{N^{\text{PO}}} \tau_{n,k} \cdot L^E f_k \right] = -E^{inc}_\text{tan} - \sum_{k=1}^{N^{\text{PO}}} \tau_k \cdot (L^E f_k) \text{.} \tag{6}
\]

Testing equation (6) with RWG basis functions in MoM-region, we can achieve the matrix equation expressed as,

\[
(Z^{\text{MoM}} + Z^{\text{MoM},PO} \cdot \tau') I^{\text{MoM}} = V - Z^{\text{MoM},PO} \cdot \tau \tag{7}
\]

where the \( Z^{\text{MoM}} \) , \( Z^{\text{MoM},PO} \) , \( \tau' \) are \( N^{\text{MoM}} \times N^{\text{MoM}} \) , \( N^{\text{MoM}} \times N^{\text{PO}} \) , \( N^{\text{PO}} \times N^{\text{MoM}} \) complex matrix respectively, \( I^{\text{MoM}} \) and \( V \) are vectors of size \( N^{\text{MoM}} \) , \( \tau \) is vector of size \( N^{\text{PO}} \). The matrix elements are written as,

\[
Z^{\text{MoM}}_{mn} = \langle f_m, L^E f_n \rangle \tag{8}
\]

\[
f_n^{\text{MoM}} = \alpha_n, \tag{9}
\]

\[
V_m = -\langle f_n^{\text{inc}}, E^{inc}_\text{tan} \rangle, \tag{10}
\]

\[
Z^{\text{MoM},PO}_{mk} = \langle f_m, L^E f_k \rangle \tag{11}
\]

\[
\tau'_{kn} = \left( \hat{t}_k^{+} + \hat{t}_k^{-} \right) g_n \hat{n} \times L^E f_n \tag{12}
\]

\[
\tau_k = \left( \hat{t}_k^{+} + \hat{t}_k^{-} \right) g_n \hat{n} \times H^{inc}(\mathbf{r}) \tag{13}
\]

B. The application of ACA algorithm in MoM-PO

For the composite scattering problems of target above rough surface, the matrix \( Z^{\text{MoM},PO} \) and \( \tau \) for interaction between MoM-region and PO-region have rank-deficient characters because the distance between source point and observation point is relatively far. The ACA algorithm fast achieves the low-rank decomposition form by using the rank-deficient character of matrix [15, 17]. The basic principle of ACA algorithm is as follow. The low-rank representation of a matrix could be got by the elements of partial rows and columns but not all the matrix elements. It means that by selecting right rows and columns, we can get the singular value decomposition form of the matrix approximately, so as to achieve the purpose of improving the computational efficiency.

Let the \( m \times n \) rectangular matrix \( Z^{\text{MoM}} \) represent the interaction between two well-separated cubes. The ACA algorithm aims to approximate \( Z^{\text{MoM}} \) by \( \tilde{Z}^{\text{MoM}} \) in the following form,

\[
\tilde{Z}^{\text{MoM}} = U^{\text{MoM}} V^{\text{MoM}} + R^{\text{MoM}} \tag{14}
\]

where \( r \) is the effective rank of the matrix \( Z^{\text{MoM}} \). The goal of ACA is to achieve,

\[
\| R^{\text{MoM}} \|_F = \| Z^{\text{MoM}} - \tilde{Z}^{\text{MoM}} \|_F \leq \epsilon_{\text{ACA}} \| Z^{\text{MoM}} \|_F \tag{15}
\]

for a given tolerance \( \epsilon_{\text{ACA}} \), where \( R \) is termed as the error matrix, \( \| \|_F \) denotes the matrix Frobenius norm. If \( r \leq \min(m,n) \), the memory requirement will be reduced significantly from \( m \times n \) to \( (m + n) \times r \).

Selecting the value of the \( \epsilon_{\text{ACA}} \) is very important. If the \( \epsilon_{\text{ACA}} \) is too small, the computational cost will be high, while if the \( \epsilon_{\text{ACA}} \) is too big, the computational accuracy will be low. Therefore, it is necessary to make a tradeoff. The more details of the ACA algorithm can be found in [15].

III. NUMERICAL EXAMPLES

In this section, several numerical examples are presented to illustrate the validity and efficiency of the proposed method. In these examples, the composite models are illuminated by tapered wave [27], which is employed to avoid rough surface edge scattering effects. The tapered wave is expressed as,

\[
E^{inc}(x, y, z) = \exp[-jk_0(z \cos \theta_x + x \sin \theta_x \cos \phi + y \sin \theta_x \sin \phi)(1 + \omega)] \exp(-t_s - t_y) \tag{16}
\]

where

\[
t_s = \frac{(x \cos \theta_x \cos \phi + y \cos \theta_x \sin \phi + z \sin \theta_x)^2}{g^2 \cos^2 \theta_x}, \tag{17}
\]
\[
t_y = \frac{(-x \sin \phi + y \cos \phi)^2}{g^2}, \quad (18)
\]
\[
\omega = \frac{1}{k^2} \left( \frac{2r_t - 1}{g^2 \cos^2 \theta} + \frac{2r_t - 1}{g^2} \right). \quad (19)
\]

Here, \( \theta_i \) and \( \phi_i \) are the elevation angle and azimuth angle of the incident wave, while \( g \) is the parameter to control the width of the tapered wave. All the computations are performed on a PC with Intel Dual-core 3.1 GHz CPU and 8 GB RAM in double precision. The terminating tolerances of the ACA is set as \( \varepsilon_{ACA} = 0.001 \).

The relative residual error at the \( k \)th iteration is used for monitoring the convergence of the proposed method, which is defined as,
\[
\varepsilon(V, k) = \frac{||V - ZI^{(k)}||_2}{||V||_2} \quad (20)
\]
where \( || \cdot ||_2 \) denotes the 2-norm of the complex vector. The iteration stops when the \( \varepsilon(V, k) \) is less than 0.001.

As the first example, the composite scattering model of a PEC sphere above a PM spectrum rough surface is considered to test the validity of the proposed method. The mesh sizes of the sphere and the rough surface are set as 0.1\( \lambda \) and 0.15\( \lambda \), respectively. The radius of the sphere is 0.5\( \lambda \) and the rough surface size is 24\( \lambda \times 24 \lambda \), whose corresponding numbers of the unknowns are 930 and 76480. The width of the sphere center from rough surface is 2.0\( \lambda \). The height of the tapered wave is 6.0\( \lambda \). The incident wave is from Theta = 30° and Phi = 0°. A total number of 77410 unknowns are involved in this example. Figure 3 shows the RCS results (VV-Polarization) at Phi=0° computed by the proposed method, which agree well with the results computed by the conventional MoM-PO. Figure 4 shows the RCS results (VV-Polarization) at Phi=0° computed by the proposed method, which agree well with the results computed by the conventional MoM-PO. Table II shows that the computational cost can be reduced significantly compared to the conventional MoM-PO.

![Fig. 3. Bistatic RCS of a PEC sphere above a PM spectrum rough surface.](image1)

![Fig. 4. Bistatic RCS of a missile above a Gaussian rough surface.](image2)
Table II: Computational cost of the second example.

<table>
<thead>
<tr>
<th></th>
<th>Memory Requirement(GB)</th>
<th>CPU Time</th>
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<tbody>
<tr>
<td>MoM-PO</td>
<td>6.817</td>
<td>&gt;3 days</td>
</tr>
<tr>
<td>MoM-PO-ACA</td>
<td>0.74</td>
<td>1.5 hours</td>
</tr>
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</table>

IV. NUMERICAL EXAMPLES

In this paper, a hybrid MoM-PO method combining ACA algorithm is applied to solving scattering from composite model of target and rough surface. Numerical examples have demonstrated that the memory requirement and CPU time can be significantly reduced without losing precision by applying the proposed method, and which can be used to analyze large scale target/rough surface scattering problems.

REFERENCES


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Null Broadening and Sidelobe Control Algorithm via Multi-Parametric Quadratic Programming for Robust Adaptive Beamforming

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Abstract — Adaptive beamforming algorithm can automatically optimize the array pattern by adjusting the elemental control weights until a prescribed objective function is satisfied. Unfortunately, it is possible that the mismatch occurs between adaptive weights and data, due to the perturbation of the interference location when the antenna platform vibrates or interference moves quickly. Besides, the traditional beamformers may have unacceptably high sidelobes when few samples are available. To solve these problems, an effective robust adaptive beamforming method is presented. In the proposed method, firstly, a tapered covariance matrix is constructed to broaden the width of nulls for interference signal sources. Secondly, multiple additional quadratic inequality constraints outside the mainlobe beampattern area are used to guarantee that the sidelobe level is strictly lower than the prescribed threshold value. Finally, the beamforming optimization problem is formulated as a multi-parametric quadratic programming problem, such that the optimal weight vector can be easily obtained by real-valued computation. Simulation results are shown to demonstrate the efficiency of the proposed approach.

Index Terms — Covariance Matrix Taper (CMT), multi-parametric Quadratic Programming (mp-QP), null broadening, robust adaptive beamforming and sidelobe control.

I. INTRODUCTION

Adaptive beamforming has been found in numerous applications from radar, sonar, wireless communications, seismology and microphone arrays. One of the most popular approaches to adaptive beamforming is the so-called Minimum Variance Distortionless Response (MVDR) processor, which minimizes the array output power while maintaining a distortionless mainlobe response toward the desired signal [1]. However, most of the conventional adaptive beamformers, such as MVDR, etc., may have unacceptably low and narrow null level in the interference direction or high sidelobes in the case of low sample support. In adaptive array systems, these may lead to significant performance degradation in the case of unexpected interference signals.

Several approaches of null broadening technique have been proposed. For example, the null broadening technique originally [2-3] is developed for robust beamforming. It is generalized by the concept of a “Covariance Matrix Taper (CMT)” [4]. A multi-parametric quadratic programming method is presented to control the null level of adaptive antenna array [5]. In the proposed method, the optimal weight vector can be easily obtained by real-valued computation. Unfortunately the sidelobe level is not controlled efficiently. A null broadening technique based on reduced rank conjugate gradient algorithm is proposed in [6]. It can obtain better performance.
nulls, but also work faster than the particle swarm optimization technique is used to obtain optimal null levels for the symmetric linear antenna array. By the linearization of the transmit model with Taylor expansion, a fast broad null beamforming technology is presented in [11] and the application of the broad null transmitting beamforming technology for the radar in anti-ARM battle is studied. A null steering beamforming algorithm is proposed to cancel unwanted signals by steering nulls of the pattern in the direction of high interference without affecting the main beam [12].

To control sidelobe, several approaches have been proposed [13-16]. A modified MVDR beamformer is presented by multiple additional quadratic inequality constraints outside the mainlobe beampattern area [13]. These constraints can guarantee that the beampattern sidelobe level remains lower than a certain prescribed value. An improved adaptive beamforming technique [14] is proposed by an adaptive dispersion invasive weed optimization. It can not only provide sufficient steering ability regarding the mainlobe and the nulls, but also work faster than the particle swarm optimization. An iterative beamforming method is proposed for sensor array with arbitrary geometry and element directivity [15]. By solving the linearly constrained least squares problem, it can effectively control the sidelobe level. Making use of multi-linear constrained minimum variance repetitiously, a low sidelobe beamforming method is presented in [16]. By searching the previously formed beampattern, the location of the highest sidelobe is found and the corresponding direction vector is added to the constrained conditions of multi-linear constrained minimum variance algorithm to receive the new weight value.

In this paper, an effective adaptive beamforming method is proposed to resolve null broadening and sidelobe control problem. Firstly, an approximation of sinc function is used to broaden null. It can be regarded as adding the coherent signals to each signal source but with two different directions, and it can place nulls in a certain range of angles instead of certain points. Secondly, multiple quadratic inequality constraints outside the mainlobe beampattern area are used to control sidelobe. These constraints can guarantee that the beampattern sidelobe level remains lower than a certain prescribed value. Thirdly, the beamforming control problem is formulated as a multi-parametric Quadratic Programming (mp-QP) problem, such that the optimal weight vector can be obtained by real-valued computation.

This paper is organized as follows. section 2 briefly introduces the signal model and presents the MVDR solution. The methods of null broadening and sidelobe control are addressed in section 3. Section 4 gives the algorithm formulation. In section 5, simulation results are presented to verify the performance of the proposed approach. Section 6 concludes the paper.

II. BACKGROUND

Consider a Uniform Linear Array (ULA), which consists of M elements. The beamformer output of the ULA at time t is given by:

\[ y(t) = w^{H} x(t), \]

where \( w = [w_1, w_2, \ldots, w_M]^T \) is the complex-valued weight vector. The superscripts \((\cdot)^T\) and \((\cdot)^H\) denote the transpose and conjugate transpose of a matrix, respectively. The \( M \times 1 \) vector of array observations \( x(t) = [x_1(t), x_2(t), \ldots, x_M(t)]^T \) is given by:

\[ x(t) = s(t) + i(t) + n(t) \]

\[ = s(t) a(\theta_0) + \sum_{j=1}^{J} i_j(t) a(\theta_j) + n(t), \]

where \( J \) is the number of interference signals. \( s(t) \) and \( i_j(t) \) stand for the signal and interference, respectively. The signal and interference Directions of Arrival (DOAs) are \( \theta_0 \) and \( \theta_j \) respectively.
\( \theta_j \) (\( j = 1, \ldots, J \)), respectively, with corresponding steering vectors \( a(\theta_0) \) and \( a(\theta_j) \). \( n(t) = [n_1(t), \ldots, n_M(t)]^T \) with \( n_i(t) \) denoting the additive noise of the \( i \)th sensor.

Let \( R \) denote the \( M \times M \) theoretical covariance matrix of the array snapshot vector. Assume that \( R \) is a positive definite matrix with the following form:

\[
R = \sigma^2 w a(\theta_0) a^H(\theta_0) + \sum_{j=1}^{J} \sigma_j^2 a(\theta_j) a^H(\theta_j) + \sigma_n^2 I_M,
\]

where \( \sigma^2, \sigma_j^2 (j = 1, \ldots, J) \) and \( \sigma_n^2 \) are the powers of the uncorrelated impinging signals \( s(t) \), \( i_j(t) \) and noise, respectively. \( I_M \) is the \( M \times M \) identity matrix. The common formulation of the beamforming problem that leads to the MVDR beamformer is described below. First determine the \( M \times 1 \) vector \( w_0 \) as the solution to the following linearly constrained quadratic problem:

\[
\min_w w^H R w \quad \text{subject to} \quad w^H a(\theta_0) = 1. \tag{1}
\]

Then the solution of equation (1) for this particular case can be given as:

\[
w_{\text{MVDR}} = \frac{R^{-1} a(\theta_0)}{a^H(\theta_0) R^{-1} a(\theta_0)}. \tag{2}
\]

In practice, the exact covariance matrix is not available and is replaced by the sample covariance matrix \( \hat{R} \),

\[
\hat{R} = \frac{1}{N} \sum_{k=1}^{N} \mathbf{x}(k) \mathbf{x}^H(k), \tag{3}
\]

where \( N \) denotes the number of snapshots.

III. ADAPTIVE BEAMFORMING WITH NULL BROADENING AND SIDELOBE CONTROL

In this section, we firstly introduce the null broadening method to widen the nulling extent and control the nulling level. Secondly, multiple quadratic inequality constraints are derived to control sidelobe level. Finally, the modified CMT-MVDR problem is given.

A. Null broadening

1. Null extent control: Assume that the narrowband interference signals impinging on the array are uncorrelated with each other as well as with the spatially white noise. According to [2], the terms in the covariance matrix \( R \) for a one-dimensional array are given as:

\[
R_{mn} = N a \delta(m,n) + \sum_j \sigma_j^2 e^{2\pi j (x_{m}-x_{n})/\lambda},
\]

where \( x_m \) is the location of the \( m \)th element. \( N_a \) is the noise power in the \( n \)th channel and \( \delta(m,n) \) is a Kronecker delta function. The sum is performed over all interference signals with averaged power \( \sigma_j^2 \) and direction cosines \( u_j = \sin \theta_j \), for \( \theta_j \) measured from the broadside. According to [2], we construct a cluster of \( q \) equal-power incoherent signals around each original interfering signal to produce a notch of width \( W \) in each of the interference directions. In this case, the additional sources can be summed in closed form, as geometric sum and can be written as:

\[
\sum_{k=1}^{q} (\sigma_j^2 / q) e^{2\pi j (x_{m}-x_{n})/\lambda} = \frac{\sin(q \Lambda_{m,n})}{q \sin(\Lambda_{m,n})} \sigma_j^2 e^{2\pi j (x_{m}-x_{n})/\lambda},
\]

where \( \Lambda_{m,n} = \pi(x_{m}-x_{n})/\lambda \) and \( \delta = W/q - 1 \). Since there is no angle dependence in the sinc function. A new covariance matrix term is given as \( \tilde{R}_{mn} = K_{mn} \frac{\sin(q \Lambda_{m,n})}{q \sin(\Lambda_{m,n})} \). In matrix form, the CMT can be expressed as:

\[
\tilde{R} = K \circ \tilde{I}_M,
\]

where “\( \circ \)” represents Hadamard product, that is multiplying the corresponding elements of the two matrices and the form of matrix \( T_{mn} \) is:

\[
T_{mn} = (\sin(q \pi \delta(m,n)/2)) / (q \sin(q \pi \delta(m,n)/2)).
\]

2. Null level control: Assume that the interference signal arrivals the received array from the angle of incident \( \theta_p \), \( (p = 1, \ldots, P) \). When the interference moves quickly, it is possible that the mismatch occurs for adaptive weight and data, due to the perturbation of the interference location. Let \( \Delta \theta \) denote the angle spread for the interference signal, which comes from \( \theta_p \). Let \( \Theta_k = [\theta_p - \Delta \theta, \theta_p + \Delta \theta] \) (\( k = 1, \ldots, K \)) be chosen grid that approximates the angle spread area. To control the null level for the angle spread area \([\theta_p - \Delta \theta, \theta_p + \Delta \theta]\), we use the following
multiple quadratic inequality constraints inside the angle spread area:
\[ |w^H a(\theta_k)|^2 \leq \xi^2, \quad k = 1, \ldots, K, \]  
where \( \xi^2 \) is the prescribed null level.

B. Sidelobe control

Let \( \theta_k \in \Theta(k = 1, \ldots, L) \) be a chosen grid that approximates the sidelobe beampattern areas \( \Theta \) using a finite number of angles. To control the sidelobe level, we use the following multiple quadratic inequality constraints outside the mainlobe beampattern area:
\[ |w^H a(\theta_k)|^2 \leq \varepsilon^2, \quad k = 1, \ldots, L, \]  
where \( \varepsilon^2 \) is the prescribed sidelobe level.

C. The modified CMT-MVDR

Adding the constraints (5) and (6) to the MVDR beamforming problem (1) and using the new tapered covariance matrix (4) instead of the sample covariance matrix (3), we obtain the following modified CMT-MVDR problem:
\[
\text{minimize} \quad w^H \hat{R} w \quad \text{subject to} \quad \begin{align*}
|w^H a(\theta_k)|^2 & \leq \xi^2, \\
|w^H a(\theta_l)|^2 & \leq \varepsilon^2,
\end{align*}
\]  
where \( k = 1, \ldots, K \) and \( l = 1, \ldots, L \).

Notice that \( |w^H a(\theta_k)|^2 \) can directly determine the output power of antenna array at the interference direction \( \theta_k \) (refer to equation (17)), thus, it can be viewed as the “directional gain” of the antenna array.

In the next section, we will convert this problem (7) to an mp-QP problem, such that the optimal weight vector is estimated by the real-valued computation.

IV. ALGORITHM FORMULATION

A. CMT-mp-QP MVDR

In this section, we present the multi-parametric programming problem for covariance matrix taper MVDR beamformer, named as CMT-mp-QP MVDR. As seen in problem (7), the data is in general complex valued. However, for convenience, we will work with real-valued data.

To do so, a pre-processing path is taken prior to the beamforming operation.

Let
\[
\begin{align*}
R_1 & = \text{Real}\{\hat{R}\}, \\
R_2 & = \text{Imag}\{\hat{R}\}, \\
w_1 & = \text{Real}\{w\}, \\
w_2 & = \text{Imag}\{w\},
\end{align*}
\]

where \( a_{01}(\theta_0) = \text{Real}\{a(\theta_0)\}, \quad a_{02}(\theta_0) = \text{Imag}\{a(\theta_0)\}, \quad a_{i1}(\theta) = \text{Real}\{a(\theta)\}, \quad a_{i2}(\theta) = \text{Imag}\{a(\theta)\}, \quad a_i(\theta) = \text{Real}\{a(\theta)\}, \quad a_{i2}(\theta) = \text{Imag}\{a(\theta)\}, \)

where \( k = 1, \ldots, K \) and \( l = 1, \ldots, L \). Real{ } and Imag{ } stand for the real and imaginary part of a complex matrix or vector, respectively.

By simple algebra, the cost function \( w^H \hat{R} w \) can be rewritten as:
\[
w^H \hat{R} w = \text{Real}\{w^H \hat{R} w\} + j\text{Imag}\{w^H \hat{R} w\} = \begin{bmatrix} w_1^T \\ w_2^T \end{bmatrix} \begin{bmatrix} R_1 & -R_2 \\ R_2 & R_1 \end{bmatrix} \begin{bmatrix} w_1 \\ w_2 \end{bmatrix} + j\begin{bmatrix} w_1^T \\ w_2^T \end{bmatrix} \begin{bmatrix} R_1 & R_2 \\ -R_2 & R_1 \end{bmatrix} \begin{bmatrix} w_1 \\ w_2 \end{bmatrix}.
\]

It is easy to know that \( \text{Imag}\{w^H \hat{R} w\} = 0 \) since \( (w^H \hat{R} w)^H = w^H \hat{R} w \in \mathbb{R} \) for \( \forall w \in \mathbb{C}^M \). Thus, the modified MVDR problem (7) can be reformulated as the following mp-QP problem:
\[
\min \quad \frac{1}{2} z^T Hz \quad \text{subject to} \quad Gz \leq b, \tag{8}
\]
where the matrices and vectors in equation (8) have the following forms:
\[
\begin{align*}
z & = [w_1^T, w_2^T] \in \mathbb{R}^M, \\
H & = \begin{bmatrix} R_1 & -R_2 \\ R_2 & R_1 \end{bmatrix} \in \mathbb{R}^{2M \times 2M}, \\
G & = [G_0^T, G_1^T, \ldots, G_K^T, G_{i1}^T, \ldots, G_{ij}^T]^T, \\
G_k & = [B_k^T, -B_k^T]^T, \\
G_{ij} & = [B_i^T, -B_j^T]^T, \\
B_k & = \begin{bmatrix} a_{i1}(\theta_k) \\ a_{i2}(\theta_k) \end{bmatrix} - \begin{bmatrix} a_{i1}(\theta_k) \\ a_{i2}(\theta_k) \end{bmatrix}, \\
B_{ij} & = \begin{bmatrix} a_{i1}(\theta_k) \\ a_{i2}(\theta_k) \end{bmatrix} - \begin{bmatrix} a_{i1}(\theta_k) \\ a_{i2}(\theta_k) \end{bmatrix}, \\
b & = [b_0^T, b_1^T, \ldots, b_K^T, b_{i1}^T, \ldots, b_{ij}^T]^T, \\
b_j & = [\sqrt{1-\lambda_j}, \sqrt{1-\lambda_j}, \sqrt{1-\lambda_j}, \sqrt{1-\lambda_j}, \sqrt{1-\lambda_j}, \sqrt{1-\lambda_j}, \sqrt{1-\lambda_j}, \sqrt{1-\lambda_j}]^T, \\
b_i & = [\sqrt{1-\lambda_i}, \sqrt{1-\lambda_i}, \sqrt{1-\lambda_i}, \sqrt{1-\lambda_i}, \sqrt{1-\lambda_i}, \sqrt{1-\lambda_i}, \sqrt{1-\lambda_i}, \sqrt{1-\lambda_i}]^T.
\end{align*}
\]
where $k=1,...,K$ and $l=1,...,L$. $\lambda_z$ and $\lambda_l \in [0,1]$. The matrix $H$ is a positive definite matrix.

B. The optimal solution

As shown in [17] and [18], CMT-mp-QP problem (8) can be solved by applying the Karush-Kuhn-Tucker (KKT) conditions,

$$Hz + G^T \lambda = 0, \quad \lambda \in \mathbb{R}^{4(K+1)},$$

$$\lambda_i G^i z - b^i = 0, \quad i=1,...,4(K+1),$$

$$\lambda \geq 0,$$

$$Gz - b \leq 0.$$  

(9)

(10)

(11)

(12)

In the sequel, let the superscript index denote a subset of the rows of a matrix or vector. Since $H$ has full rank, equation (9) gives:

$$z = -H^{-1} G^T \lambda.$$  

(13)

**Definition 1:** Let $z^*$ be the optimal solution to problem (8). We define active constraints the constraints with $G^i z - b^i = 0$ and inactive constraints the constraints with $G^i z - b^i < 0$. The optimal active set $A^* = \{ i | G^i z^* = b^i \}$.

**Definition 2:** For an active set, we say that the Linear Independence Constraint Qualification (LICQ) holds if the set of active constraint gradients are linearly independent, i.e., $G^i$ has full row rank.

Assuming that LICQ holds, equation (10) and equation (13) lead to:

$$\lambda^* = -(G^4 H^{-1} G^4)^{-1} b^4.$$  

(14)

Equation (14) can now be substituted into equation (13) to obtain:

$$z = H^{-1} (G^4)^T (G^4 H^{-1} G^4)^{-1} b^4.$$  

(15)

Partition, now, the vector $z$ into $z_1, z_2 \in \mathbb{R}^M$, by $z = [z_1, z_2]^T$ and define $w_0$ as follows:

$$w_0 = z_1 + jz_2 \in \mathbb{C}^M.$$  

(16)

It is clear that the optimal solution of problem (7) for this particular case is $w_0$.

**Summary of the proposed algorithm**

1. Collect the sample data and estimate the covariance matrix $\hat{R} = \frac{1}{N} \sum_{k=1}^{N} x(k) x^H(k)$.

2. Modify the sample covariance matrix using the CMT matrix $T$, $\hat{R} = \hat{R} T^T$.

3. Add the constrains of null level control $|w^H a(\theta_j)|^2 \leq \varepsilon^2, (k=1,...,K)$ and sidelobe level control $|w^H a(\theta_l)|^2 \leq \varepsilon^2, (l=1,...,L)$ to the MVDR beamformer (1).


C. Output SINR and array gain

To investigate the performance of the proposed method, this section gives the definitions of Signal-to-Interference-and-Noise Ratio (SINR) and array gain for an adaptive antenna array system. From $y(k) = w^H x(k)$, the mean square power output of the beamformer can be expressed as:

$$P = E \left( |y(k)|^2 \right) = \sigma_0^2 |w^H a(\theta_j)|^2 + w^H R_{jn} w$$

$$= \sigma_0^2 |w^H a(\theta_j)|^2 + \sum_{j=1}^{J} \sigma_j^2 |w^H a(\theta_j)|^2 + \sigma_n^2 w^H p_n w,$$  

(17)

where $E\{\}$ denotes the statistical expectation. The $M \times M$ interference-plus-noise covariance matrix $R_{jn}$ is expressed as:

$$R_{jn} = E \{ p p^H \} = \sum_{j=1}^{J} \sigma_j^2 a(\theta_j) a^H(\theta_j) + \sigma_n^2 p_n,$$

where $p = \sum_{j=1}^{J} j(k) a(\theta_j) + n(k)$ and $\sigma_0^2$ is the desired signal power. $\sigma_j^2 (j=1,...,J)$ and $\sigma_n^2$ are the interference signal power and noise power, respectively. $p_n$ is the Hermitian cross-spectral density matrix of the noise normalized to have its trace equals to $M$.

SINR is defined as follows:

**Definition 3:** For an adaptive antenna array system, SINR is defined as the output signal power divided by the output interference-and-noise power and is given by:

$$\text{SINR}(w) = \frac{\sigma_0^2 |w^H a_0|^2}{w^H R_{jn} w}.$$  

(18)

The signal power $\sigma_0^2$ can be estimated according to the following formulation:
From equation (18) and equation (19), SINRs of CMT-mp-QP MVDR beamformer and mp-QP MVDR beamformer can be achieved, respectively.

**Definition 4** For an adaptive antenna array system, the array gain is defined as the output Signal-to-Interference-and-Noise Ratio (SINR) divided by the input SINR and is given by:

$$G = \frac{\text{SINR}_{\text{out}}}{\text{SINR}_{\text{in}}} = \sigma_n^2 \left| \mathbf{w}^\mathsf{H} a(\theta) \right|^2 / \left( \sigma_n^2 \sum_{j=1}^{J} \sigma_j^2 + \sigma_n^2 \right).$$

(20)

To calculate the output due to the desired signal, a distortionless constraint is imposed on \( \mathbf{w} \) that \( \mathbf{w}^\mathsf{H} a(\theta) = 1 \). Consider the special case of spatial white noise and identical noise spectra at each sensor, the noise cross-spectral density matrix \( \mathbf{R}_n \) reduces to an identity matrix. Thus the array gain for white noise is given by:

$$G = \sum_{j=1}^{J} \sigma_j^2 + \sigma_n^2$$

$$= \sum_{j=1}^{J} \sigma_j^2 \left| \mathbf{w}^\mathsf{H} a(\theta) \right|^2 / \left( \sum_{j=1}^{J} \sigma_j^2 + \sigma_n^2 \right)$$

$$= \sum_{j=1}^{J} \sigma_j^2 / \sigma_n^2 \left| \mathbf{w}^\mathsf{H} a(\theta) \right|^2 + \left\| \mathbf{w} \right\|^2,$$

(21)

where \( \left\| \cdot \right\| \) stands for the Euclidean norm.

From equation (21), the array gain of CMT-mp-QP MVDR beamformer and mp-QP MVDR beamformer can be calculated, respectively, according to the weight vectors given by equation (16) and equation (2).

V. SIMULATION RESULTS

In this section, we conduct some simulations to validate the proposed approach. Assume that the Uniform Linear Array (ULA) consists of seventeen sensors \( (M=17) \) equispaced by half-wavelength. Assume that the desired signal and two interference signals are plane waves impinging on the ULA from the directions 0° and 30°, respectively. In these simulations, the Signal-to-Noise Ratio (SNR) is set to 0 dB, 35 dB and 35 dB for the desired signal and the two interference signals, respectively. The notch width \( W = 9^o \) and \( q = 5 \). The sidelobe beampattern areas \([-90^o , 10^o ] \cup [10^o , 90^o ]\) are chosen and a uniform grid is used to obtain the angles. Sensor noises are modeled as spatially and temporally white Gaussian processes. It is assumed that \( \varepsilon^2 = 10^{-5} \) and \( \varepsilon^2 = 10^{-2} \), i.e., we require the beampatterns null level below -50 dB and sidelobe level to be below -20 dB.

In the first simulation, 1024 snapshots are used to compute the direction patterns of mp-QP MVDR and CMT-mp-QP MVDR, which are plotted in Fig. 1. As shown in this figure, all the beampatterns place deep nulls at the DOAs of the interference signals and maintain a distortionless response for the signal-of-interest. However, mp-QP MVDR is not able to broaden the interference nulls and it has very high sidelobe level. The proposed algorithm can overcome these above shortcomings, with tapered covariance matrix that is and it can not only broaden the null width but also achieve lower sidelobe level.

**Fig. 1.** Directional pattern curves of the proposed CMT-mp-QP method and pure mp-QP method.

In the second simulation, Fig. 2 gives the array gain curves (which are computed by equation (21)) of the aforementioned two beamformers, based on 200 independent trials under the hypothesis that SNRs range from -20 dB to 20 dB and the number of snapshots is equal to 1024. When SNR ≤ 0 dB, mp-QP MVDR has a little better array gain than the proposed method. With the increase of SNR, it can be seen that CMT-mp-QP MVDR shows better array gain.
The third simulation considers that the number of snapshots is varied. Figure 3 shows the average output SINR curves (which are computed by equation (18) and equation (19)) of the aforementioned methods, based on 200 independent trials with the SNR equal to 10 dB. From the figure, it can be seen that the proposed method has better convergence even few samples are available. As the number of snapshots increases, the performances of both methods tend to stabilize, while CMT-mp-QP MVDR has higher output SINR than mp-QP MVDR.

VI. CONCLUSIONS
This paper presents an effective robust adaptive beamforming method with null broadening and sidelobe control. By modifying the measured covariance matrix, null broadening can be regarded as adding the coherent signals to each signal sources. Multiple quadratic inequality constraints outside the mainlobe beampattern area are used to guarantee the beampattern sidelobe level are strictly below some prescribed threshold. Then, the robust adaptive beamforming problem is formulated as a multi-parametric quadratic programming problem, such that the optimal weight vector can be estimated by real-valued computation. The performance of the presented method is verified by simulation.

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Optimal Design of Electromagnetic Absorbers

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Abstract — A procedure for the optimal design of compact and light-weight electromagnetic absorbers is presented. The absorbers are designed to damp resonances inside metallic enclosures on the basis of Jaumann’s theory; several layers of lossy (artificial) dielectrics are separated by two high/low impedance frequency selective surfaces and one resistive sheet. The constitutive parameters of the absorbing layers are optimized by means of the Particle Swarm Optimization method in order to maximize bandwidth and absorption rate of the structure in the GHz frequency range, where typically the first resonances of small enclosures occur.

Index Terms - Absorbing materials, electromagnetic shielding, finite integration technique, particle swarm optimization and shielding effectiveness.

I. INTRODUCTION

The design of layers capable of absorbing Electromagnetic (EM) waves has been afforded either in numerical studies, e.g., [1], [2] and in practical applications [3]. One of the most challenging benchmarks to test a design procedure for EM absorbers is represented by the resonances occurring in typical shielding enclosures. Metallic cabinets behave as overmoded cavities at working frequencies, and the twofold goal of avoiding unwanted emissions and improving immunity of shielded devices is accomplished by damping the EM field in the shielded region. The Shielding Effectiveness (SE) of an enclosure is usually quantified according to the IEEE Std. 299 in terms of the ratio of the incident over the shielded electric (magnetic) field at a selected point inside the enclosure; although, integral definitions have been proposed and adopted [4], [5]. It is well known that resonances can dramatically deteriorate the screening performance of the enclosure, because the excited internal modes can enhance the field in the victim proximity and apertures may increase these effects. Resonances can be suppressed by lining the internal enclosure surfaces with properly designed absorbers.

Research into EM wave absorbers started in the 1930’s, shortly after the advent of radar with the aim of developing materials capable to reduce the radar cross-section (radar absorbing materials) and camouflaging military devices (stealth technologies) [6]. Through the years, the interest in absorbing materials has spread to the commercial sector to reduce interferences resulting from a growing number of EM apparatus, reaching into the radio frequency band [7]. Nowadays, there is a rising demand for lighter weight and more highly absorbing materials in a number of applications [8]. Absorbers can be roughly classified into impedance matching and resonant absorbers, though many absorbers have features of both these classifications; traditional designs include pyramidal, tapered loading, matching layers and ferrite-based absorbers [9], as well as Salisbury [10], [11] and Dallenbach [12] screens, Jaumann absorbers [13], [14] and circuit analog absorbers based on the use of lossy frequency selective surfaces [14], [15].

In this work, a procedure for the design of absorbing materials for damping resonances inside shielded enclosures is presented. The aims of the study are to design and optimize the performances of an absorbing structure in the GHz range, where the first resonances of commercial enclosures occur and to maximize the bandwidth and minimize the reflection coefficient for oblique
angles of incidence. The layered structure is based on Jaumann’s theory and on circuit analogy. The main idea of Jaumann’s theory is based on the achievement of a cancelling interference between incident and reflected wave by means of adequate dimensions and properties of the multi-layer structure. Absorption is enhanced alternating dielectrics and inductive, capacitive or resistive sheets; four layers of lossy dielectrics, one loaded with lossy wires are considered and alternated with three lossy sheets, two of which are lossy capacitive/inductive frequency selective surfaces [16], [17].

II. SYSTEM CONFIGURATION

The electromagnetic problem under analysis is sketched in Fig. 1 (a): a metallic enclosure, whose walls are assumed to be perfectly conducting and infinitesimally thin, presents an aperture on one side. When the enclosure is illuminated by an impinging external field \( \mathbf{E} \) or driven by internal sources, e.g., electric \( \mathbf{J} \) or magnetic \( \mathbf{M} \), dipoles [5], internal resonances can be excited deteriorating the shielding performances of the enclosure. As an example, in Fig. 1 (b), it is reported the electric field shielding effectiveness \( \text{SE}_E \) of a commercial enclosure, having dimensions \( a \times b \times c = 30 \times 40 \times 12 \) cm, with a rectangular aperture with dimensions \( w \times h = 15 \times 3 \) cm on its front side. Results are obtained by means of a Method of Moments (MoM) formulation [5] from 100 MHz to 3 GHz. The enclosure is illuminated with a y-directed uniform plane wave with the electric field linearly polarized along the shortest side of the aperture (i.e., the z-axis) and the observation point is in the center of the enclosure \( (E_z = 1 \text{ V/m}) \). In this frequency range, the enclosure exhibits 81 resonant frequencies as shown in Fig. 1 (b), with vertical grey lines (only the first 27 up to 2 GHz are reported in Table 1). It is evident that resonances and anti-resonances appear in the frequency spectrum of the \( \text{SE}_E \), depending on the position of the observation point inside the cavity and on the resonant modes that are excited by the source field. Furthermore, Fig. 1 (c) shows the pattern of the first mode, while Fig. 1 (d) shows the maps of the magnitude of the electric field on the \( xy \)-plane and \( yz \)-plane passing for the center of the enclosure. As it is evident, the magnitude of the electric field at the first resonance is maximum at the center of the shielded box where the \( \text{SE}_E \) is evaluated, so that the shielding effectiveness presents there’s a minimum.

In order to improve the \( \text{SE}_E \), especially in the neighborhood of the resonant frequencies, the interior cabinet surfaces are lined with an artificial absorbing material to damp the excited resonant fields, as shown in Fig. 1 (a).

The design procedure proceeds in two steps: first, the layers characteristics are determined with reference to a set of plane wave sources having different incidence angles in the assigned frequency bandwidth. Then, the multilayer effectiveness is tested inside the enclosure in order to ascertain whether the initial goals have been met with the absorber in place. An iteration with more stringent constraints may be necessary or not, depending on the configuration and on the experience in fixing the initial target values.
Fig. 1. (a) Shielded enclosure, with an aperture on one side, loaded with absorbing material in order to damp its resonances; (b) SE $E$ versus frequency at the center of the shielded enclosure illuminated by an $y$-directed impinging plane wave: $a=30$ cm, $b=12$ cm, $c=40$ cm, rectangular aperture with dimensions $w=15$ cm, $h=3$ cm, centered in the $xz$-side; (c) electric field map at the first resonant frequency and (d) electric field patterns at the first resonant frequency in the center of $xy$-plane and $yz$-plane.

Table 1: Natural resonant frequencies up to 2 GHz of the $30 \times 12$ cm enclosure, along with the mode indices $l, m$ and $n$

<table>
<thead>
<tr>
<th>$l$</th>
<th>$m$</th>
<th>$n$</th>
<th>$f$ [GHz]</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>1</td>
<td>0</td>
<td>0.624</td>
</tr>
<tr>
<td>2</td>
<td>2</td>
<td>0</td>
<td>0.900</td>
</tr>
<tr>
<td>2</td>
<td>1</td>
<td>0</td>
<td>1.067</td>
</tr>
<tr>
<td>1</td>
<td>3</td>
<td>0</td>
<td>1.230</td>
</tr>
<tr>
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<td>0</td>
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<td>0</td>
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<td>1</td>
<td>1.304</td>
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</tr>
<tr>
<td>1</td>
<td>4</td>
<td>0</td>
<td>1.580</td>
</tr>
</tbody>
</table>

III. EQUIVALENT CIRCUIT REPRESENTATION

The proposed absorbing material consists of four dielectric layers alternated with three lossy sheets, as depicted in Fig. 2. The outer sheet is a capacitive array of patches, the middle sheet is a homogeneous resistive sheet and the inner sheet is an inductive strip-grid. Furthermore, the dielectric layer adjacent to the PEC surface is a wire-medium, realized by means of orthogonal wires in order to achieve a better polarization insensitivity.

By proper selection of the geometrical and physical parameters of conductive sheets and dielectric layers, the result consisting in minimum EM field reflection over a prescribed frequency range and for oblique incidence angles is achieved. Particular effort has been devoted to make the absorber’s characteristics as independent as possible of the incident angle [18]. To perform the design of the artificial material, an equivalent circuit representation is established by means of
equivalent surface impedances and quasi-dynamic formulas for effective parameters obtained through homogenization procedures [19]. The equivalent circuit representation is more suitable for the optimal tuning of the constitutive parameters.

The propagation of uniform plane-waves in homogeneous stratified media can be modeled by means of the analogy with voltage and current waves on uniform equivalent Transmission Lines (TLs) [20], [21]; any arbitrary incident plane wave is decomposed into its fundamental TE and TM polarized waves, as in Fig. 2. The propagation constants $k_{z,i}$ and the intrinsic characteristic impedance $Z_{i,i}$ of the $i$-th medium at angular frequency $\omega$, are functions of both polarization and angle of incidence $\theta_{inc}$ as:

$$k_{z,i} = k_0 \sqrt{\varepsilon_{r,i} - \sin^2 \theta_{inc}}, \quad (1a)$$

$$Z_{i,i}^{TE} = \frac{\omega \mu_0}{k_{z,i}}, \quad (1b)$$

$$Z_{i,i}^{TM} = \frac{k_{z,i}}{\omega \varepsilon_0 \varepsilon_{r,i}}, \quad (1c)$$

where $j = \sqrt{-1}$, $k_0$ is the wavenumber in free space, $\varepsilon_0$ and $\mu_0$ are, respectively, the absolute permittivity and permeability of vacuum and $\varepsilon_{r,i}$ is the relative electric permittivity of $i$-th medium. The free space surrounding the structure is modeled by a port with interior impedance $Z_b$, according to (1b)-(1c), where $\varepsilon_{r,i}$ is set equal to 1. Thus, the equivalent circuit reported in Fig. 3 is obtained.

![Fig. 3. Equivalent circuit model of the overall system; an equivalent EM field source (frequency-, polarization- and angle-of-incidence-dependent) incident on a multi-layer absorber.](image)

To account for the three sheets located at the interfaces between dielectrics, equivalent lumped impedances are derived and inserted transversally in the equivalent circuit. An array of patches (outer interface #3), at frequencies well below the array resonances, exhibits an equivalent grid capacitance $C_g$ [22]:

$$C_g^{TM} = \frac{2\alpha}{\eta_{eff} \omega}, \quad (2a)$$

$$C_g^{TE} = \frac{2\alpha}{\eta_{eff} \omega} \left(1 - \frac{\sin^2 \theta_{inc}}{2\eta_{eff}}\right), \quad (2b)$$

where $\alpha$ is the grid parameter:

$$\alpha = k_0 D \ln \left(\frac{1}{\sin \left(\frac{\pi w}{2D}\right)}\right), \quad (3)$$

and the effective (complex) permittivity is
function of the two adjacent dielectric permittivity’s:

$$\varepsilon_{r,\text{eff}} = \varepsilon_0 \left[ \frac{\varepsilon_{r,j} + \varepsilon_{r,j+1}}{2} \right].$$

(4)

In (3), $w$ is the width of the gap between adjacent patches, $D$ is the period of the structure (as shown in Fig. 2), $\varepsilon_{r,\text{eff}}$ is the effective intrinsic wave impedance, $\varepsilon_{r,\text{eff}} = \frac{\mu_0}{\varepsilon_0 k_{r,\text{eff}}}$ and $k_{r,\text{eff}}$ is the effective wavenumber, $k_{r,\text{eff}} = k_e \sqrt{\varepsilon_{r,\text{eff}}}$. If the sheet is lossy, an equivalent resistance $R_g$ should be inserted in series with the capacitance. According to [23], the resistance can be computed as:

$$R_g = \frac{R_s}{D - w \varepsilon_{r,\text{eff}}},$$

(5)

where $R_s$ is the surface resistance of the sheet (measured in $\Omega$ sq).

For a grid of strips (inner interface #1), the grid impedance is inductive. Through the Babinet’s principle, the inductance can be derived straightforwardly from (2) as [22]:

$$L_{g}^{TM} = \frac{\eta_{r,\text{eff}} \alpha}{2\omega} \left( 1 - \frac{\sin^2 \theta_{\text{inc}}}{2\mu_0 \varepsilon_{r,\text{eff}}} \right),$$

(6a)

$$L_{g}^{TE} = \frac{\eta_{r,\text{eff}} \alpha}{2\omega},$$

(6b)

where (3)-(4) still hold, considering $w$ as the strip width (see Fig. 2). If the sheet is lossy, an equivalent resistance $R_g$ should be inserted in series with the inductance. According to [23], the resistance is:

$$R_g^{TM} = \frac{R_g^{TM}}{w \mu_{r,\text{eff}}},$$

(7)

The homogeneous resistive sheet located at the interface #2, is represented by its surface resistance $R_s$ as shown in Fig. 3.

The layers between sheets are assumed to be lossy dielectric with relative dielectric constant $\varepsilon_{r,j} = \varepsilon_{r,j}' - j\varepsilon_{r,j}'' = \varepsilon_{r,j}' (1 - j\tan \delta_i)$, where $\tan \delta_i$ is the loss tangent of the i-th dielectric medium. The inner dielectric adherent to the PEC surface of the enclosure is an artificial lossy wire-medium, as shown in Fig. 2. In the long-wavelength limit, such a structure behaves like an homogeneous material whose effective relative permittivity is a frequency-dependent scalar quantity. In the case of E-polarized incident plane wave and lossy wires with finite conductivity $\sigma$, the dielectric constant reads [24]:

$$\varepsilon_{r,j} = \varepsilon_{r,j}' - j\varepsilon_{r,j}'' = \varepsilon_{r,j}' + \frac{c_0 \varepsilon_0}{j\omega d^2 (j\omega L + \varepsilon_{r,j})},$$

(8)

where $c_0$ is the free speed of light, $\varepsilon_0$ is the dielectric constant in free space, $d$ is the period of the wire grid and $\varepsilon_{r,j}$ is the relative permittivity of the host medium. In (8), the per unit length external inductance of wires $L$ is given by:

$$L = \frac{\mu_0 L_{\text{wire}}}{2\pi} \ln \left( \frac{d^2}{4\rho_0 (d - \rho_0)} \right),$$

(9)

being $\rho_0$ the wire radius. The per unit length impedance $\zeta_s$, accounting for wire losses is:

$$\zeta_s = \frac{1 + j}{2\pi \rho_0 \sqrt{\varepsilon_0} \sigma} I_1(\xi)$$

(10)

In (10), $I_0(\cdot)$ and $I_1(\cdot)$ are, respectively, the modified Bessel functions of the first kind of order 0 and 1, and:

$$\xi = (1 + j) \sqrt{\varepsilon_0 \mu_0 \sigma \rho_0}.$$  

(11)

In order to obtain a material effective under both TM and TE polarizations, two sets of wires mutually perpendicular have been introduced in order to achieve polarization insensitivity.

The actual reflection coefficient $\Gamma$ of the multilayered material at air/absorber interface #4, must be computed by using the well-known transmission line theory. Starting from the inner layer #1 and moving toward the free space, the input impedance $Z_i$ at the beginning of the i-th line (assuming the line terminated at $l = h_i$ on the impedance $Z_{i-1}$, with $Z_0 = 0$ since the first line is terminated on the PEC surface of the enclosure) is evaluated:

$$Z_i = Z_{c,j} + Z_{c,j+1} \tanh(jk_{z,i}h_j)$$

(12)

Then the impedance value $Z_i$ is updated by performing the parallel of the computed $Z_i$ with the lumped sheet admittance $Y_{\text{sheet}}^i$ (we assume $Y_{\text{sheet}}^4 = 0$ since there is no sheet at the air/absorber interface), according to:

$$Z_i = \left( \frac{1}{Z_i} + Y_{\text{sheet}}^i \right)^{-1}. $$

(13)
Finally, once the $Z_4$ is available, the reflection coefficient (to be minimized) of the absorbing material can be computed as:

$$
\Gamma = \frac{Z_4 - Z_m}{Z_4 + Z_m}.
$$

(14)

It should be noted that $\Gamma$ has to be minimized in a prescribed frequency range and for various incidence angles and polarizations.

IV. OPTIMIZATION

To optimize all the geometrical and constitutive parameters of the absorbing material in order to reduce its reflection coefficient $\Gamma$ in the prescribed frequency range, the Particle Swarm Optimization (PSO) has been used. Several metaheuristic evolutionary algorithms are available today, such as the Genetic Algorithm (GA), the Ant Colony Optimization (ACO), the Bees Algorithms and the PSO, and they have been applied successfully in the design of absorbing materials [25]-[28]. Since its proposal by Kennedy and Eberhart in 1995 [29], the PSO has rapidly become very popular as an efficient optimization method for solving single objective and multi-objective optimization problems. In addition, a large number of works dealing with the application of the PSO technique to engineering problems, is available in literature [30]-[34] and the authors have developed a good expertise with this method in several past works [7], [34]-[36]. For these reasons, the PSO have been selected for the optimization process.

In the present work, a simple variation of the original PSO, referred to as Meta Particle Swarm Optimization (MPSO) [27] is adopted. It simply consists in subdividing the entire swarm in more subgroups of particles moving through the space domain $\mathcal{D}$. Each j-th group is composed of $M_j$ particles flying with a velocity vector $v_{i}^{m,j} = [v_{1,j}^{m,j}, v_{2,j}^{m,j}, ..., v_{N,j}^{m,j}]^T$, at time $t$ (with $m = 1, 2, ..., M_j$), around a multidimensional search space $\mathcal{D}$. During its flight, each particle updates its position $x_{i}^{m,j} = [x_{1,j}^{m,j}, x_{2,j}^{m,j}, ..., x_{N,j}^{m,j}]^T$ according to its own experience, to that of the group, which the particle belongs to and to the experience of the entire warm. The MPSO method combines three search levels:

1. A local single-particle search: the $m$-th particle knows its personal best position $b_{L}^{m,j}$ (local optimum solution); i.e., the coordinates associated with the best solution that the particle has achieved so far.
2. A group local search: the $m$-th particle exchanges information with other particles of the same group and knows the $b_{S}^{j}$ value (global optimum solution inside the $j$-th group); i.e., the coordinates associated with the best position even tracked by the group giving the best fitness value in the group population.
3. A swarm global search: the $m$-th particle exchanges information with all other particles and knows the best global value $b_{G}$ (global optimum solution); i.e., the coordinates associated with the best position even tracked by the swarm giving the best fitness value in the entire population.

At each algorithm step, $b_{L}^{m,j}$, $b_{S}^{j}$ and $b_{G}$ are computed, updated and used by the particles to adjust their velocities and positions in order to improve their current fitness through the following two updating equations:

$$
v_{i+1}^{m,j} = wv_{i}^{m,j} + c_1\varphi_1(b_{L}^{m,j} - x_{i}^{m,j}) + c_2\varphi_2(b_{S}^{j} - x_{i}^{m,j}) + c_3\varphi_3(b_{G} - x_{i}^{m,j}),
$$

$$
x_{i+1}^{m,j} = x_{i}^{m,j} + v_{i+1}^{m,j}.
$$

(15a)

(15b)

The entire optimization relies on the correct manipulation of the particles’ velocities; $w$ is the inertia factor, which keeps the particle in its current trajectory. The last three terms inject deviation according to the distances from $b_{L}^{m,j}$, $b_{S}^{j}$ and $b_{G}$ best locations through the cognitive factor $c_1$, the group social factor of the $j$-th group $c_2$ and the global social factor $c_3$. The numbers $\varphi_1$, $\varphi_2$ and $\varphi_3$ are random variables distributed in the range $[0,1]$, which inject the unpredictability of the particles’ movement. The convergence of the algorithm depends on the proper tuning of the acceleration coefficients and on the boundary conditions used to prevent the explosion of the particles [37].

It should be noted that all the groups interact among them in the optimization process. In the
early time of the iteration process, the global best particle $b_0$ appears in every subgroup alternately, which shows the well global exploration performance that tends to concentrate in the territory around the group best particle $b_j^c$ in the medium time of the process. Then, when the algorithm enters the latter time of the process, $b_0$ is almost fixed in some subgroup and the algorithm begins with a local research around the best position tracked by the swarm. Several papers [38], [39] have introduced additional factors in the velocity equation; named repulsive factors in order to encourage individual particles, located in the territory of other groups to escape from the other groups’ territory in efficient manners and consequently search for multiple optima in the solution space.

In the problem at hand, each particle $x_{j,t}$ consists of a possible set of values for the constitutive/geometrical parameters of the absorbing material. The inertia and acceleration coefficients have been chosen according to [37] and the reflecting boundary conditions have been used to relocate the particles that fly outside the allowed solution space. The cost function $F$ that needs to be minimized is defined as:

$$F = \frac{1}{2N_{freq}N_{inc}} \sum_{l=TE,TM} \sum_{n=1}^{N_{inc}} \left| \Gamma_l(\omega_m, \theta_{inc,n}) - \Gamma_0 \right|^2,$$

where $N_{freq}$ is the number of sampling frequencies $\omega_m$ distributed over the band of interest, $N_{inc}$ is the number of angles of incidence $\theta_{inc,n}$, $\Gamma_0$ is the target value to be achieved and $\Gamma_l(\omega_m, \theta_{inc,n})$ is the computed actual value of the reflection coefficient.

V. RESULTS

The methodology has been applied for the design of two absorbers in the frequency range from 100 MHz up to 3 GHz. The first one is constrained to possess best performance below 1 GHz because it is aimed at absorbing the first resonant modes. The second one is designed to absorb the higher modes of the enclosure occurring at frequencies above 1 GHz.

The following constraints have been enforced:
1. The periods $D_1$ and $D_3$ of both the selective surfaces (patch array and strip-grid) have been considered equal (through several simulations we have found that the optimum solution is near the condition $D_2=D_4$, so that this constrain has been directly enforced in the optimization process in order to obtain a simpler absorber).
2. The host dielectric of the wire-medium (medium #1) has been considered to be foam, $\varepsilon_{rel}=1$.
3. The loss tangent of all the remaining dielectrics has been set equal to $1.25 \times 10^{-4}$.
4. The two frequency selective surfaces have been considered lossless.

The geometrical variables and physical parameters that have been optimized are reported in Table 2 with their “optimum” values.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Absorber #1</th>
<th>Absorber #2</th>
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</thead>
<tbody>
<tr>
<td>$\varepsilon_r$</td>
<td>9.9</td>
<td>4.5</td>
</tr>
<tr>
<td>$\varepsilon_i$</td>
<td>2.6</td>
<td>1.2</td>
</tr>
<tr>
<td>$\varepsilon_{rel}$</td>
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<td>2.9</td>
</tr>
<tr>
<td>$h_1$</td>
<td>4.8 mm</td>
<td>1.6 mm</td>
</tr>
<tr>
<td>$h_2$</td>
<td>5.5 mm</td>
<td>1.6 mm</td>
</tr>
<tr>
<td>$h_3$</td>
<td>5.8 mm</td>
<td>1.7 mm</td>
</tr>
<tr>
<td>$h_4$</td>
<td>6 mm</td>
<td>2 mm</td>
</tr>
<tr>
<td>$D_1=D_3$</td>
<td>22.8 mm</td>
<td>18.4 mm</td>
</tr>
<tr>
<td>$w_1$</td>
<td>0.8 mm</td>
<td>0.2 mm</td>
</tr>
<tr>
<td>$w_3$</td>
<td>2.2 mm</td>
<td>1.8 mm</td>
</tr>
<tr>
<td>$R_{s2}$</td>
<td>81.5 $\Omega$/sq</td>
<td>243.7 $\Omega$/sq</td>
</tr>
<tr>
<td>$d$</td>
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<td>0.3 mm</td>
</tr>
<tr>
<td>$r_0$</td>
<td>10 $\mu$m</td>
<td>5 $\mu$m</td>
</tr>
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</table>

The frequency spectrum of the reflection coefficients $\Gamma$ for both TMz and TEz polarizations are reported in Figs. 4, for different angles of incidence $\theta_{inc}$ for either of the absorbers. In all the Figs., it also reported the frequency trend of the prescribed reflectivity $\Gamma_0$, that has been assumed equal to a band-pass Chebyshev filter of second type of order 4, for the design of the absorber #1 and 5, for the design of the absorber #2, with stop-band ripple equal to 10 dB and stop-band-edge
frequencies equal to 100 MHz and 3 GHz in either cases. The absorbing performance are not substantially degraded, increasing the incidence angle $\theta_{inc}$. It is possible to note that under TM$_z$ polarization, the absorbing performance of the two absorbers are less stable with respect to the angle of incidence; nevertheless, they show higher absorbing coefficients for all the angles of incidence.

Fig. 4. Reflection coefficients $\Gamma$ of the two absorbers versus frequency $f$ for TM$_z$ and TE$_z$ polarizations for different angles of incidence $\theta_{inc}$.

For the sake of completeness, the best value $b_G$ and the mean value of the swarm (that has been subdivided in 4 tribes) during the search are reported in Figs. 5, for both the absorbers. It is evident that after a very steep descent in the first steps, the tribes move slowly but quite constantly toward the best solution.

Finally, the shielding effectiveness $SE_E$ of the commercial enclosure studied in Figs. 1, having dimensions $a \times b \times c = 30 \times 40 \times 12$ cm, with a rectangular aperture with dimensions $w \times h = 15 \times 3$ cm on its front side, has been newly computed twice under a perpendicularly impinging plane wave ($E = Ez$ and $k = ky$); a first time with absorber #1 placed on its interior walls, then with absorber #2. The results have been obtained by means of the Commercial Software CST Microwave Studio based on the Finite Integration Technique in the Time Domain. The results are reported in Figs. 6. It is possible to note that the
Absorbers are effective in damping the effect of the first resonant frequencies of the shielded enclosure and their performance are reasonable compared to their geometrical dimensions and the values of their constitutive parameters. The attenuation shown by both the absorbers when placed inside the test enclosure are substantially in accordance with the theoretical predictions under plane wave incidence, with variable angle of incidence $\theta_{inc}$.

Finally, Figs. 7 show the pattern of the electric field at the first resonant frequency of the enclosure $f = 0.624$ GHz, without (Fig. 7 (a)) and with (Fig. 7 (b)) the absorber #1 on the interior sides of the walls of the enclosure.

The absorber is effective in reducing the magnitude of the resonant field mode, especially in the central zone of the enclosure.

Fig. 5. Best value and mean value of the swarm during the optimization process.

Fig. 6. Shielding effectiveness of the enclosure reported in Figs. 1, with and without the designed absorbers placed on its interior walls.
VI. CONCLUSION

In this work, an optimal design procedure is presented, aimed at achieving compact and lightweight absorbing structures for damping resonances inside shielded enclosure. The proposed absorber is based on the concepts of Jaumann’s layers and circuit analog absorbers; it employs four layers of lossy (artificial) dielectrics, separated by two high/low impedance frequency selective surfaces and one resistive sheet. The parameters of the absorber have been optimized by means of a PSO, in order to maximize its performance in the frequency range between 0.1 GHz and 3 GHz, for various incidence angles. The results show that an absorber with the expected performance can be obtained, demonstrating the effectiveness of the proposed design procedure and absorber configuration.

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ARANEO, CELOZZI: OPTIMAL DESIGN OF ELECTROMAGNETIC ABSORBERS

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Bandwidth Improvement of Omni-Directional Monopole Antenna with a Modified Ground Plane

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Abstract — This study introduces a new design of low profile, multi-resonance and omni-directional monopole antenna for Ultra-Wideband (UWB) applications. The proposed antenna configuration consists of an ordinary square radiating patch and a ground plane with pairs of inverted fork-shaped slits and inverted Γ-shaped parasitic structures, which provides a wide usable fractional bandwidth of more than 135%. By cutting a pair of inverted fork-shaped slits in the ground plane and also by inserting a pair of inverted Γ-shaped conductor-backed plane in the feed gap distance, additional resonances are excited and hence much wider impedance bandwidth can be produced; especially at the higher band. By obtaining the third and fourth resonances, the usable lower frequency is decreased from 3.12 GHz to 2.83 GHz and also the usable upper frequency of the presented monopole antenna is extended from 10.3 GHz to 14.87 GHz. The proposed antenna has symmetrical structure with an ordinary square radiating patch; therefore, displays a good omni-directional radiation patterns, even at the higher frequencies. The antenna radiation efficiency is greater than 87% across the entire radiating band. The measured results show that the proposed antenna can achieve the Voltage Standing Wave Ratio (VSWR) requirement of less than 2.0 GHz in frequency range from 2.83 GHz to 14.87 GHz, which is suitable for UWB systems.

Index Terms — Omni-directional radiating patterns and printed monopole antenna.

I. INTRODUCTION

After allocation of the frequency band from 3.1 GHz to 10.6 GHz for the commercial use of Ultra-Wideband (UWB) systems by the Federal Communication Commission (FCC) [1], ultra-wideband systems have received phenomenal gravitation in wireless communication. Designing an antenna to operate in the UWB frequency range is quite a challenge, because it has to satisfy the requirements such as ultra-wide impedance bandwidth, omni-directional radiation pattern, constant gain, high radiation efficiency, constant group delay, low profile, easy manufacturing, etc. [2]. In UWB communication systems, one of key issues is the design of a compact antenna while providing wideband characteristic over the whole operating band. Consequently, a number of microstrip antennas with different geometries have been experimentally characterized [3-4]. Some methods are used to obtain the multi-resonance function in the literature [5-8].

In this paper, a different method is proposed to obtain the very wideband bandwidth for the compact monopole antenna. In the proposed antenna, we use pairs of inverted fork-shaped slits and Γ-shaped conductor-backed plane in the ground plane, which provides a wide usable fractional bandwidth of more than 135%. Regarding Defected Ground Structures (DGS) theory, the creating slits in the ground plane provide additional current paths. Moreover, these structures change the inductance and capacitance of the input impedance, which in turn leads to change the bandwidth [9-11]. Therefore, by cutting a pair of inverted fork-shaped slits in the ground plane, much enhanced impedance bandwidth may be achieved. In addition, based on Electromagnetic Coupling Theory (ECT), by adding a pair of inverted Γ-shaped conductor-
backed plane in the air gap distance, additional coupling is introduced between the bottom edge of the square patch and the ground plane and its impedance bandwidth is improved without any cost of size or expense. Good VSWR and radiation pattern characteristics are obtained in the frequency band of interest. The designed antenna has a small size of 12×18 mm².

II. MICROSTRIP ANTENNA DESIGN

The presented small monopole antenna fed by a microstrip line is shown in Fig. 1, which is printed on an FR4 substrate of thickness of 1.6 mm, permittivity of 4.4 and loss tangent 0.018.

![Fig. 1. Geometry of proposed omni-directional monopole antenna: (a) side view and (b) modified ground plane.](image)

The basic monopole antenna structure consists of a square radiating patch, a feed line and a ground plane. The square radiating patch has a width \( W \). The patch is connected to a feed line of width \( W_f \) and length \( L_f \). The width of the microstrip feed line is fixed at 2 mm, as shown in Fig. 1. On the other side of the substrate, a conducting ground plane with two inverted fork-shaped slits and a pair of \( \Gamma \)-shaped parasitic structures is placed. The proposed antenna is connected to a 50-Ω SMA connector for signal transmission.

The DGS applied to a ground plane causes a resonant character of the structure transmission with a resonant frequency controllable by changing the shape and size of the slits [3]. In addition, based on ECT, by using a parasitic structure in air gap distance, an additional coupling is introduced between the bottom edge of the square patch and the ground plane and its impedance bandwidth is improved without any cost of size or expense. Therefore, by cutting two inverted fork-shaped slits and also by embedding a pair of inverted \( \Gamma \)-shaped parasitic structures and carefully adjusting these parameters, much enhanced impedance bandwidth may be achieved. The final values of proposed design parameters are displayed in Table 1.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
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<tbody>
<tr>
<td>( W_{sub} )</td>
<td>12 mm</td>
</tr>
<tr>
<td>( L_{sub} )</td>
<td>18 mm</td>
</tr>
<tr>
<td>( h_{sub} )</td>
<td>1.6 mm</td>
</tr>
<tr>
<td>( W_f )</td>
<td>2 mm</td>
</tr>
<tr>
<td>( L_f )</td>
<td>7 mm</td>
</tr>
<tr>
<td>( W_s )</td>
<td>0.5 mm</td>
</tr>
<tr>
<td>( L_s )</td>
<td>0.5 mm</td>
</tr>
<tr>
<td>( W_{s1} )</td>
<td>2.5 mm</td>
</tr>
<tr>
<td>( W_{s2} )</td>
<td>1 mm</td>
</tr>
<tr>
<td>( L_{s1} )</td>
<td>2.5 mm</td>
</tr>
<tr>
<td>( W_{s3} )</td>
<td>0.5 mm</td>
</tr>
<tr>
<td>( W_p )</td>
<td>2.25 mm</td>
</tr>
<tr>
<td>( L_{p1} )</td>
<td>125 mm</td>
</tr>
<tr>
<td>( L_{p2} )</td>
<td>3 mm</td>
</tr>
<tr>
<td>( W_{p1} )</td>
<td>0.5 mm</td>
</tr>
<tr>
<td>( L_{p3} )</td>
<td>0.75 mm</td>
</tr>
<tr>
<td>( L_{gnd} )</td>
<td>3.5 mm</td>
</tr>
</tbody>
</table>

III. RESULTS AND DISCUSSIONS

The proposed microstrip monopole antenna with various design parameters was constructed and the numerical and experimental results of input impedance and radiation characteristics are presented and discussed. Ansoft HFSS simulations are used to optimize the design and agreement between the simulation and measurement is obtained [11].

Figure 2 shows the structure of the various antennas used for simulation studies. VSWR characteristics for ordinary square patch antenna (Fig. 2 (a)), square antenna with a pair of inverted fork-shaped slits (Fig. 2 (b)) and the proposed antenna (Fig. 2 (c)) are compared in Fig. 3. As shown in Fig. 3, in the proposed antenna configuration, the ordinary square monopole can provide the fundamental and next higher resonant
radiation band at 4.8 GHz and 8.2 GHz, respectively; in the absence of the inverted fork-shaped slits and a pair of inverted \( \Gamma \)-shaped conductor-backed plane. It is observed that by using these modified elements including pairs of inverted fork-shaped slits and \( \Gamma \)-shaped conductor-backed plane, additional third (10.9 GHz) and fourth (14.2 GHz) resonances are excited, respectively, and hence the bandwidth is increased. As illustrated in Fig. 3, the embedded structures in the ground plane and also the generation of extra resonances have an effect of disappear for the second resonance of the antenna. This is because the coupling between the ground plane and radiating patch for the second resonance becomes weaker.

Fig. 2. (a) Ordinary square monopole antenna, (b) the antenna with a pair of inverted fork-shaped slits in the ground plane and (c) the proposed monopole antenna.

In order to understand the phenomenon behind these additional resonances performance, the simulated current distributions on the ground plane for the proposed antenna at 10.9 GHz and 14.5 GHz (third and fourth resonances) are presented in Figs. 4 (a) and (b), respectively. As shown in Fig. 4 (a), the currents concentrated on the edges of the interior and exterior of the inverted fork-shaped slits at third resonance frequency (10.9 GHz). Also, as illustrated in Fig. 4 (b), the current concentrated on the edges of the interior and exterior of the inverted \( \Gamma \)-shaped parasitic structures at fourth resonance frequency (14.2 GHz).

Fig. 3. Simulated VSWR characteristics for the various monopole antennas shown in Fig. 2.

Fig. 4. Simulated surface current distributions on the ground plane for the proposed antenna: (a) at 10.9 GHz and (b) at 14.2 GHz.

Fig. 5. Photograph of the realized printed monopole antenna: (a) top view and (b) bottom view.
Fig. 6. Measured and simulated radiation patterns of the proposed antenna: (a) 5 GHz, (b) 8 GHz, (c) 11 GHz and (d) 14 GHz.

As shown in Fig. 5, the proposed antenna was fabricated and tested. The VSWR characteristic of the antenna was measured using a network analyzer in an anechoic chamber. The radiation patterns have been measured inside an anechoic chamber using a double-ridged horn antenna as a reference antenna placed at a distance of 2 m. Also, a two-antenna technique using a spectrum analyzer and a double-ridged horn antenna as a reference antenna placed at a distance of 2 m is used to measure the radiation gain in the z axis direction (x-z plane).

Figure 6 depicts the measured and simulated radiation patterns of the proposed antenna, including the co-polarization and cross-polarization in the H-plane (x-z plane) and E-plane (y-z plane). It can be seen that quasi-omnidirectional radiation pattern can be observed on x-z plane over the whole UWB frequency range, especially at the low frequencies. The radiation pattern on the y-z plane displays a typical figure-of-eight, similar to that of a conventional dipole antenna. It should be noticed that the radiation patterns in E-plane become imbalanced as frequency increases because of the increasing effects of the cross-polarization. The patterns indicate at higher frequencies and more ripples can be observed in both E and H-planes, owing to the generation of higher-order modes [14-16].

Figure 7 shows the measured and simulated VSWR characteristics of the proposed antenna. The fabricated antenna has the frequency band of 2.83 GHz to 14.87 GHz.

Fig. 7. Measured and simulated VSWR characteristics of the proposed monopole antenna.

The simulated radiation efficiency characteristic of the proposed antenna is shown in Fig. 8. Results of the calculations using the software HFSS indicated that the proposed antenna features a good efficiency, being greater than 87% across the entire radiating band. In addition, the simulated and measured maximum gains of the antenna against frequency are illustrated in Fig. 8.
The antenna gain has a flat property, which increases by the frequency. As seen, the proposed antenna has sufficient and acceptable gain levels in the operation bands [17-18].

In the UWB communication systems, antennas should be able to transmit the electrical pulse with minimal distortion. If group delay variation exceeds more than 1 ns, phases are no more linear in far field and phase distortion occurs, which can cause a serious problem for UWB applications. Figure 9 shows the simulated group delay property of the proposed monopole antenna. As illustrated, the variation is less than 0.25±0.4 over the frequency band from 3 GHz to 14.5 GHz. It shows that the antenna has low-impulse distortion and is suitable for UWB applications [18-21].

Table 2: Comparison of previous designs with the proposed antenna

<table>
<thead>
<tr>
<th>Ref.</th>
<th>FBW (%)</th>
<th>Dimension (mm)</th>
<th>Gain (dBi)</th>
</tr>
</thead>
<tbody>
<tr>
<td>[13]</td>
<td>47%</td>
<td>33×33</td>
<td>3.5~6</td>
</tr>
<tr>
<td>[14]</td>
<td>87%</td>
<td>22×24</td>
<td>1~5.5</td>
</tr>
<tr>
<td>[15]</td>
<td>87%</td>
<td>32×25</td>
<td>2~5.5</td>
</tr>
<tr>
<td>[16]</td>
<td>91%</td>
<td>26×26</td>
<td>3~7</td>
</tr>
<tr>
<td>[17]</td>
<td>112%</td>
<td>20×20</td>
<td>2~4.7</td>
</tr>
<tr>
<td>[18]</td>
<td>118%</td>
<td>40×10</td>
<td>2.3~6.3</td>
</tr>
<tr>
<td>[19]</td>
<td>130%</td>
<td>12×18</td>
<td>2.7~5.5</td>
</tr>
<tr>
<td>[20]</td>
<td>132%</td>
<td>25×26</td>
<td>not reported</td>
</tr>
<tr>
<td>This Work</td>
<td>136%</td>
<td>12×18</td>
<td>3.3~6.5</td>
</tr>
</tbody>
</table>

Table 2 summarizes the previous designs and the proposed antenna. As seen, the proposed antenna has a compact size with very wide bandwidth in comparison with the pervious works. In addition, the proposed antenna has good omni-directional radiation patterns with low cross-polarization level, even at the higher and upper frequencies. As the proposed antenna has symmetrical structure and an ordinary square radiating patch without any slot and parasitic structures at top layer, in comparison with previous multi-resonance UWB antennas, the proposed antenna displays a good omni-directional radiation pattern, even at lower and higher frequencies [20]. Also, the proposed microstrip-fed monopole antenna has sufficient and acceptable radiation efficiency, group delay and antenna gain levels in the operation bands [25-27].
IV. CONCLUSION
In this manuscript, a novel compact Printed Monopole Antenna (PMA) with multi-resonance characteristics has been proposed for UWB applications. The fabricated antenna can operate from 2.83 GHz to 14.87 GHz. In order to enhance the bandwidth, we insert a pair of inverted fork-shaped slits in the ground plane and also by adding two inverted Γ-shaped conductor-backed plane with variable dimensions, additional resonances are excited and hence much wider impedance bandwidth can be produced. The designed antenna has a simple configuration with small size of 12×18 mm² and an ordinary square radiating patch, which its radiation efficiency is greater than 87%. Simulated and experimental results show that the proposed antenna could be a good candidate for UWB systems application.

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Compact Microstrip Lowpass Filter with Ultra-Wide Stopband using Stepped-Impedance Trapezoid Resonators

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Abstract — A new microstrip lowpass filter with compact size and ultra-wide stopband is presented. The resonance properties of a microstrip main transmission line parallel loaded with the stepped-impedance trapezoid resonator are studied. Analysis results reveal that a compact size and ultra-wide stopband lowpass filter can be realized by properly introducing multiple stepped-impedance trapezoid resonators in the design. A demonstration filter with 3 dB cutoff frequency at 0.8 GHz has been designed, fabricated, and measured. Results indicate that the proposed filter is able to suppress the 17th harmonic response by 15 dB, together with a small size of 0.057 λg × 0.077 λg, where λg is the guided wavelength at 0.8 GHz.

Index Terms — Compact, lowpass filter, microstrip, trapezoid resonator, and ultra-wide.

I. INTRODUCTION

Planar lowpass filters with compact size and high performance are in great demand for modern wireless communication systems to suppress harmonics and spurious signals. Conventional lowpass filters using shunt stubs or high-low impedance transmission lines have been widely used in microwave systems for their excellent characteristics [1]. However, compact size and high performance are hard to achieve simultaneously. Therefore, many works report attempts to achieve both size reduction and performance enhancement [2–11].

In general, there are two methods to design a lowpass filter with compact size and wide stopband. The first method is to form a lowpass filter by cascading multiple resonators [2-5]. With this method, Li et al. designed a lowpass filter by cascading multi-radial patch resonators [2]. Although, sharp roll-off was achieved, the size of the filter was relatively large and only the 6th harmonic response was suppressed. A microstrip lowpass filter with low insertion loss and sharp roll-off was proposed by cascading modified semi-circular and semi-elliptical microstrip patch resonators [3]. However, the circuit size and passband performance still need improvement. Therefore, to further improve the stopband performance, Ma et al. proposed a lowpass filter by cascading LC resonant structures and transformed radial stubs. Although, better than 13th harmonic suppression was realized, this method increased design complexity and circuit area [4]. As for the work reported in [5], a microstrip lowpass filter with compact size and ultra-wide stopband had been achieved but at the cost of a relatively complex circuit design. The second method is to design a lowpass filter by using modified stepped impedance hairpin resonators [6-8]. Using stepped impedance hairpin resonators with radial stubs, Wei et al., proposed a lowpass filter with 7th harmonic suppression performance [6]. Although, a compact design had been realized with this method, further improvement should be carried out in stopband bandwidth. The stopband performance should also be improved in [7], as the compact lowpass filter using a coupled-line hairpin unit, one spiral slot, and two open stubs only achieve 10 dB attenuation up to 20 GHz inside the stopband. A very wide stopband lowpass filter is achieved with a novel application of shunt open-stubs at the feed points of a center fed coupled-line hairpin resonator [8], but the
reflection loss is relatively large. In addition, using defected ground structure is also a popular and useful way [9-11]. A lowpass filter composed of semi-circular defected ground structures and semi-circular stepped-impedance shunt stubs is proposed in [9], which increases the circuit complexity and circuit size.

The motivation of this paper is to design a new microstrip lowpass filter with both compact size and ultra-wide stopband. To achieve compact size and ultra-wide stopband rejection, stepped-impedance trapezoid resonators are introduced and parallel loaded on the main transmission line of the filter. A demonstration filter with 3 dB cutoff frequency at 0.8 GHz has been designed, fabricated, and measured. Measured results indicate that the designed filter has an ultra-wide stopband with better than 15 dB suppression up to 13.7 GHz. Furthermore, the size of the filter is only $11.6 \times 15.7$ mm$^2$, which corresponds to a compact electrical size of $0.057 \lambda_g \times 0.077 \lambda_g$, where $\lambda_g$ is the guided wavelength at 0.8 GHz.

**II. FILTER DESIGN**

Figure 1 shows the layout of the proposed lowpass filter, which is composed of a high impedance microstrip main transmission line and five stepped-impedance trapezoid resonators. Each stepped-impedance trapezoid resonator is composed of a high impedance transmission line and a trapezoidal patch, which are connected in series. Figure 2 shows the lumped-element equivalent circuit of the presented lowpass filter. In the circuit, the high impedance line is mainly represented by the inductance $L_0$ and $L_1$. The symbols $C_a$, $C_b$, and $C_c$ mainly represent the capacitances between the resonator 1, 2, 3, and the ground plane, respectively, while $L_a$, $L_b$, and $L_c$ mainly represent the inductances of the high impedance of the stepped-impedance resonator 1, 2, and 3. The symbol $C_{bc}$ means the coupling capacitance between resonators 1 and 3. The capacitor and inductor values of the lumped-element equivalent circuit of the proposed lowpass filter are given as follows: $C_a = 4.6$ pF, $C_b = 1.9$ pF, $C_c = 6.6$ pF, $C_{bc} = 0.1$ pF, $L_0 = 5.6$ nH, $L_1 = 15.8$ nH, $L_a = 0.9$ nH, $L_b = 0.92$ nH, and $L_c = 0.27$ nH.

To illustrate the design theory of the proposed filter, frequency responses of four stepped-impedance rectangular trapezoid resonators and one stepped-impedance isosceles trapezoid resonator have been studied, respectively. As can be seen from Fig. 3 (a), a microstrip main transmission line with only two stepped-impedance rectangular trapezoid resonators, i.e., resonator 1, exhibits a wide stopband together with one transmission pole (TP) at about 4.4 GHz. In order to suppress the undesired frequency response, another two stepped-impedance rectangular trapezoid resonators, i.e., resonator 2, are also introduced to the design.

![Fig. 1. Layout of proposed lowpass filter.](image1)

![Fig. 2. Lumped-element equivalent circuit of the proposed lowpass filter.](image2)

Figure 3 (b) investigates the resonant properties of the stepped-impedance trapezoid resonators. It can be seen that one transmission zero (TZ) at about 4.4 GHz in the stopband is achieved. This transmission zero is caused by the resonance of loaded stepped-impedance trapezoid resonators and its frequency location can be controlled by the structure parameters of the stepped-impedance trapezoid resonators. Based on the investigation mentioned above, if we can properly combine the four stepped-impedance rectangular trapezoid resonators in a filter, the mutual suppression of spurious passbands and thereby a better stopband performance is
expected to be achieved. Figure 3 (c) shows the frequency response of the filter with four stepped-impedance rectangular trapezoid resonators, i.e., resonators 1 and 2. As expected, by locating the transmission zero in Fig. 3 (b) around the position of spurious response appeared at about 4.4 GHz in Fig. 3 (a), we finally achieve the new lowpass filter with an enhanced stopband performance.

In order to achieve sharp roll-off rate, one stepped-impedance isosceles trapezoid resonator, i.e., resonator 3, is also introduced to the filter. It is can be seen in Fig. 3 (d) a sharp roll-off rate is achieved by the adoption of resonator 3. Therefore, if we properly combine the three types of stepped-impedance trapezoid resonators, i.e., resonators 1, 2, and 3, in one filter, a compact microstrip lowpass filter with ultra-wide stopband can be realized.

The new lowpass filter is designed and fabricated based on the analysis mentioned above. The structure parameters are as follows: \( l_1 = 2.2 \) mm, \( l_2 = 0.9 \) mm, \( l_3 = 4.5 \) mm, \( l_4 = 0.2 \) mm, \( l_5 = 0.2 \) mm, \( l_6 = 9.3 \) mm, \( w_1 = 2.6 \) mm, \( w_2 = 3.4 \) mm, \( w_3 = 1.16 \) mm, \( w_4 = 1.4 \) mm, \( w_5 = 3.4 \) mm, and \( w_6 = 9.3 \) mm. The substrate used here has a relative dielectric constant of 3.38 and a thickness of 0.508 mm. Figure 4 is the photograph of the proposed lowpass filter.

![Fig. 3. Simulated S-parameters of the studied resonators; a) filter with stepped-impedance rectangular trapezoid resonator 1, b) filter with stepped-impedance rectangular trapezoid resonator 2, c) filter with stepped-impedance rectangular trapezoid resonator 1 and 2, and d) filter with stepped-impedance isosceles trapezoid resonator 3.](image)

![Fig. 4. Photograph of the proposed lowpass filter.](image)

### III. SIMULATION AND MEASUREMENT RESULTS

Simulation was accomplished by using EM simulation software ANSOFT HFSS 12. The comparisons among the circuit model EM simulated results and the equivalent lumped element circuit results are given in Fig. 5.
Measurement was carried out on an Agilent 8722ES network analyser. Figure 6 shows the simulated and measured results, which are in good agreement. As can be observed from Fig. 6, the measured 3 dB cutoff frequency $f_c$ is located at 0.8 GHz, as expected. Figure 6 also shows that the spurious frequencies suppressed by better than 15 dB from 2.36 GHz up to 13.7 GHz. Thus, the proposed filter has a property of 17th harmonic suppression. Furthermore, the proposed filter exhibits a small electrical size of $0.057\lambda_g \times 0.077\lambda_g$, where $\lambda_g$ is the guided wavelength at 0.8 GHz. For comparison, Table I summarizes the performance of some published lowpass filters. As can be seen from the table, our proposed filter has the properties of compact size, simple circuit topology, and ultra-wide stopband among the quoted filters.

### Table I: Performance comparisons among published filters and proposed ones.

<table>
<thead>
<tr>
<th>Ref.</th>
<th>Harmonic suppression</th>
<th>Cutoff frequency (GHz)</th>
<th>Circuit size</th>
</tr>
</thead>
<tbody>
<tr>
<td>2</td>
<td>6th</td>
<td>2.4</td>
<td>$0.351\lambda_g \times 0.106\lambda_g$</td>
</tr>
<tr>
<td>3</td>
<td>6th</td>
<td>3.12</td>
<td>$0.395\lambda_g \times 0.151\lambda_g$</td>
</tr>
<tr>
<td>4</td>
<td>13th</td>
<td>3</td>
<td>$0.310\lambda_g \times 0.240\lambda_g$</td>
</tr>
<tr>
<td>5</td>
<td>16th</td>
<td>1</td>
<td>$0.111\lambda_g \times 0.091\lambda_g$</td>
</tr>
<tr>
<td>6</td>
<td>7th</td>
<td>1.67</td>
<td>$0.104\lambda_g \times 0.104\lambda_g$</td>
</tr>
<tr>
<td>7</td>
<td>10th</td>
<td>2</td>
<td>$0.101\lambda_g \times 0.150\lambda_g$</td>
</tr>
<tr>
<td>8</td>
<td>9th</td>
<td>0.5</td>
<td>$0.104\lambda_g \times 0.214\lambda_g$</td>
</tr>
<tr>
<td>9</td>
<td>5th</td>
<td>2.7</td>
<td>$0.134\lambda_g \times 0.323\lambda_g$</td>
</tr>
<tr>
<td>This work</td>
<td>17th</td>
<td>0.8</td>
<td>$0.057\lambda_g \times 0.077\lambda_g$</td>
</tr>
</tbody>
</table>

**IV. CONCLUSION**

A new microstrip lowpass filter is presented in this letter. One prototype filter with 3 dB cutoff frequency at 0.8 GHz has been demonstrated. Results indicate that the demonstrator has the properties of compact size, good passband performance, and ultra-wide stopband. With all these good features, the proposed filter is applicable for modern communication systems.

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Optimization of Interior Permanent Magnet Motor on Electric Vehicles to Reduce Vibration Caused by the Radial Force


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Abstract – The vibration and noise level of a driven motor on electric vehicles has a great influence on the overall comfort of the whole vehicle. In this paper, the main vibrational harmonic waves with high amplitudes of Interior Permanent Magnet (IPM) motor were investigated via an experiment. In order to lower the amplitudes of these harmonic waves, the paper carries out the optimization of radial force of the IPM motor based on the parameter sensitivity analysis and also implemented the structural optimization, according to the dynamic response of the stator. The final dynamic simulation of the optimized model excited by the optimized radial force, show that the optimization results in a better performance of the vibration behaviors. This study could provide some guidelines for the optimal design of the interior permanent magnet motor to reduce vibration.

Index Terms – Interior permanent magnet motor, radial force optimization, stator structural optimization and vibration behavior.

I. INTRODUCTION

An electric vehicle is one of the energy-saving and environmental-friendly transportation tools. The driven motor is one of the most important parts on an electric vehicle. However, the vibration and noise of the motors on vehicles is more serious than that of normal ones, which has a bad influence on the comfort of the vehicle. Thus it is necessary to conduct the study of the causes, factors and optimization of the vibration behaviors of the driven motors on electric vehicles. Interior Permanent Magnet (IPM) motors are most often used to drive the electric vehicles. For this kind of motor, the radial force inside stimulates the stator, which leads to the vibration and noise [1]. Accordingly, the existing studies on the vibration of IPM motors mainly focuses on the calculation and analysis of the radial force and the vibration modes of the stator.

Zhu is one of the earliest scholars to study the vibration and noise characteristics of permanent magnet motors. In his study, the instantaneous magnetic field distribution of brushless permanent magnet DC motor was calculated and the effects of four factors were analyzed: open-circuit field, armature-reaction field, stator slotting and loads. The results of analytical method agreed with the ones of finite element method [2]-[5], which laid the theoretical foundation of the calculation of the radial force; although, he didn’t study the vibrational optimization of the motor. In the 21st century, the electromagnetic numerical method finite element method, as an example, is used more often for the motor analysis. Chen made use of finite element method to analyze the relationship between the radial force and stator teeth, width of rotor back iron and slot opening of the permanent magnet brushless motor. He found that the lower amplitudes of the radial forces resulted from larger stator teeth width, smaller rotor back iron width and slot opening [6]; however, he only pointed out the relationship and didn’t further study the parameter optimization. Zhang studied the radial force of switched reluctance motors using skewed slot structure based on finite element method. The simulated results show that using skewed slot could reduce the radial force [7]; however, he didn’t consider the effects of other parameters on the radial force, such as air gap length, thickness and depth of
permanent magnet. Abbas used analytical and FEM to calculate the radial forces exerted on different parts of the coils [8]; however, he didn’t consider the effects of the radial force on the vibration of the motor.

In addition, there were also some scholars who studied the structural optimization of motor stators. Choi presented a topology optimization method of IPM motor stator cores to reduce cogging torque. The optimization results show that configurations having dummy slots were desirable, but the number of dummy slots differed [9]. Kwack used the level set method to optimize the structure of motor stator in order to reduce torque ripples, by minimizing the difference between torque values at defined rotor positions and the constant target torque value under constrained material usage [10]. Both studied the torque, instead of the vibration of the motors. Additionally, they didn’t use a 3-D model for the optimization.

This paper aims at the vibration reduction of IPM motor by the optimization of radial force and stator structure, which are the two main factors of motor vibration. It discusses the vibration behavior of IPM motor on electric vehicles via an experiment to investigate the main vibrational harmonic waves with high amplitudes. Then the parameters that have influences on the radial forces were optimized according to the sensitivity analysis, as well as the 3-dimentinal model of stator was optimized to reduce the dynamic response of the stator. Lastly, the final simulation of the optimized stator model excited by the optimized radial forces showed that the vibration level of the stator had declined to a large extent than that of the original model, which meant that the vibration behavior of the IPM stator was much better.

II. EXPERIMENTAL ANALYSIS

In this paper, in order to measure the vibration level of the IPM motor a test bench was established, which can take into account the effects of changing loads of torques. The noise was also measured; although, in this experiment we mainly focus on the vibration analysis.

A. Experiment introduction

In this paper, a test bench system was established to measure the vibration of the IPM motor used to drive electric vehicles, as shown in Fig. 1. The system included electric eddy current dynamometer, its controller and cooling water box as the loading subsystem, as well as the protective cover and sound absorption material as the sound isolating subsystem. Nine microphones were arranged around the motor, as shown in Fig. 2 (a) (black points); there were also 4 vibration acceleration transducers pasted on the surface of the motor, as shown in Fig. 2(b).

In order to study the real vibration levels of the IPM motor on an electric vehicle, the working conditions of the testing motor should be the same as those working on electric vehicles. The working conditions of the motor in this experiment included steady ones and transient ones. The steady states of the motor corresponded to those when the vehicle drove with the velocity from 5 Km/h to 40 Km/h; every 5 Km/h as an increment. The transient conditions of the motor corresponded to those when the vehicle sped up with the velocity from 5 Km/h to 40 Km/h. The correspondent rotational speeds and torques were calculated according to the vehicles driving conditions. In this way, the vibration levels of the motor on the test bench were almost the same as on the electric vehicle.
B. Vibration analysis of experimental results

In order to study the relationship of the vibration level and the rotational speed of the motor, the paper carried out the frequency spectrum analysis of the vibration acceleration signals of the motor surface while the motor sped up. The waterfall of motor radial vibration acceleration is shown in Fig. 3.

It could be found from Fig. 3 that there were vibration harmonic waves that changed with the rotational speed; because the law is the same for any rotational speed, it was reasonable to choose one steady state of the motor for analysis to study the harmonic waves of the vibration spectrum. In addition, the rated condition was most often used and was representative; therefore, this paper chose the rated condition for study. The rotational speed was \( n = 2400 \, \text{rpm} \) and the torque was \( T = 30 \, \text{Nm} \). The radial vibration frequency spectrum of the motor under this condition is shown in Fig. 4.

For this working condition, the rotational frequency of the motor was \( f = n / 60 = 40 \, \text{Hz} \). The frequency of controlling current was \( f_i = n \cdot p / 60 = 240 \, \text{Hz} \); where \( p = 6 \) was the number of pole-pairs of the motor. It was shown in Fig. 4 that the main components of motor vibration included the rotational frequency and its second harmonic frequency as well as the second, fourth, and sixth harmonic frequencies of the current frequency; where the sixth harmonic frequency has the highest amplitude.

As the rotational frequency and its second harmonic frequency were caused by the mechanical effects, this paper mainly studied the second, fourth and sixth harmonic waves of the current frequency of the vibration for optimization. Since the vibration of the IPM motor stator is mainly determined by the radial force and the stator structure, this paper implemented the optimization of the radial force and the stator structure, respectively in the following parts.

III. OPTIMIZATION OF RADIAL FORCE

A. Analytical calculation of radial force

According to Maxwell’s law, the radial force was expressed as:

\[
p_n(\theta, t) = \frac{b^2(\theta, t)}{2\mu_0},
\]

where \( \mu_0 \) was the magnetic conductivity of air, \( \mu_0 = 4\pi \times 10^{-7} \, \text{H/m} \) and \( b(\theta, t) \) was air gap magnetic flux density.

The air gap magnetic flux density could be calculated by:

\[
b(\theta, t) = \lambda(\theta, t) f(\theta, t),
\]

where \( \lambda(\theta, t) \) was the air gap magnetic permeance and \( f(\theta, t) \) was the air gap magnetic potential.

The air gap magnetic permeance was composed by 4 parts, as in:

\[
\lambda(\theta, t) = \lambda_0 + \sum_{i=1}^1 \lambda_{1i} + \sum_{i=2}^2 \lambda_{2i} + \sum_{i=1}^1 \sum_{j=2}^2 \lambda_{1i} \lambda_{2j},
\]

where \( \lambda_0 \) was the constant component, \( \lambda_{1i} \) was the harmonic wave magnetic permeance when the rotor was smooth and the stator was slotted and \( \lambda_{2i} \) was the harmonic wave magnetic permeance when the stator was smooth and the rotor was slotted.

The air gap magnetic potential was caused mainly by stator field current and rotor permanent
magnet, as in:

\[ f_c = \frac{N_i I_c}{2}, \]

\[ f_m(\theta,t) = f_0(\theta,t) + \sum_{\mu} f_{\mu}(\theta,t) + \sum_{n} f_n(\theta,t), \]

where \( f_c \) was the air gap magnetic potential caused by stator field current, \( N_i \) was the number of coil in one slot, \( I_c \) was the current in one coil, \( f_m(\theta,t) \) was the air gap magnetic potential caused by rotor permanent magnet, \( f_0(\theta,t) \) was the synthetic magnetic potential of fundamental wave, \( f_n(\theta,t) \) was the harmonic wave magnetic potential of stator winding and \( f_{\mu}(\theta,t) \) was the harmonic wave magnetic potential of rotor.

As for the real motor in this paper, the rotor was smooth, the stator was slotted and there was no air gap permeance. Leaving out the vibration force waves of high orders and low amplitudes, the radial force was expressed as:

\[ p_{r}(\theta,t) = \frac{B_i}{2} \cos(2p\theta - 2\alpha f - 2\phi_0) + \frac{B_i}{2} \sum_{\mu} \sum_{\nu} B_{\mu\nu} \cos[(\mu \pm \nu)\theta - (\alpha \pm \phi_0)t - (\phi_1 \pm \phi_2)]. \]

In this formula, \( B_i \) is the magnetic potential of the basic wave, \( P \) is the number of pole-pairs of the motor, \( \theta \) is the angular displacement in the air gap, \( t \) is the time, \( \mu, \nu \) are the orders of the harmonic waves, \( \omega \) is the angular frequency of the harmonic wave and \( \phi \) is the phase of the harmonic wave.

Similar with the working condition in the experiment, the radial force of the rated condition was calculated and the frequency spectrum is shown in Fig. 5.

Figure 5 shows that the main components of radial force included 480 Hz, 960 Hz and 1439 Hz, which are in accordance with the vibration frequency components measured in the experiment. They were respectively the second, fourth and sixth harmonic frequencies of the current frequency. Accordingly, the analytical calculation could reflect the harmonic waves of the radial force.

**B. Magnetic finite element simulation**

Magnetic finite element modeling:

For the 3-phase, 6-state IPM permanent motor with 18 teeth and 12 poles that was studied in this paper, the number of slots per pole per phase is 1/2. It was equal to 6 unit motors and each unit motor had 3 teeth and 2 poles. Since every unit motor had the same magnetic field, this paper used unit motors for magnetic simulation to save time. According to the real parameters of the motor, the magnetic model of 2 unit motors was established in ANSOFT. Then the stator, rotor, air gap and magnets were meshed with different sizes. After meshing of the model, the paper got the finite element model of the unit motors, as shown in Fig. 6.

**Fig. 6. Magnetic model and finite element model of the unit motors.**

Calculation of radial force:

Under the rated working condition, the
rotating speed was 2400 r/min, the load was 18 N/m, the voltage was 120 v and the electric current frequency was 240 Hz. The Maxwell 2-D transient magnetic field solver was used for the simulation. The simulation step was 0.0001 s and the time duration was 0.5 s. For the finite element model of the unit motors under rated condition, the radial force in the air gap was calculated and its frequency spectrum is shown in Fig. 7.

Fig. 7. Frequency spectrum of radial force in the air gap by simulation.

It was shown in Fig. 7 that the main frequencies of the radial force were the second, fourth and sixth harmonic frequencies of the current frequency. The results were in accordance with the main frequencies of analytical calculation, as in Fig. 5 and the experimental result, which confirmed the validity of the simulation. It was shown in both Figs. 5 and 7 that the energy of the second harmonic waves was more than 40% of the whole energy; the fourth and sixth harmonic waves were respectively about 20%. This means that the energy distribution and the ratio of harmonic waves were consistent. Considering that in the analytical calculation the equation (2) is an approximation, there is some deviation of the amplitudes between the analytical results and the simulation result. Therefore, the paper used the finite element model, which was more accurate than analytical model, for further analysis.

It was known from the experiment that the sixth harmonic wave of the current of the vibration had the highest amplitude, so the reduction of the amplitude of this wave in the radial force is the key for the improvement of the vibration behavior.

C. Parameter sensitivity analysis

It was obvious that the thickness and depth of the permanent magnet of IPM motor had an influence on the air gap flux density, which had an effect on the radial force. Additionally, the existing studies showed that the air gap length and the slot opening width also had an influence on the air gap flux density, which then affected the frequency and amplitude of the radial force [11]-[12]. This section used these four parameters for sensitivity analysis of the radial force under rated condition in order to choose the sensitive parameter intervals for optimization to get lower radial forces. Only one parameter changed at one time, the other settings remained unchanged. It should be noticed that the parameter ranges in this section all meet the requirement of design handbook of permanent magnetic motor, which means that the variations won’t affect the performances of the motor too much [1].

The previous study in this paper showed that second, fourth and sixth harmonic waves of the radial force was the main exciting source of the vibration of the motor; therefore, this section mainly studied the parameter sensitivity to the second, fourth and sixth harmonic waves of radial force. During the study of the influence on the radial force of one parameter, other parameters remained unchanged. The studied ranges of the objective parameters were around and symmetrical about the original values of the motor. As the values were quite small, it is difficult and costly to manufacture extremely precise motors. Thus, it is not necessary to conduct a statistical study. The paper divided each parameter range into four intervals and four analysis, as shown in Table 1. Then when the value of the studied parameter is respective to the boundary values of the intervals, the second, fourth and sixth harmonic frequencies of the radial force are calculated. The results of the radial forces when the four parameters differed, are shown in Fig. 8. The comparison of the results indicated the laws that the radial force increased slowly with the increase of the width of stator slot openings, decreased slowly with the increase air gap length, increased rapidly with the increase of thickness of the permanent and decreased rapidly with the increase of depth of the permanent.
Table 1: Parameter intervals division

<table>
<thead>
<tr>
<th>Parameters (mm)</th>
<th>Interval 1</th>
<th>Interval 2</th>
<th>Interval 3</th>
<th>Interval 4</th>
<th>Original Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>Slot opening width</td>
<td>(0.5, 1.5)</td>
<td>(1.5, 2.5)</td>
<td>(2.5, 3.5)</td>
<td>(3.5, 4.5)</td>
<td>2.5</td>
</tr>
<tr>
<td>Air gap length</td>
<td>(0.7, 0.8)</td>
<td>(0.8, 0.9)</td>
<td>(0.9, 1.0)</td>
<td>(1.0, 1.1)</td>
<td>0.9</td>
</tr>
<tr>
<td>Thickness of permanent magnet</td>
<td>(2.5, 3.5)</td>
<td>(3.5, 4.5)</td>
<td>(4.5, 5.5)</td>
<td>(5.5, 6.5)</td>
<td>4.5</td>
</tr>
<tr>
<td>Depth of permanent magnet</td>
<td>(0.1, 0.7)</td>
<td>(0.7, 1.3)</td>
<td>(1.3, 2.3)</td>
<td>(2.3, 3.3)</td>
<td>1.3</td>
</tr>
</tbody>
</table>

As the value of these four parameters differed so much, this paper used the following method to carry out the sensitivity analysis. Take the air gap length for example, the original air gap length was \( l_0 \) and the amplitude of sixth harmonic wave of the radial force was \( b_0 \). One interval of the air gap length was \( l_i \), when the air gap length was respectively \( l_1 \) and \( l_2 \), the corresponding amplitudes of sixth harmonic waves of the radial forces were \( b_1 \) and \( b_2 \). Then the change rate of air gap length in this interval was \( k_i \), which was taken as the sensitivity index [13], was defined as:

\[
k_i = \left[ \frac{(b_2 - b_1)/b_0}{(l_2 - l_1)/l_0} \right].
\]

It could be seen that \( k_i \) was a dimensionless quantity. It considered the original dimensions and the change at the same time so it could scientifically reflect the parameter sensitivity to radial force.

The change rates of other parameters were calculated in the same way with air gap length. The sensitivities of the four parameters in all intervals to the second, fourth and sixth harmonic waves are shown in Fig. 9. In Fig. 9, the positive and negative values meant that the amplitudes of radial force increased or decreased with the increase of the studied parameter.

Fig. 8. Calculated radial forces of different parameters.
D. Parameter optimization

It was known from the previous analysis that it was important to reduce the amplitudes of the sixth harmonic waves of radial forces to reduce vibration. Those parameter intervals with high sensitivities to the sixth harmonic waves were interval 1 of width of permanent magnet, interval 3 of depth of permanent magnet, interval 1 and interval 2 of the width of stator slot opening, and all the intervals of the air gap length. Thus, the parameter values should be determined in these intervals respectively.

The paper determined the parameter values in order to achieve lower amplitudes of the radial forces according to what is analyzed above. The parameters should be chosen in the sensitive intervals and could result in smaller radial forces. In this way, the desired radial forces could be achieved without changing the parameters too much. Accordingly, the width of permanent magnet should be as small as possible in interval 1, the depth of permanent magnet should be as large as possible in interval 3 and the air gap length should be as large as possible in the four intervals. As for the stator slot opening width, the values in interval 1 were too small, which might lead to some errors and high costs of manufacture and installation. Also, with consideration of the sensitivity of the second and fourth harmonic waves, the value of slot opening width should be as small as possible in interval 2. Under the comprehensive consideration, the finally determined values of the parameters were shown in Table 2.

Table 2: Optimized parameter values

<table>
<thead>
<tr>
<th>Parameters</th>
<th>Values (mm)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Thickness of permanent magnet</td>
<td>2.5</td>
</tr>
<tr>
<td>Depth of permanent magnet</td>
<td>2.3</td>
</tr>
<tr>
<td>Air gap length</td>
<td>1.1</td>
</tr>
<tr>
<td>Slot opening width</td>
<td>1.5</td>
</tr>
</tbody>
</table>

According to the optimized parameters in Table 2, the new finite element model of unit motors was established in software ANSOFT, to calculate the radial forces. The main frequencies of the radial force after optimization were the second, fourth and sixth harmonic frequencies of the current frequency; which were the same with those before optimization, but the amplitudes of the radial forces are optimized. The comparison of amplitudes of radial forces before and after optimization are shown in Table 3. It can be seen that the amplitudes of second, fourth and sixth harmonic waves had been reduced to a large extent; especially the amplitude of sixth harmonic wave, which declined by two-thirds.

IV. STATOR STRUCTURAL OPTIMIZATION

The purpose of the stator structural optimization in this paper was that when the weight and boundary dimension remained unchanged, the dynamic displacement response caused by the given excitation should be as small as possible. That is to say, when the excitation remains unchanged, the vibrational
energy of the stator should be reduced to improve the vibration behavior of the motor.

Table 3: Force amplitudes of main harmonic waves

<table>
<thead>
<tr>
<th>Harmonic Waves</th>
<th>Second</th>
<th>Fourth</th>
<th>Sixth</th>
</tr>
</thead>
<tbody>
<tr>
<td>Original amplitude (mm)</td>
<td>74160</td>
<td>34360</td>
<td>29790</td>
</tr>
<tr>
<td>Optimized amplitude (mm)</td>
<td>39870</td>
<td>21990</td>
<td>9967</td>
</tr>
<tr>
<td>Optimized percentage</td>
<td>46%</td>
<td>36%</td>
<td>67%</td>
</tr>
</tbody>
</table>

A. Determination of parameters and objective function for optimization

This chapter carried out the stator structural optimization of IPM motor on electric vehicles. The study aimed to improve the dynamic response behavior of the stator without changing the performance of the motor. During the design of motors, inner diameter and effective length of the iron core were the main parameters that determined the performance of the motor. The original values of the motor structural parameters are shown in Table 4.

Table 4: Motor parameter values

<table>
<thead>
<tr>
<th>Parameters</th>
<th>Values</th>
</tr>
</thead>
<tbody>
<tr>
<td>No. of pole-pairs</td>
<td>$2p = 6$</td>
</tr>
<tr>
<td>Inner diameter of iron core</td>
<td>$D_{11} = 130, \text{mm}$</td>
</tr>
<tr>
<td>Outer diameter of iron core</td>
<td>$D_1 = 180, \text{mm}$</td>
</tr>
<tr>
<td>Shell thickness</td>
<td>$\Delta h = 4, \text{mm}$</td>
</tr>
<tr>
<td>Yoke thickness</td>
<td>14 mm</td>
</tr>
<tr>
<td>Length of iron core</td>
<td>86 mm</td>
</tr>
</tbody>
</table>

The existing study indicated that the parameters that had an influence on the vibration modes of motor stator were the shell thickness, outer diameter of iron core and thickness of stator yoke [14]. As the yoke thickness was associated with the inner and outer diameters of the stator, when the inner diameter remains unchanged, the yoke thickness changes with the outer diameter. Thus, this paper chose the outer diameter of iron core and shell thickness as optimization variables. During the optimization, the unchanged inner diameter and effective length of the iron core ensured the performance of the motor.

The original value of shell thickness was 4 mm, which was quite small. With consideration of the requirement of structural strength, the range of shell thickness for optimization should be a little larger, as 4–8 mm in this paper. In addition, the volume of the motor shouldn’t be too large, so the stator boundary dimension was set as less than 5% larger than the original value. At the same time, the minimum of the stator yoke thickness was set as 5 mm to avoid the magnetic saturation. Accordingly, the range of outer diameter of the iron core was 81–90 mm.

The optimization was achieved in software ANSYS and the whole process was conducted with ANSYS Parameter Design Language. The range between the maximum and the minimum of the dynamic displacement that was the fluctuating amplitude of the dynamic displacement of the stator surface, reflected the vibration energy of the stator. In order to reduce the vibration level of the motor stator, it was effective to reduce the fluctuating amplitude of the dynamic displacement of the stator surface, which was chosen as the objective function for optimization in this paper.

Accordingly, the parameters were set as follows during the optimization:

1. Design variables:
   - Outer radius of the iron core: 81–90 mm;
   - Shell thickness: 4–8 mm.
2. State variables:
   - $R_1 - R_{11} \geq 16\, \text{mm}$ (stator yoke thickness is not smaller than 5 mm)
   - Stator boundary dimension $\leq 1.05 \times 190\, \text{mm}$ (original value)
3. Objective function:
   - Fluctuating amplitude of dynamic displacement of the stator surface.

B. Optimization result

The process was programmed with ANSYS Parameter Design Language. First, the 3-D model of stator was established, as shown in Fig. 10. Second, the model was excited by the time-domain radial forces calculated by simulation in chapter 3.2.2. Then the transient analysis started. During the calculation, the shell thickness and outer radius of the iron core of the model changed to another set of values, according to equal step sweeping method of ANSYS and another step of transient analysis began. During every step, the objective function was recorded and it changed with the step number of the optimization, as shown in Fig. 11. Figure 11 showed that the objective function of the 26th step...
number was the smallest, which meant the corresponding parameters were approximately the best. Then these parameters were rounded off, as shown in Table 5.

<table>
<thead>
<tr>
<th>Parameters</th>
<th>Boundary Dimension (mm)</th>
<th>Outer Diameter of Iron Core (mm)</th>
<th>Shell Thickness (mm)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Values</td>
<td>192</td>
<td>176</td>
<td>8</td>
</tr>
</tbody>
</table>

Fig. 10. Finite element model of the stator.

Fig. 11. Changes of objective function during optimization.

It was shown that after optimization the stator boundary dimension was only 2 mm larger than the original value, which met the requirement that it was less than 5% larger than the original value. Additionally, the outer diameter of the iron core decreased by 4 mm and the shell thickness increased by 4 mm. The transient displacement responses of the stator surface before and after optimization with the same excitation are shown in Fig. 12. The corresponding amplitudes of the main frequencies are shown in Table 6. In Table 6 it shows that after optimization the amplitudes of the main frequencies decreased to quite a large extent. That means that the dynamic response of the motor stator with the same excitation reduced after optimization.

Fig. 12. Transient dynamic displacement responses of the stator before and after optimization.

<table>
<thead>
<tr>
<th>Harmonic Frequencies</th>
<th>Second (480Hz)</th>
<th>Fourth (960Hz)</th>
<th>Sixth (1440Hz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Original amplitude (mm)</td>
<td>6.69e-5</td>
<td>2.83e-5</td>
<td>5.12e-5</td>
</tr>
<tr>
<td>Optimized amplitude (mm)</td>
<td>3.85e-5</td>
<td>2.04e-5</td>
<td>3.68e-5</td>
</tr>
<tr>
<td>Percentage of optimization</td>
<td>42%</td>
<td>28%</td>
<td>28%</td>
</tr>
</tbody>
</table>

V. TRANSIENT DYNAMIC ANALYSIS OF THE OPTIMIZED MODEL

After the optimization of the radial force and the stator structure, a new model was established for simulation to verify the optimization effect. The optimized finite element model of the stator was excited by the optimized radial force for frequency response analysis.

Then the paper conducted the frequency
response analysis of the rated condition, which was a steady state. The results of the frequency response analysis showed that after optimization the main harmonic frequencies of the vibration acceleration were the same with those before optimization. They were 480 Hz, 960 Hz and 1439 Hz, which were the second, fourth and sixth harmonic frequencies of the current frequency; especially the sixth harmonic wave with the highest amplitude. The amplitudes of the main harmonic waves decreased to a large extent. The comparison of the amplitudes of the harmonic waves are shown in Table 7. It could be found from Table 7 that the average decrease of the amplitudes was 60%, which meant the optimization reduced the vibration level greatly. Accordingly, the optimization was quite ideal.

Table 7: Vibration acceleration amplitudes of main frequencies

<table>
<thead>
<tr>
<th>Frequency (Hz)</th>
<th>480</th>
<th>960</th>
<th>1439</th>
</tr>
</thead>
<tbody>
<tr>
<td>Original amplitudes (m/s²)</td>
<td>0.7</td>
<td>0.6</td>
<td>1.9</td>
</tr>
<tr>
<td>Optimized amplitudes (m/s²)</td>
<td>0.17</td>
<td>0.20</td>
<td>1.0</td>
</tr>
<tr>
<td>Percentage of optimization</td>
<td>76%</td>
<td>67%</td>
<td>37%</td>
</tr>
</tbody>
</table>

VI. CONCLUSION

The following conclusions could be drawn from this study on the analysis of the vibration behaviors of the motor and its optimization.

In this paper, the main vibration harmonic waves of the IPM motor on electric vehicles were investigated by frequency spectrum analysis of the experimental result. It was found that the second, fourth and sixth harmonic waves of the current frequency in the vibration had the highest amplitudes.

The optimization of the radial force based on the parameter sensitivity analysis could result in smaller radial forces without changing the parameters too much. It was also desirable to use the 3-D model of the stator for structural optimization to reduce the dynamic response of the stator.

The final transient dynamic simulation of the optimized model excited by the optimized radial forces showed that the vibration level decreased by 60% averagely, which verified the good effects of the optimization in this paper.

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REFERENCE


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