Determining the Number of Coherent/Correlated Sources Using FBSS-based Methods

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Abstract Determining the number of sources from observed data, is a fundamental problem in array signal processing. In this paper, first we focus on two popular estimators based on information theoretic criteria, AIC and MDL. Then another algorithm based on eigenvalue grads, namely EGM is presented. The computer simulation results prove the effective performance of the EGM for non-coherent signals but in the small differences between the incident angles of non-coherent sources, MDL and AIC have a much better detection performance than EGM. These methods can detect only non-coherent signals, and the performance of them will be sharply declined even signals are coherent and/or correlated. So, first forward/backward spatial smoothing (FBSS) method is used as a pre-processing step to solve the coherency/correlation, and then MDL, AIC and EGM algorithms are run to estimate the number of signals. Numerical results show that FBSS-based EGM offers higher detection probability rather than FBSS-based MDL and AIC in the case of coherent sources as well as correlated ones. Also, the higher detection probability can be achieved for correlated case compared to coherent one.

Keywords Information Theoretic Criteria, AIC, MDL, EGM, Coherent Signals, Correlated Signals, Spatial Smoothing, FBSS

1. Introduction

Wireless direction finding is the procedure for determining signal sources by observing signal direction of arrivals (DOAs). Its history came back to the beginning of wireless communications. This technology is used in many fields such as radar, sonar, mobile communications, astronomy, seismology and etc. So, DOA estimation using sensor arrays and direction finding, are important subjects in signal processing[1]. Most of the DOA estimation algorithms, assume that the number of sources is known a priori and may give misleading results if the wrong number of sources is used. So, determining the number of signals is an important problem in field of array signal processing[2].

In order to determine the number of sources to satisfy the requirement of DOA estimator, many parametric and nonparametric detection methods have been proposed[3]. A number of related methods have been widely studied in[4]. One of the most widely-used approaches is that of information theoretic criteria, which introduced by Wax and Kailath for the first time. They proposed an approach to this problem, based on the Akaike’s information criterion (AIC) and minimum description length (MDL). According to these two algorithms, the number of signal sources is determined as the value for which the AIC or MDL criterion is minimized[5].

However, despite these methods are efficient in detection point of view, they involve the estimation of a covariance matrix and its eigenvalue decomposition which introduce generally high complexity. In order to reduce the computational complexity and attain accurate detection performance, a low-complexity MDL method is developed in[6]. This method employs the training data of the desired signal, to quickly partition the array data into two orthogonal components in signal and noise subspaces.

To enhance the probability of detection, an accurate minimum description length (MDL) method was devised in[7]. Since the training sequence of the desired signal is used to calculate the minimum mean square error (MMSE), the performance of proposed method is significantly superior to the traditional MDL method at low SNRs and small number of snapshots.

Luo Jing Qing[8] proposed a set of eigenvalue gradient methods (EGMs), which like AIC and MDL methods, it also determines the number of non-coherent sources according to the eigenvalues of auto-correlation matrix. Another algorithm that considers a bound or threshold for eigenvalues is investigated in[9]. This approach is a simple and efficient way to increase detection probability in the case of unequal power sources.

Above discussed techniques assume that noise is spatially uncorrelated. In practice, signals experience colored noise which causes decreasing the performance of conventional
methods, rapidly. To overcome this problem, a signal enumerator for spatially correlated noise has been proposed in [10]. They combine the combined information theoretic criteria and eigenvalue correction to estimate the number of coherent signals.

Li Jing-hua and Li Rong [11] established a novel effective detection algorithm in spatial color noise by using the canonical correlation coefficients of the joint covariance matrix. Compared with the other algorithms, this algorithm has a better performance and lower complexity.

In the real environment, the coherent source is common, and the above-mentioned methods are suitable for non-coherent signals, and their performance will be sharply declined even signals are coherent or correlated [12]. A partial solution to this problem, applicable to coherent or correlated sources, was proposed in [13]. The method uses the MDL principle and decomposes data into signal and noise components. The MDL descriptor is then computed from the sub-array output covariance matrices [14, 15]. The second method is more applicable because SS is a pre-processing required for both determining the number of sources and DOA estimation.

The main purpose of this paper is to determine the number of sources in three types of sources, non-coherent, coherent, and correlated, based on MDL, AIC and EGM methods. First, conventional MDL, AIC and EGM algorithms are applied to detect the number of non-coherent sources. Then, by applying spatial smoothing as a pre-processing part, above mentioned algorithms are used for detecting the number of coherent/correlated sources.

The paper is organized as follows. After the statement and formulation of the problem in Section 2, Section 3 presents the derivation of the three detection criteria for non-coherent sources (MDL, AIC and EGM). In addition, two experiments and associated simulation results are presented. Section 4 presents forward/backward spatial smoothing (FBSS) technique for determining the number of coherent/correlated sources. Moreover, it illustrates simulation results of two experiments, coherent and correlated sources. Finally, Section 5 concludes this research.

2. Signal Model

Consider P narrowband signals emitted from the far field impinging on array of M sensors \( P < M \). The \( M \times 1 \) measurements of the output of the array corrupted by additive noise can be expressed as

\[
X(t) = A(\theta)S(t) + N(t)
\]  

(1)

where \( A(\theta) = [a_1(\theta), a_2(\theta), \ldots, a_P(\theta)] \) is the matrix of array manifold, \( S(t) = [S_1(t), S_2(t), \ldots, S_P(t)]^T \) is the vector of signal, and \( n(t) \) is the additive noise. The additive noise is assumed to be complex, ergodic Gaussian vector stochastic process, independent of the signals, with zero mean and covariance matrix given by \( \sigma_n^2 I \) [16]. The covariance matrix for array is

\[
\hat{\Sigma} = \frac{1}{N} \sum_{i=1}^{N} X(t_i)X^H(t_i)
\]  

(3)

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

To solve the problem, determining the number of sources from \( \hat{\Sigma} \), we consider the following assumptions [13].

1. The array manifold, defined as the set of steering vectors, \( a(\theta) \) is known.
2. Any subset of \( M \) steering vectors from the array manifold is linearly independent.
3. The number of sensors is greater than the number of sources, namely, \( M > P \).

3. Determining the Number of Non-coherent Sources

In this section, first, three well-known algorithms, MDL, AIC and EGM, are described for non-coherent signals. Consequently, related simulation results that show the performance of these algorithms are illustrated with more details.

3.1. Estimating the Number of Signals with Information Theoretic Criteria

The information theoretic criteria for model selection, introduced by Akaike Schwartz and Rissanen, address the following general problem: Given a set of \( N \) observation \( X = \{x(1), x(2), \ldots, x(N)\} \) and a family of models, that is, a parameterized family of probability densities \( f(X|\theta) \) selects the model that best fits the data.

Akaike’s proposal was to select the model which gives the minimum AIC, defined by

\[
AIC = -2 \log f(X|\theta) + 2k
\]  

(4)

where \( \theta \) is the maximum likelihood estimate of the parameter vector \( \theta \), and \( k \) is the number of free adjusted parameters in \( \theta \). Inspired by Akaike’s pioneering work, Schwartz and Rissanen approached the problem from quite different points of view. Rissanen’s approach is based on information theoretic arguments. Since each model can be used to encode the observed data, Rissanen proposed to select the model that yields the minimum code length. It turns out that in the large-sample limit, both Schwartz’s and Rissanen’s approaches yield the same criterion, given by...
MDL = \(- \log f(X|\theta) + \frac{1}{2} k \log N\) \hspace{1cm} (5)

To apply the information theoretic criteria to detect the number of signals, we can say that \([16]\)

\[
f(x(t_1), ..., x(t_k)) \theta^{(k)}
\]

\[
= \prod_{i=1}^{N} \frac{1}{\pi^d \det R^{(k)}} \exp - x(t_i)^{-1} R^{(k)} \neg x(t_i)
\]

(6)

Taking the logarithm and omitting terms that do not depend on the parameter vector \(\theta^{(k)}\), we find the log-likelihood function \(L(\theta^{(k)})\) as

\[
L(\theta^{(k)}) = -N \log \det R^{(k)} - \text{tr}[R^{(k)}]^{-1} R
\]

(7)

The maximum-likelihood estimate is the value of \(\theta^{(k)}\) that maximizes (7). These estimates are

\[
\hat{V}_i = \xi_i; \quad i = 1, ..., k
\]

(8a)

\[
\hat{\lambda}_i = l_i; \quad i = 1, ..., k
\]

(8b)

\[
\sigma_n = \frac{1}{M-k} \sum_{i=k+1}^{M} l_i
\]

(8c)

Substituting the maximum likelihood estimates (8) in the log-likelihood (7), with some straightforward manipulations, we obtain

\[
L(\theta) = \log \left[ \prod_{i=k+1}^{M} \frac{1}{(M-k)} \right]^{(M-k)N}
\]

(9)

The form of AIC for this problem is therefore given by equation (10).

\[
AIC(k) = -2 \log \left[ \prod_{i=k+1}^{M} \frac{1}{(M-k)} \right]^{(M-k)N} + 2k(2M-k)
\]

while the MDL criterion is given by equation (11).

\[
MDL(k) = - \log \left[ \prod_{i=k+1}^{M} \frac{1}{(M-k)} \right]^{(M-k)N} + \frac{1}{2} k (2M-k) \log N
\]

(11)

The number of signals \(\hat{d}\) is determined as the value of \(k \in \{0,1, ..., M-1\}\), for which either the AIC or the MDL is minimized [16].

3.2. Estimating the Number of Signals Using EGM

Just like traditional AIC and MDL methods, EGM family also determine the number of non-coherent sources according to the eigenvalues of auto-correlation matrix[8]. The first step is calculating the spatial auto-correlation matrix of the output data \(X(t)\) of the sensor array by (3). The second step is applying eigen-decomposition on auto-correlation \(R_X\), and arranges the eigenvalues in descending order,

\[
\hat{\lambda}_1 \geq \hat{\lambda}_2 \geq \hat{\lambda}_{k+1} \geq ... \geq \hat{\lambda}_M
\]

(12)

There is a significant difference between \(\hat{\lambda}_k\) and \(\hat{\lambda}_{k+1}\). So, the number of signal \(k\) can be determined by checking the difference between neighbour eigenvalues. This is the main idea of the set of EGM methods. A common used checking method named EGM1[8] is cited as follows:

1. Define the average grads of all eigenvalues by

\[
\Delta \lambda = (\lambda_i - \lambda_{i+1}) / (M-1)
\]

(13)

2. Calculate the gradients of all neighbor eigenvalues as

\[
\Delta \hat{\lambda}_i = \hat{\lambda}_i - \hat{\lambda}_{i+1}
\]

(14)

3. Find out all \(i\) satisfying \(\Delta \hat{\lambda}_i \leq \Delta \lambda\) to construct the set \(\{i_k\}\).

4. Take the \(i_0\) that is the first one of the last continuous block of \(i\) in the set \(\{i_k\}\), and the estimated number of signals is \(\hat{d} = i_0 - 1\).

3.3. Simulation of Non-Coherent Signals

Computer simulations have been carried out to examine the effectiveness of above mentioned algorithms in the case of non-coherent sources. Detection probability is the performance metric to evaluate and also compare MDL, AIC and EGM algorithms.

The performance of determining the number of non-coherent sources with MDL, AIC and EGM under different SNRs is evaluated in the first experiment. In this experiment, the uniform linear array with 10 sensors is used, element spacing is half-wavelength and SNR is defined as

\[
\text{SNR} = 10 \log \left( \frac{\sigma_s^2}{\sigma_n^2} \right)
\]

(15)

Two non-coherent signals with equal powers impinge on the array at 0° and 10°, SNR changes from -15 dB to 30 dB with step size 1 dB and the number of snapshots is 100. For each SNR, 1000 Monte Carlo trials are run to find probability of success.

In Fig. 1, the probabilities of detection for MDL and AIC are compared with EGM in the case of non-coherent sources. According to this Figure, for SNRs lower than -10 dB, the performance of EGM is better than both MDL and AIC methods. For higher SNRs, greater than -5 dB, MDL and EGM show 100% success but AIC offers detection probability a bit lower than 100%.

![Figure 1. The detection probability versus SNR for experiment 1](image-url)
8 sensors is used. Two non-coherent signals with equal powers impinge on the array, the angular distance changes from 1 to 10 degrees and the number of snapshots is 100.

The simulation results of experiment 2 are shown in Fig. 2. As depicted in this figure, in small differences between the incident angles (lower than 6°), MDL and AIC have a much better detection performance than EGM. In addition, for differences lower than 6°, EGM performance degrades drastically. In the other hand, MDL and AIC performance will be decreased for the angular differences lower than 2°. Moreover, for the differences greater than 7°, EGM is a much better than MDL and also AIC.

A major problem with above methods (MDL, AIC and EGM) is that, they are not applicable to the case of fully correlated signals, referred to as the coherent signals. This case appears especially in multipath propagation and therefore it is of great practical importance.

In this investigation, we focused on spatial smoothing method to solve the coherency/correlation, and then the number of coherent sources is estimated considering the pre-processed output signal of array antenna.

The spatial smoothing scheme first suggested by Evans and extensively studied by Shan[14]. Their solution is based on a pre-processing scheme that divides the total array of sensors into overlapped sub-arrays and then, generates the average of the sub-array output covariance matrices. This forward-only smoothing scheme makes use of a larger number of sensor elements than the conventional ones, and in particular requires 2K sensor elements to estimate any K directions of arrival. In[15], it is proved that by simultaneous use of a set of forward and complex conjugated backward sub-arrays, it is always possible to estimate any K directions of arrival using at most[3K/2] sensor elements. So, in this research work, we pre-processed coherent signals using FBSS as the first step of the process of determining the number of coherent sources.

4.1. Forward/Backward Spatial Smoothing Technique for Coherent/Correlated Signal Identification

FBSS scheme starts by dividing a uniform linear array with M sensors into uniformly overlapping sub-arrays of size $M_0$ (see Fig. 3). Let $x_1^l(t)$ stands for the output of the $l$th sub-array for $l = 1,2,...,L ≡ M - M_0 + 1$ where L denotes the total number of these forward sub-arrays[15].
Then, the covariance matrix of the $l$th sub-array is given by
\[ R_l = E[x_l(t)x_l^T(t)] \] (17)

Forward spatially smoothed covariance matrix $R^f$ as the mean of the forward sub-array covariance matrices is
\[ R^f = \frac{1}{L} \sum_{l=1}^{L} R_l \] (18)

The covariance matrix of the $l$th backward sub-array is given by
\[ R^b = E[x_l(t)x_l^T(t)] \] (19)

Finally, the forward/backward smoothed covariance matrix $\bar{R}$ as the mean of $R^f$ and $R^b$ is given by
\[ \bar{R} = \frac{R^f + R^b}{2} \] (20)

Since the smoothed covariance matrix $\bar{R}$ in (21) has exactly the same form as the covariance matrix for some non-coherent situations, the eigenstructure-based techniques can be applied to this smoothed covariance matrix to successfully estimate the number of coherent sources[15].

4.2. Simulation of Coherent/Correlated Sources

Computer simulations have been carried out to examine the effectiveness of above mentioned algorithms in coherent case. An uniform linear array with 9 sensors is used, SNR changes from $-10 \, \text{dB}$ to $10 \, \text{dB}$ with step size 1 dB and the number of snapshots is 100. For each SNR, 1000 Monte Carlo trials are run.

As the experiment 3, the performance of FBSS-based MDL, AIC and EGM algorithms under different SNRs for coherent sources is investigated. Three coherent signals with equal powers impinge on array at $0^\circ$, $10^\circ$ and $20^\circ$. Correlated signals are obtained by filtering the signal through a first order auto-recursive (AR1) filter, given by
\[ y(i) = \alpha y(i-1) + x(i) \] (22)

where $\alpha \in [0,1]$ is the correlation coefficient. In this experiment $\alpha = 0.8$ is assumed. Simulation results are shown in Fig. 5. In this case, FBSS-based methods can effectively detect the number of sources and offer high detection probabilities even much better than experiment 3. According to Fig. 5, EGM offers higher performance with respect to MDL and AIC methods. As a highlighted remark, EGM method shows high detection probability for low SNRs (lower than $-4 \, \text{dB}$).

5. Conclusions

In this paper, we focused on the performance evaluation of three popular methods, AIC, MDL as well as EGM. As shown in simulation results, these algorithms are appropriate for determining the number of non-coherent signals and the performance of them will be decreased when signals are coherent and/or correlated. A solution to this problem, applicable to coherent as well as correlated non-coherent sources, is to use spatial smoothing method to solve the coherency or correlation, and then estimating the number of signals.

Computer simulations showed that, in the case of non-coherent signals, in low SNRs, EGM is better than MDL and AIC. Also, in small angular differences, MDL and AIC algorithms offer a higher detection performance with respect to EGM.

Moreover, simulation results for FBSS-based algorithms, in the case of coherent sources as well as correlated ones, showed the higher detection probability for EGM compared to AIC and MDL.

In this research we compared MDL, AIC and EGM algorithms in additive white Gaussian noise (AWGN) channel. Eigen increment threshold (EIT) method which is proposed as a method to determine the number of sources can be researched in AWGN channel and compared to AIC, MDL and EGM algorithms in both non-coherent and coherent cases. In addition, these methods can be investigated in colored noise scenario and in the case that
exist different groups of non-coherent sources containing coherent signals. Modeling and determining the number of sources with unequal powers is another new research work to be done in this field.

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REFERENCES


