

## Research of DOA Estimation Based on MUSIC Algorithm

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**Abstract.** This paper studies MUSIC algorithm in direction of arrival(DOA) estimation of signals, expounds the principle of this algorithm, conducts simulation for uniform linear array with Matlab, concludes that MUSIC algorithm has different performances in estimating direction of arrival with different number of arrays, different array element spacing or different SNR. The algorithm has high resolution, estimation precision and stability.

### Introduction

Array signal processing is an important branch of signal processing, direction of arrival(DOA) estimation is an important part of array signal. MUSIC algorithm is used in DOA estimation. [1] MUSIC algorithm decomposes the covariance matrix of received signal, promotes the rise of structure characteristics classification algorithm, has a good performance in DOA estimation and becomes symbol algorithm of spatial spectrum estimation theory. [2] This paper compares and simulates with Matlab the characteristics of MUSIC algorithm in different conditions.

### MUSIC algorithm

Suppose there are M uniform linear array, N incident narrowband source waves, the source direction is  $\theta_1 \theta_2 \dots \theta_N$ ,  $s_i(k)$  is the i-th complex amplitude of the source,  $S(k) = [s_1(k), \dots, s_N(k)]^T$ . Array direction vector is  $a(\theta_i) = [1, e^{-j\omega_i}, \dots, e^{-j(M-1)\omega_i}]^T$  ( $i = 1 \dots N$ ),  $A = [a(\theta_1), \dots, a(\theta_N)]$ ,  $\omega_i = \frac{2\pi d}{\lambda} \cdot \sin(\theta_i)$ ,  $d$  is the distance between array elements,  $\lambda$  is the wavelength of carrier. [3]

According to the N received signals, get the estimation value of covariance matrix, decompose the eigenvalue of covariance matrix.

$$R_x = E[X(k)X^H(k)] = AR_s A^H + \sigma^2 I \quad (1)$$

According to the order of the characteristic value, get N eigenvalues  $\lambda_1, \lambda_2 \dots \lambda_N$  (by the order), take the number of eigenvalues and eigenvectors equal to the signal number for signal space, take the rest of the eigenvalues and eigenvectors for noise space, get the noise matrix. With the noise characteristic vector for columns, make the noise matrix

$$E_n = [v_{N+1}, v_{N+2} \dots v_M]^T \quad (2)$$

The eigenvectors correspond to noise eigenvalues and the column of matrix A correspond to source direction is orthogonal, so

$$A^H v_i = 0 \quad (i = N + 1, N + 2 \dots M) \quad (3)$$

Define the space spectrum

$$P_{music}(\theta) = \frac{1}{a^H(\theta) E_n E_n^H a(\theta)} \quad (4)$$

When the columns of  $a(\theta)$  and  $E_n$  is orthogonal, the denominator is zero. Actually, because

of the existed noise, the denominator is a small value, corresponds to the peak of  $P_{music}$ , the job is to find spectral peak to get the DOA estimation. [5]

## Simulation and comparison

### DOA estimation performance of different number of array elements

Experiment uses 3 separate narrowband signal source, the center frequency is 1000 Hz, signal-to-noise ratio is 10, incident to the linear array from 60 degrees, 30 degrees, 45 degrees, the sampling number is 128, take half wavelength as array element spacing. Take different numbers of array element of uniform linear array for simulation, the results are shown in figure 1.

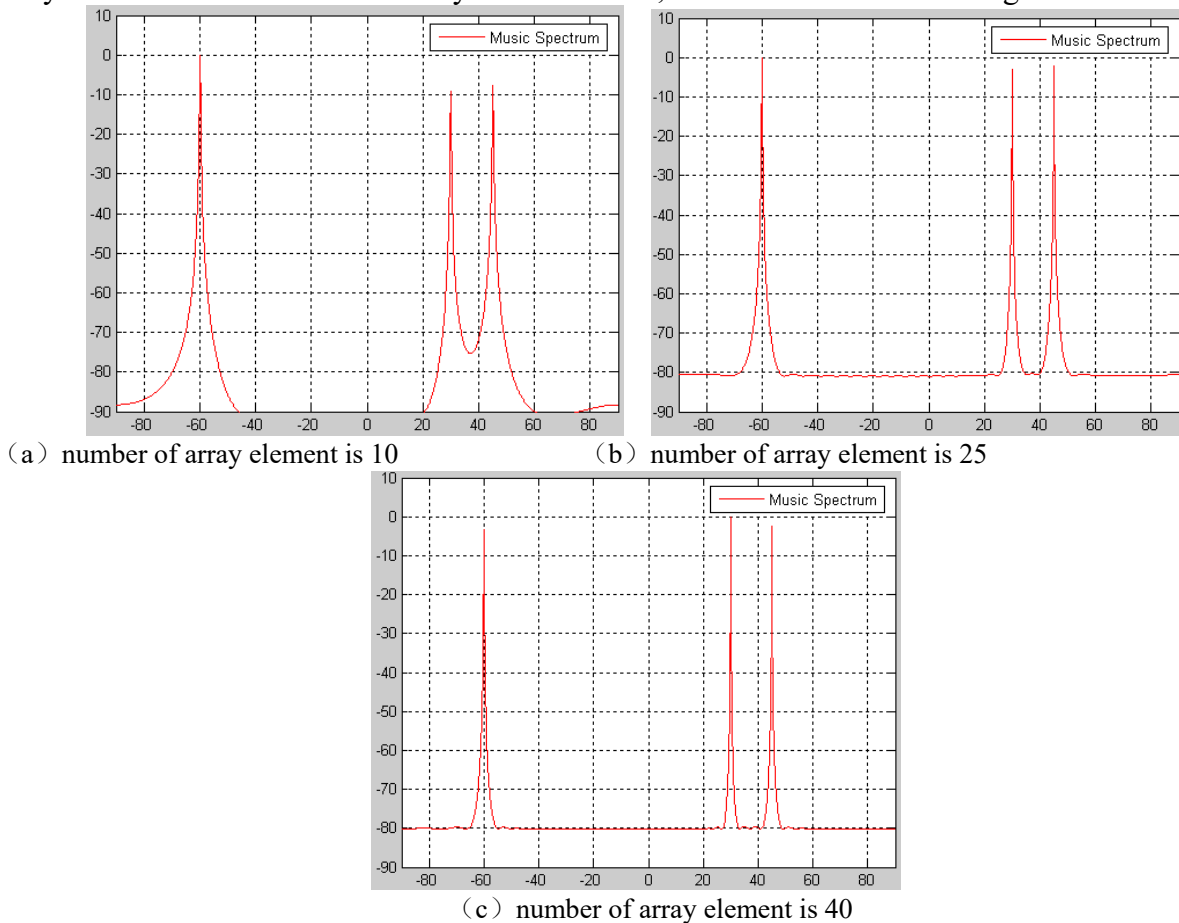


Fig.1. Estimation performance of different number of array elements

Figure 1 shows that with the increase of distance between array elements, the resolution of MUSIC algorithm for DOA estimation has improved.

### DOA estimation performance of different distances between array elements

Experiment uses 10 array element, 3 separate narrowband signal source, the center frequency is 1000 Hz, signal-to-noise ratio is 10, incident to the linear array from 60 degrees, 30 degrees, 45 degrees, the sampling number is 128. Take different distances between array elements of uniform linear array for simulation, the results are shown in figure 2.

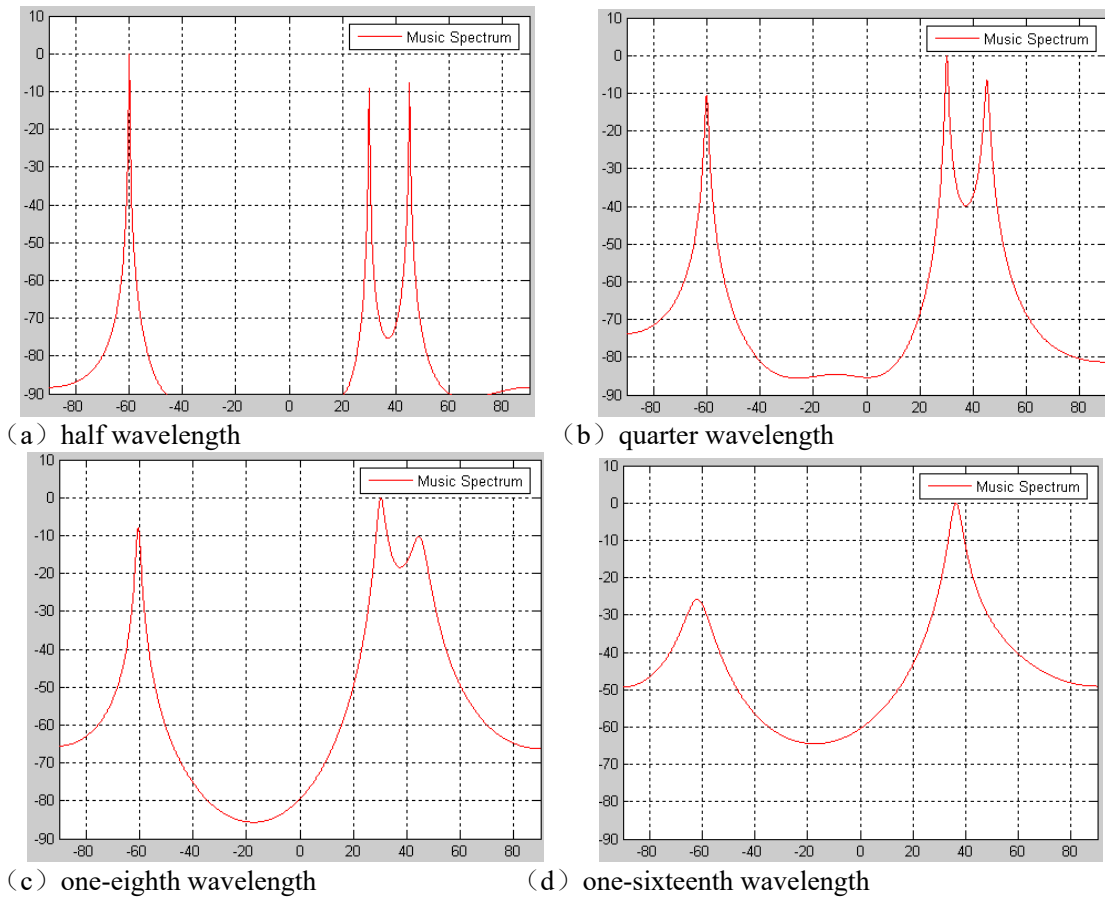
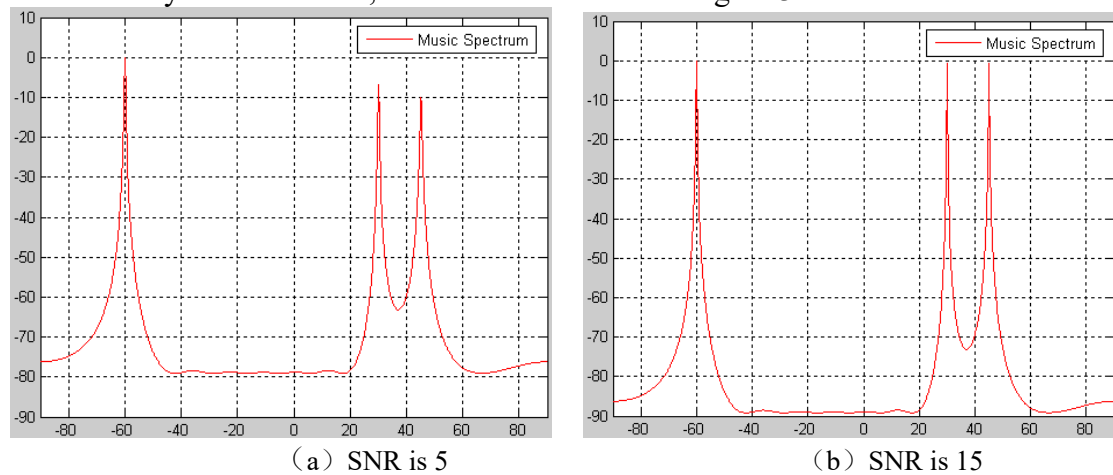


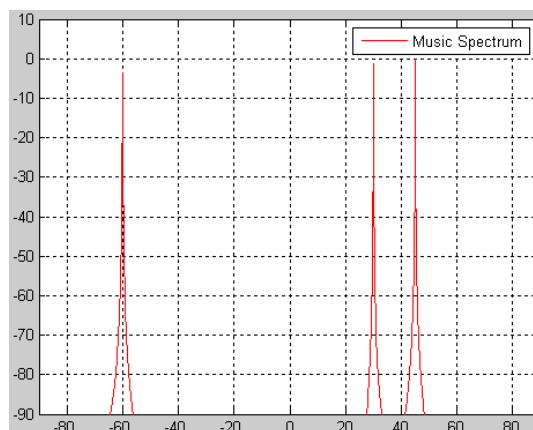
Fig.2. Estimation performance of different distance between array elements

Figure 2 shows that with the increase of distance between array elements, the resolution of MUSIC algorithm for DOA estimation has improved.

**DOA estimation performance of different signal-to-noise ratio**

Experiment uses 10 array element, 3 separate narrowband signal source, the center frequency is 1000 Hz, incident to the linear array from 60 degrees, 30 degrees, 45 degrees, the sampling number is 128, take half wavelength as array element spacing. Take different signal-to-noise ratio of uniform linear array for simulation, the results are shown in figure 3.





(c) SNR is 25

Fig.3. Estimation performance of different signal-to-noise ratio

Figure 3 shows that with the increase of signal-to-noise ratio, the resolution of MUSIC algorithm for DOA estimation has improved.

## Conclusion

This paper analyzes the principle of MUSIC algorithm, simulates and compares the DOA estimation performance with different number of arrays, different array element space, different SNR. Experimental results show that with the increase of array element number, array element space, and signal-to-noise ratio, MUSIC algorithm for DOA estimation has higher resolution. Under certain conditions the algorithm has high estimation precision and stability.

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