A New Method for Linear Antenna Array Design

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Abstract

The inherent relation between the linear antenna array and the finite impulse response filter of digital signal processing is revealed and the least square linear-phase finite impulse response filter design method is introduced for the linear antenna array to obtain a similar main lobe width of uniform linear antenna array but much lower sidelobe level.

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Introduction: The linear antenna array (LAA) is a widely used for airborne target detection. The main efforts are focusing on reducing its sidelobe level while trying to keep the main lobe being narrow. In addition, the phase control concept was introduced to allow the LAA scanning, and the uniform linear antenna array (ULAA) is always the research platform [1]. From the LAA phase point of view, the ULAA is just the finite impulse response (FIR) filter of digital signal processing (DSP) [2] with the rectangle window function (RWF). Beside RWF, there are many high-performance ones available for LAA design. However, the availability of low sidelobe level is at the price of the wider main lobe for the window functions based LLA. A more effective FIR filter design method called least-square linear-phase design can be employed for the LLA design to balance the tradeoff between the low sidelobe level and narrow main lobe.

The relation between LLA and FIR Filer: The LAA can be shown as Fig.1. The distance d between the adjacent elements is $\lambda/2$, where λ is the wavelength; and the incident angle is θ . So, the phase difference between the adjacent elements is kdcos θ , where k is $2\pi/\lambda$.



Fig. 1. Uniform Linear Antenna Array diagram

Then, the LLA has the antenna factor (AF) :

 $AF(\theta) = \sum_{n=0}^{N-1} I_n e^{-jnkd\cos\theta}$ (1)

where I_n is the exciting current of the nth element, N is the total number of the elements of the LLA.

Let $\varphi = kdcos\theta = \pi cos\theta$, then formular (1) becomes:

$$AF(\theta) = \sum_{n=0}^{N-1} I_n e^{-jn\varphi}(2)$$

On the other hand, the frequency response definition of the FIR filter is:

$$H(z) = \sum_{n=0}^{N-1} h(n) z^{-n}$$
(3)

When z is $e^{\vartheta\omega}$, formular (2) meets this definition. So, the AAA san be regarded as the PIP filter of $\Delta\Sigma\Pi$, and the VAAA is dust the PIP filter with PDP.

ΛΛΑ δεσινν ωπη λεαστ-σχυαρε λινεαρ-πηασε μετηρό: Λετ της ΦΙΡ φιλτερ λενγτη βε Λ+1 σαμπλες, ωιτη Λ εεν. Ινιτιαλλψ ιτ ωιλλ βε ςεντερεδ αβουτ της τιμε οριγιν, ι.ε., ζερο πηασε. Τηςν της φρεχυενςψ ρεσπονσε ςαν βε ςαλςυλατεδ ατ της σπεςιφιεδ φρεχυενςψ $ω_k$ βψ

$$H(\omega_k) = \sum_{n=-L/2}^{L/2} h(n) e^{-j\omega_k n}, \quad k = 0 \text{ to } N - 1.$$
 where N>>A (4)

Ρεγαρδινγ εεν σψμμετρψ ιν της ιμπυλσε ρεσπονσε, ι.ε., η(ν)=η(-ν), γιες α ζερο-πηασε ΦΙΡ φιλτερ τηατ ςαν βε λατερ ριγητ-σηιφτεδ $\Lambda/2$ σαμπλες το μαχε α ςαυσαλ, λινεαρ πηασε φιλτερ. τηεν, της φρεχυενςψ ρεσπονσε ρεδυζες το α συμ οφ ςοσινες:

$$H(\omega_k) = h(0) + 2 \sum_{n=1}^{L/2} h(n) \cos(\omega_k n),$$
where
x=0,1, 2, ..., N-1, N>> Λ . (5)

Λετ της δεσιρεδ Φ IP φιλτερ φρεχυενςψ ρεσπονσε βε $D(\omega_k)$, τηςν της δεσιγν ταργετ ις το μινιμιζε της φολλοωινη σχυαρε ερρορ:

 $\min ||H(\omega_k) - D(\omega_k)||^2 (6)$

Ωηεν της λεαστ σχυαρε ερρορ ις ρεαςηεδ, της ρελατεδ η(ν) ςαν βε οβταινεδ, ανδ η(ν) ις ρεαλ νυμβερ σινςε της Φ IP φιλτερ ις ενφορςεδ το βε εεν σψμμετρψ. Ιν ουρ ςασε, τηις η(ν) ςαν βε ρεγαρδεδ ας της εξςιτινη ςυρρεντ οφ εαςη ελεμεντ οφ της ΛΑΑ.

 $\Sigma o,$ the AAA design procedure of using method can be:

- 1. Σπεςιφψ της τοταλ ελεμεντ νυμβερ οφ ΛΛΑ.
- 2. Σπεςιφψ της νορμαλιζεδ ςυτ-οφφ πηασε οφ της μαιν λοβε.
- 3. Σπεςιφψ τηε νορμαλιζεδ σταρτ πηασε οφ τηε σιδελοβε.
- 4. Υσε της φιρλς φυνςτιον ιν σιγναλ προςεσσινγ τοολβοξ οφ ΜΑΤΛΑΒ [3],

Here is a design example with the MATAAB source sodes for an 8-edement AAA. For the normalized phase, i.e., φ is between 0 and 1, set the cut-off phase of the main lobe to 0.1, the start phase of the sidelobe from 0.2 to 1. By the exciting current of each element for this LAA is as the follows:

$$I_0 = I_7 = 0.0804,$$

$$I_1 = I_6 = 0.1147,$$

$$I_2 = I_5 = 0.1422$$

$$I_2 = I_4 = 0.1574$$

The pattern diagram of this LAA vs that of ULAA is as the follows:



Fig. 2. Pattern diagram, ULAA vs Least square Linear Phase LAA

Obviously, the 3dB main lobe of this LAA is very close to that of ULAA, while the 1st side lobe level of this LAA is about 7dB lower than that of ULAA.

The MATLAB source codes is listed as the follows:

```
N=8;
 \% 8-element LAA, the distance between two adjacture elements is 1/2 wavelend
 A=zeros(1,8); %Initial the exciting current of the ULAA
 B=zeros(1,8); % Initial the exciting current of the LAA
 rectangleoutput=zeros(1,99); % Initial the frequency response of ULAA
 leastsquareoutput=zeros(1,99); % Initial the frequency response of LAA
 theta=zeros(1,99); % initial the incident angles
 sum1=0;% DC component of the ULAA
 sum2=0;% DC component of the LAA
 f = [0 \ 0.1 \ 0.2 \ 1];
                             % desired and normalized Frequency band edges
 a = [1 1 0 0];
                          % desired and normalized Amplitudes
 B = firls(7,f,a); % Least Square Linear phase function in the toolbox
□ for j=1:N
     A(j)=1; % assign 1 for each element of ULAA
     sum1=sum1+A(j); % calculate DC component of the ULAA
     sum2=sum2+B(j); % calculate DC component of the LAA
 end
 theta_step=pi/100; % The step for the incident angle
☐ for K=1:99
     theta(K)=theta_step*K; % obtain all of the incident angles
 end
□ for K=1:99
     temp_variable1=0;
     temp_variable2=0;
     for j=1:N
        temp_variable1=temp_variable1+A(j)*exp(-1i*pi*cos(theta(K))*(j-1));
        temp_variable2=temp_variable2+B(j)*exp(-1i*pi*cos(theta(K))*(j-1));
    end
        rectangleoutput(K)=20*log10(abs(temp_variable1)/sum1);
        leastsquareoutput(K)=20*log10(abs(temp_variable2)/sum2);
 end
 plot(theta/pi,rectangleoutput,theta/pi,leastsquareoutput);
 legend('ULAA','firls design','FontSize',14);
 xlabel('Normalized Incident Angle (\times\pi rad/sample)','FontSize',16);
 ylabel('Magnitude in dB', 'FontSize',16);
```

Conclusion: Generally, the LAA can be regarded as the FIR filter of DSP with respect of its phase. And ULAA is just the FIR filter with RWF. To have the better performance, square linear-phase design method for FIR filter can be directly employed for the LAA design.

References

- 1. COLLIN, R. E.:' Antennas and Radiowave Propagation'. (McGraw-Hill, 1985)
- 2. OPPENHEIM, A. v., and SCHAFER. R.W.: 'Digital signal processing' (Prentice-Hall, New Jersey, 1975)
- 3. MATLAB R2018 User Manual



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