

# A new direction-of-arrival estimation method using automotive radar sensor arrays

Seunghoon Cho, Heemang Song, Kyung-Jin You and Hyun-Chool Shin

## Abstract

This article presents a new signal processing method of estimation of direction of arrival using the phase difference between sensors. The new method is essentially a narrow-band technique for automotive radar. A delay of the reception time is caused by the physical gap between sensors in the incident signals to the sensor arrays, and it results in a phase difference between input signals. The new method predicts the ideal phase value and the phase value of the input signal. The point that minimizes the phase error for every sensor is estimated as direction of arrival. The simulation result shows that the new method offers significantly improved estimation resolution and direction-of-arrival estimation compared to conventional methods.

## Keywords

Automotive radar, radar sensor signal processing, sensor array, radar sensor beamforming, automotive direction-of-arrival estimation

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## Introduction

Recently, the number of vehicles equipped with an automotive radar is increasing steadily. The automotive radar is typically used for advanced driver-assistance systems (ADASs),<sup>1–3</sup> which include advanced smart cruise control (ASCC) for automatic adjustment of distance between vehicles and blind spot detection (BSD). The most important factor for using these systems is the location of objects because the locations of objects must be accurately identified for the stable use of the systems.

The conventional radar sensors were used to get information about the distance and speed of objects, but various systems have been introduced with the development of radar technology. For example, the importance of the direction-of-arrival (DOA) estimation algorithm, which has combined the high-resolution digital beamforming method with sensor array, has increased. The DOA is estimated as the direction of

reflection of the incident signals. The digital beamforming method uses a sensor array<sup>4</sup> and is divided into parametric and spectral methods depending on the signal processing method.<sup>5</sup> In this study, the spectral method based on spatial spectrum was used. The spectrum-based beamforming method finds the point that has the maximum value in the results of spatial spectrum that have the incident direction as a variable and is classified into beamforming techniques and subspace-based methods. The beamforming techniques include Bartlett and Capon algorithms,<sup>6,7</sup> and the subspace-based methods include multiple signal classification (MUSIC) algorithm.<sup>8,9</sup>

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further permission provided the original work is attributed as specified on the SAGE and Open Access pages (<http://www.uk.sagepub.com/aboutus/openaccess.htm>).

In this article, a new algorithm to improve the DOA estimation is proposed. The new method is essentially a narrow-band technique for automotive radar. In the new method, DOA is estimated by investigating the phase difference by a time delay. The steering vector is built by calculating a phase difference by the physical gap between sensors. The phase of the steering vector can be determined by the arrival angle to sensor arrays, but the actual angle information for the phase of the input signal is hard to extract because the steering vector is combined with the complex envelope. To obtain the angle information of objects in the phase of input signal, the actual phase difference was estimated. The new method estimates the DOA as the point where the difference between the estimated ideal phase difference in the sensor array and the phase difference from the actual input signal is minimized. The new method was experimentally verified to estimate the DOA more accurately than the conventional algorithms, and the spectrum results confirmed a higher resolution of the new method. The spectral resolution was quantified using kurtosis,<sup>10</sup> which is used as a measure of the sharpness of spectral beamforming, and it was found that the new method has a higher resolution compared to the conventional algorithms.

## Radar system

Figure 1 shows the structure of the frequency-modulated continuous-wave (FMCW) radar.<sup>11</sup> It consists of the antenna unit where an electrical radio frequency (RF) signal is converted to an electromagnetic wave, the transceiver unit that generated the RF signal and processes the received RF signal, and the signal control and processing unit that controls the RF signal and processes the received signal. The antenna unit contains the linear patch array antenna, and the signal control and processing unit handles the schedule of FMCW modulation and radar scanning. The transmission channel is a single

channel, and the receiving channel consists of  $M$  uniform linear arrays.

## DOA estimation algorithms

In the DOA estimation algorithm, the spectral technique is a method of estimating the DOA as  $\hat{\theta}$  that has the maximum value on the spectrum with the incident direction  $\theta$  as a variable. The spectral methods are divided into beamforming technique or subspace-based method depending on the estimation method of the spectrum.

### Beamforming techniques

The beamforming technique is one of the basic DOA estimation methods using a sensor array. The spatial spectrum is formed using the output values obtained by orienting the sensor array to all directions. The direction representing the maximum value is estimated as the final DOA

$$\mathbf{A} = [a(\theta_1), a(\theta_2), \dots, a(\theta_K)] \quad (1)$$

$$\begin{aligned} \mathbf{a}(\theta) &= [a_1(\theta), a_2(\theta), \dots, a_M(\theta)]^T \\ &= [e^{-j\phi_1}, e^{-j\phi_2}, \dots, e^{-j\phi_M}]^T \end{aligned} \quad (2)$$

$$\phi_i = 2\pi f \tau_i = 2\pi f \frac{d(i-1) \sin \theta}{f\lambda} = 2\pi \frac{d(i-1) \sin \theta}{\lambda}, \quad (3)$$

for  $i = 1, 2, \dots, M$

In equation (1),  $\mathbf{A}$  denotes the steering matrix,  $\mathbf{a}(\theta)$  denotes the steering vector of the sensor array response to a specific direction  $\theta$ , and  $K$  denotes the number of plane waves. In equation (2),  $M$  denotes the number of sensors. In equation (3),  $\phi_i$  and  $\tau_i$  denote the phase difference and time delay at the  $i$ th sensor, respectively. Here,  $f$  denotes frequency,  $d$  denotes the distance between sensors, and  $\lambda$  denotes the wavelength.

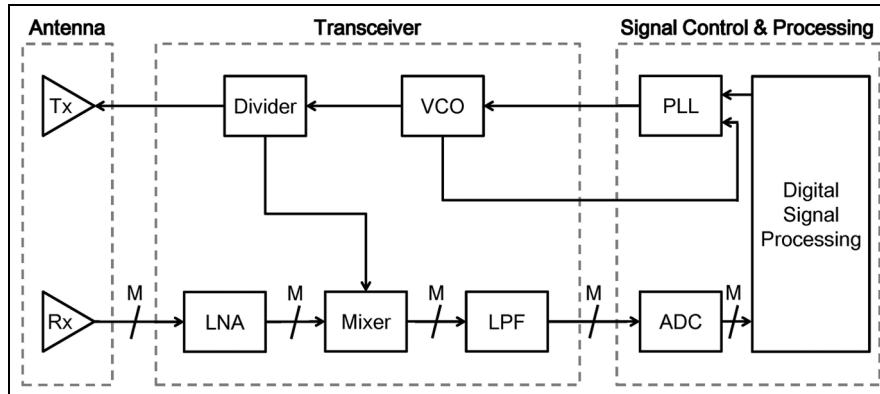


Figure 1. Radar system structure.

The input signal  $x_i$  at the  $i$ th sensor is expressed as follows

$$x_i(t) = a_i(\theta_{opt})s_{opt}(t) + n_i(t), \quad \text{for } i = 1, 2, \dots, M \quad (4)$$

where  $s_{opt}$  denotes the complex envelope of the plane wave that appears when the object in the actual input signal is  $\theta_{opt}$ ,  $n_i$  denotes the noise at the  $i$ th sensor. Thus, the input signal  $X$  in the sensor array antenna is expressed as follows

$$X = [x_1(t), x_2(t), \dots, x_M(t)]^T \quad (5)$$

The output  $y$  of the sensor array consisting of  $M$  sensors is a linear combination of sensor outputs with the weight of each sensor as a coefficient and is expressed as follows

$$y = \sum_{i=1}^M w_i^* x_i(t) = \mathbf{w}^H X \quad (6)$$

where  $w_i$  denotes the weight at the  $i$ th sensor,  $*$  denotes the complex conjugate, and  $H$  denotes the conjugate transpose. The output power  $P(\mathbf{w})$  of the sensor array is defined as follows

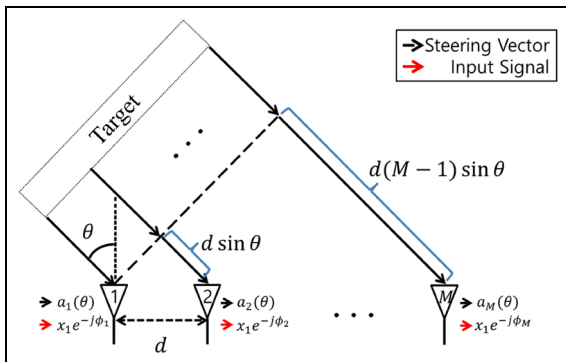
$$P(\mathbf{w}) = E[|y|^2] = \mathbf{w}^H E[XX^H] \mathbf{w} = \mathbf{w}^H \mathbf{R} \mathbf{w} \quad (7)$$

where  $\mathbf{R}$  denotes the covariance matrix of the input signal. The weighting vector  $\mathbf{w}$  varies between the Bartlett beamforming algorithm and the Capon beamforming algorithm.

Figure 2 shows that a time delay is generated according to the sensor by the gap of the sensor array, which causes a phase difference.

### Bartlett beamformer

The Bartlett beamforming algorithm maximizes the signal output by giving a large weight to the input signal from a specific direction.



**Figure 2.** Time delay and phase difference.

The weighting vector  $\mathbf{w}_{BA}$  to maximize the output of the sensor array for input signals from a specific direction  $\theta$  is as follows

$$\mathbf{w}_{BA} = \frac{\mathbf{a}(\theta)}{\sqrt{\mathbf{a}^H(\theta)\mathbf{a}(\theta)}} \quad (8)$$

By substituting equation (8) in equation (7), the spatial spectrum of Bartlett beamforming can be expressed as follows

$$P_{BA}(\theta) = \frac{\mathbf{a}^H(\theta)\mathbf{R}\mathbf{a}(\theta)}{\mathbf{a}^H(\theta)\mathbf{a}(\theta)} \quad (9)$$

### Capon beamformer

The Capon beamforming algorithm maintains a constant gain for the input signals from a specific direction and gives a smaller weight to the noise. The optimization problem to satisfy these conditions is as follows

$$\min_{\mathbf{w}} \mathbf{w}^H \mathbf{R} \mathbf{w} \quad \text{subject to } \mathbf{w}^H \mathbf{a}(\theta) = 1 \quad (10)$$

The above optimization problem maintains the gain of a specific direction at 1 and minimizes the noise to increase the signal-to-noise ratio (SNR). To find the solution of equation (10) using this, the weight vector  $\mathbf{w}_{CA}$  of the Capon beamforming algorithm can be expressed as follows

$$\mathbf{w}_{CA} = \frac{\mathbf{R}^{-1}\mathbf{a}(\theta)}{\mathbf{a}^H(\theta)\mathbf{R}^{-1}\mathbf{a}(\theta)} \quad (11)$$

By substituting equation (11) in equation (7), the spatial spectrum of Capon beamforming can be expressed as follows

$$P_{CA}(\theta) = \frac{1}{\mathbf{a}^H(\theta)\mathbf{R}^{-1}\mathbf{a}(\theta)} \quad (12)$$

### The MUSIC algorithm

The MUSIC algorithm is a subspace-based algorithm which uses the characteristic that every steering vector corresponding to the incident signal is at right angle to the noise space. The covariance matrix  $\mathbf{R}_X$  of the input signal is expressed as follows using eigen decomposition

$$\begin{aligned} \mathbf{R}_X &= E[XX^H] \\ &= E[(\mathbf{A}\mathbf{S} + \mathbf{N})(\mathbf{A}\mathbf{S} + \mathbf{N})^H] \\ &= \mathbf{A}E[\mathbf{S}\mathbf{S}^H]\mathbf{A}^H + E[\mathbf{N}\mathbf{N}^H] \\ &= \mathbf{A}\mathbf{R}_S\mathbf{A}^H + \sigma^2\mathbf{I} \end{aligned} \quad (13)$$

$$\begin{aligned} R_X e_i &= (AR_S A^H + \sigma^2 I) \\ e_i &= \mu_i e_i + \sigma^2 e_i = (\mu_i + \sigma^2) e_i \quad \text{for } i = 1, 2, \dots, M \end{aligned} \quad (14)$$

where  $S$  denotes wavefront signal,  $N$  denotes noise vector,  $R_S$  denotes the correlation matrix of  $S$ ,  $\sigma^2$  denotes the power of noise, and  $I$  denotes the unit matrix. In equation (14),  $\mu_i$  is the eigenvalue excluding noise, and  $e_i$  is the eigenvector corresponding to the eigenvalue. If the eigenvalue including noise is  $\lambda_i$ , the following relational expressions are valid

$$\lambda_i = \mu_i + \sigma^2 \quad \text{for } i = 1, 2, \dots, M \quad (15)$$

$$\lambda_1 \geq \lambda_2 \geq \dots \geq \lambda_K > \lambda_{K+1} = \dots = \lambda_M = \sigma^2 \quad (16)$$

$$\begin{aligned} R_X e_i &= (AR_S A^H + \sigma^2 I) e_i \\ &= \lambda_i e_i = \sigma^2 e_i, \quad \text{for } i = K + 1, \dots, M \end{aligned} \quad (17)$$

From the above equations, the following results are obtained

$$AR_S A^H e_i = 0 \quad \text{for } i = K + 1, \dots, M \quad (18)$$

$$A^H e_i = 0 \quad \text{for } i = K + 1, \dots, M \quad (19)$$

$$a^H(\theta_k) e_i = 0 \quad \begin{matrix} \text{for } k = 1, 2, \dots, K \\ i = K + 1, \dots, M \end{matrix} \quad (20)$$

When the power of noise is equal to the eigenvalue, the eigenvector is at right angle to the steering vector. This eigenvector is called noise eigenvector and the noise subspace  $E_N$  is expressed as follows using this noise eigenvector

$$E_N = [e_{K+1}, e_{K+2}, \dots, e_M] \quad (21)$$

Using a random steering vector  $a(\theta)$  and a noise subspace, the spatial spectrum of MUSIC is defined as follows

$$P_{MU}(\theta) = \frac{1}{a^H(\theta) E_N E_N^H a(\theta)} \quad (22)$$

## A new DOA estimation method

A delay of the reception time is caused by the gap between sensors in the incident signals to the sensor arrays, and it results in a phase difference between input signals. The new method uses the ideal phase value for  $M$  sensors in the specific direction  $\theta$  and the phase value of the input signal. The DOA is estimated by comparing the phase difference between two sensors for all sensors using the phase angle of the steering vector with  $\theta$  as a variable and the phase angle of the input signal.

The equation for finding the difference between the ideal phase angle and the actual phase angle of the input signal is expressed as follows

$$\Delta\phi_i = \phi_i - \phi_{i,opt} \quad \text{for } i = 1, 2, \dots, M \quad (23)$$

where  $\phi_i$  denotes the phase angle of the steering vector, and  $\phi_{i,opt}$  denotes the phase containing the angle of objects in the input signal. Equation (23) can be used to express the phase difference for every sensor as a vector as follows

$$\Delta\phi = [\Delta\phi_1, \Delta\phi_2, \dots, \Delta\phi_M] \quad (24)$$

The point that minimizes the difference for every sensor in equation (24) is estimated as the DOA. The optimization problem to satisfy this condition is as follows

$$\min_{\theta} \|\Delta\phi\|^2 = \max_{\theta} \frac{1}{\|\Delta\phi\|^2} \quad (25)$$

The optimization problem in equation (25) is to estimate the DOA as the  $\theta$  that minimizes  $\|\Delta\phi\|^2$ . For actual input signals, the accurate value of  $\phi_{i,opt}$  cannot be known because of equation (4). Thus, the optimization problem of equation (35) is solved by the following method. Here, it is assumed that the input signals have no noise

$$\|\Delta\phi\|^2 = \sum_{i=1}^M |\phi_i - \phi_{i,opt}|^2 \quad (26)$$

$$\phi_i - \phi_{i,opt} = 2\pi \frac{d(i-1)}{\lambda} (\sin \theta - \sin \theta_{opt}) \quad (27)$$

The above equations can be used to determine  $\Delta\phi$  at different  $\theta$  values. However,  $\phi_{i,opt}$  cannot be determined because  $\theta_{opt}$ , which indicates the angle of the actual object, cannot be known. Thus, this problem is solved using the following method

$$\phi_{i,opt} = 2\pi \frac{d(i-1)}{\lambda} \sin \theta_{opt} \quad (28)$$

Equation (28) expresses the phase angle  $\phi_{i,opt}$  that contains the angle data of objects in the input signal. The phase angle of the input signals is expressed as follows

$$\angle x_i = \angle s_{opt} - 2\pi \frac{d(i-1)}{\lambda} \sin \theta_{opt} \quad (29)$$

$$\angle x_1 = \angle s_{opt} \quad (30)$$

where  $\angle$  denotes the phase angle. Using equations (29) and (30), the phase difference between the phase angle of the input signal obtained from the first sensor and the phase angle of the input signals obtained from the  $i$ th sensor is obtained as follows

$$\angle x_1 - \angle x_i = 2\pi \frac{d(i-1)}{\lambda} \sin \theta_{opt} = \phi_{i,opt} \quad (31)$$

Equation (31) can be used to determine the phase angle  $\phi_{opt}$  that contains the angle data of the objects in the input signals. Furthermore, the phase angle  $\phi_i$  of the steering vector can be expressed as follows

$$\angle a_1 - \angle a_i = 2\pi \frac{d(i-1)}{\lambda} \sin \theta = \phi_i \quad (32)$$

Using equations (31) and (32),  $\Delta\phi_i$  can be expressed as follows

$$\Delta\phi_i = (\angle a_1(\theta) - \angle a_i(\theta)) - (\angle x_1 - \angle x_i) \quad (33)$$

From the above equation, it can be seen that the difference in the phase difference of input signals between the two sensors is the same as the phase difference of the steering vector in the sensors 1 and  $i$ .

Figure 3 illustrates the new method for the direction  $\theta$ . Figure 3(a) shows the input signal and Figure 3(b) shows the phase angle of the input signal. Figure 3(c) shows the phase difference at the first sensor and the  $i$ th sensor. Figure 3(d) shows the phase difference at the first sensor and  $i$ th sensor of the steering vector and changes by  $\theta$ . This data can be used to represent all angles by changing  $\theta$ , and the spatial spectrum in Figure 3(e) can be obtained.

Equation (33) can be rewritten as  $(\angle a_1(\theta) - \angle x_1) - (\angle a_i(\theta) - \angle x_i)$ , representing the difference between the phase angle of the steering vector and the phase angle of the input signal for each sensor. The difference at the  $i$ th sensor is defined as  $c_i(\theta)$  as follows

$$c_i(\theta) = \angle a_i(\theta) - \angle x_i \quad (34)$$

$$P_{NM}(\theta) = \frac{1}{\sum_{i=2}^M |c_1(\theta) - c_i(\theta)|^2} \quad (35)$$

The above equations can be used to form the spatial spectrum and applied to equation (25) to obtain the following equation to estimate the final DOA

$$\hat{\theta} = \arg \max_{\theta} P_{NM}(\theta) \quad (36)$$

## Experiment results

The data obtained using the 77-GHz FMCW short-range radar (SRR) in a chamber environment was used as input signal. Figure 4 shows the chamber photograph. In this chamber, radar signal was generated by single and dual target simulator.

The input signal ranged between  $-47^\circ$  and  $47^\circ$  in  $0.5^\circ$  intervals, and a total of 1952 scans with 8 – 12

scans for each angle were experimented. Scan is the time unit which consists of transmission, reception, and signal processing. A single scan duration period is 50-ms long. Additionally, scan-based single snapshot method was used when applying the Bartlett, MUSIC, and the new method. The Capon was applied by 21 snapshots. The number of sensors in a sensor array was four and the sensor gap was  $0.6\lambda$ , and the number of targets for each angle was one.

Figure 5 shows the spatial spectrum calculated with each method when the actual angles of objects are  $28^\circ$ ,  $8.5^\circ$ ,  $-11^\circ$ , and  $30.5^\circ$ . The maximum point in the corresponding spectrum is estimated as the DOA. The actual estimation of the DOA is limited to the range of  $-50^\circ$  to  $50^\circ$ .

Figure 6 shows the averages and standard deviations of errors resulting from the difference between the actual angle and the estimated angle for a total of 1952 data with 8–12 data for each angle. Bartlett and MUSIC have the same values for all data. The standard deviation of Capon is relatively large. This result confirms that the new method has a smaller error of DOA estimation than the Capon algorithm.

Table 1 shows the average errors of the actual and estimated angles and the average standard deviation at each angle for all data. The new method has the smallest average error, which is smaller than  $1^\circ$ . Furthermore, the new method has the smallest standard deviation compared to the conventional Bartlett and Capon algorithms.

Compared to the conventional algorithms, the new method forms a sharper main lobe for the estimated angle and has almost no side lobe, as shown in Figure 5. To quantify this sharpness, kurtosis was used. Kurtosis is a measure of deviation of the probability distribution and is used to measure how much the observation values are crowded at the center.

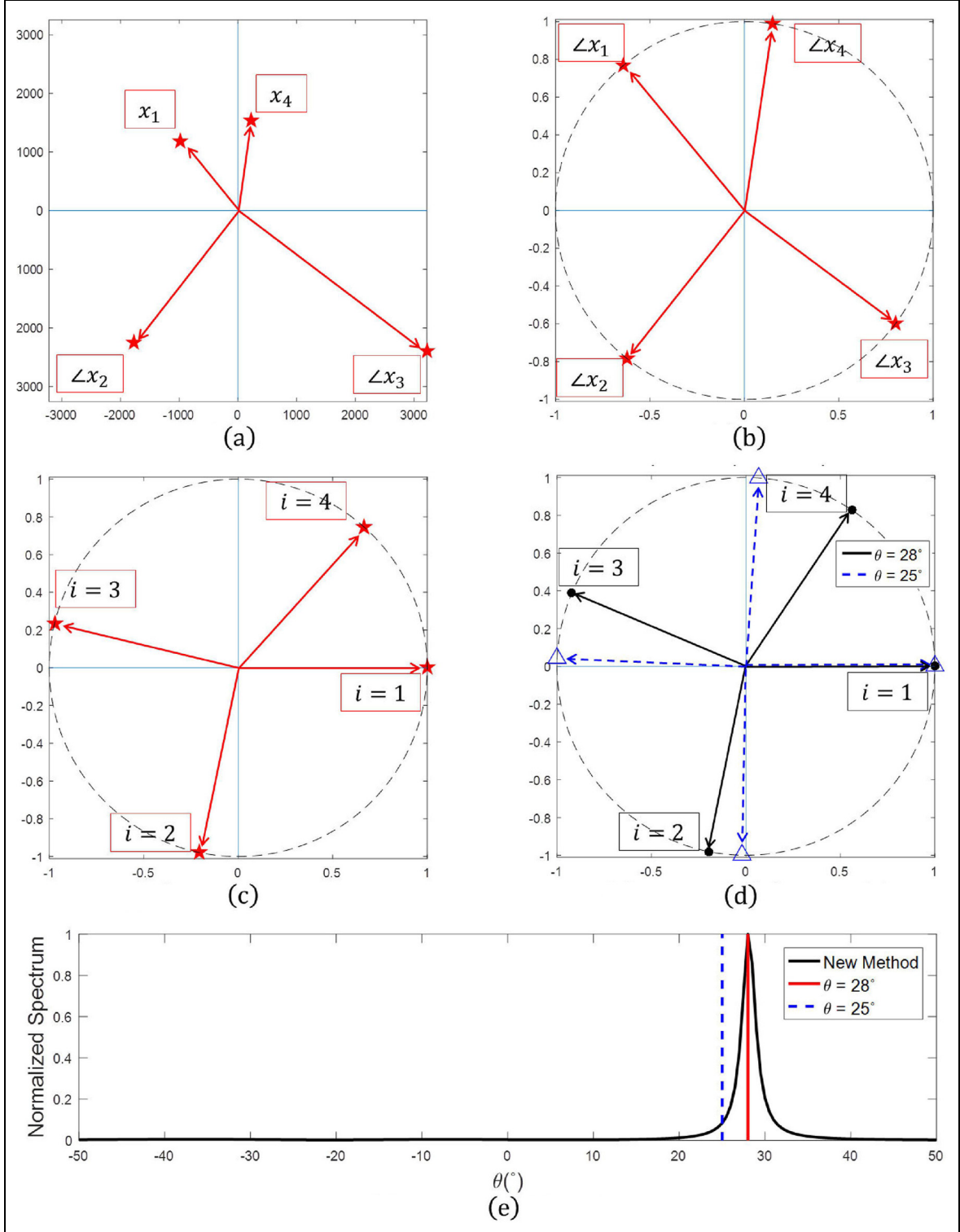
The equation for kurtosis is expressed as follows

$$Kurt[X] = \frac{\mu_4}{\sigma^4} = \frac{E[(X - \mu)^4]}{E[(X - \mu)^2]^2} \quad (37)$$

where  $\mu$  denotes the average of  $X$ ,  $\sigma$  denotes the standard deviation of  $X$ , and  $E[\cdot]$  denotes the expectation operator.

The kurtosis of normal distribution is three, and it approaches zero as the distribution becomes more even. When the distribution becomes biased, the kurtosis becomes larger than three. Thus, the kurtosis value can be used to determine whether the distribution is even or biased. The greater the kurtosis value is, the higher the resolution becomes.

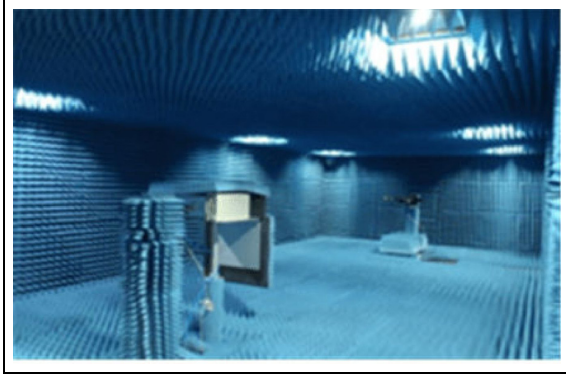
Figure 7 shows the kurtosis values for the DOA estimation results of each algorithm. To apply these



**Figure 3.** Input signal phase difference ( $\phi_{i,opt}$ ) and steering vector phase difference ( $\phi_i$ ): (a)  $X$ , (b)  $\angle X$ , (c)  $\phi_{i,opt} = \angle x_1 - \angle x_i$ , (d)  $\phi_{i,} = \angle a_1(\theta) - \angle a_i(\theta)$ , and (e) normalized spectrum.

kurtosis values, only the values within  $\pm 20^\circ$  from the angle that has the maximum value in the normalized spectrum were used. The results confirmed that the

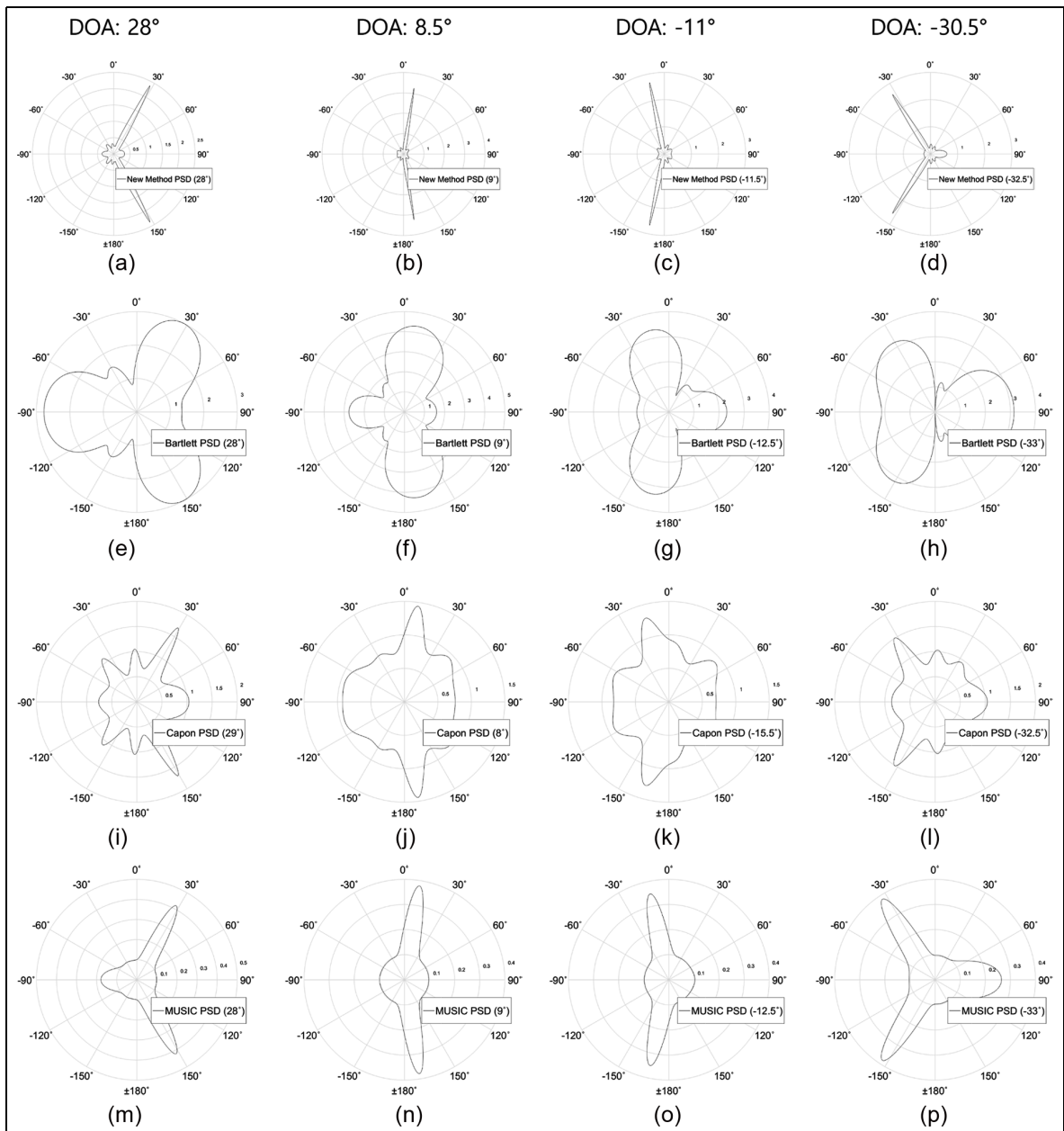
new method has more main lobes estimated as the DOA and almost no side lobe compared to the conventional algorithms.



**Figure 4.** Radar chamber.

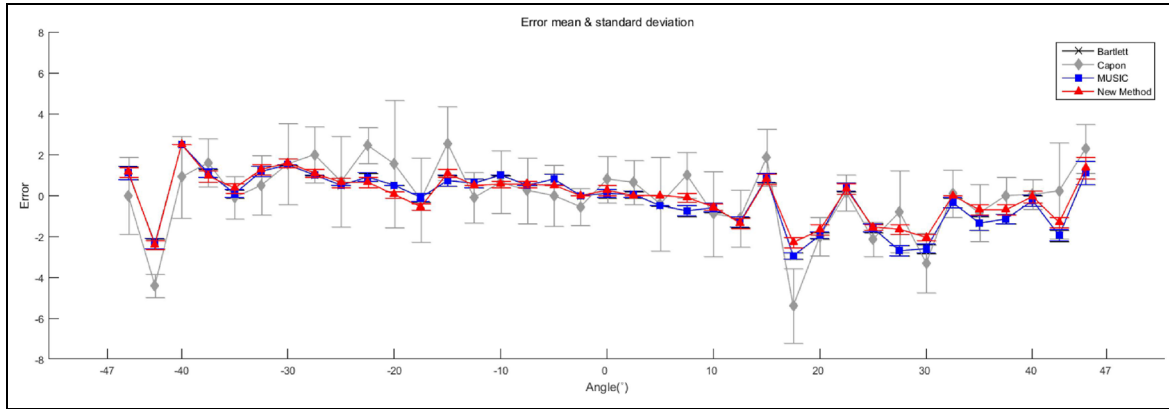
Figure 8 shows the average kurtosis value for each error when there is an error between the estimated angle and actual angle of the DOA for each algorithm. This result confirms that the DOA estimation error of the new method is smaller than those of the conventional algorithms and the average of kurtosis values is very large when the DOA estimation has an error.

Table 2 shows the average of kurtosis values for each algorithm for all data. The average kurtosis values of the conventional algorithms are smaller than three, indicating that they are more even than the normal distribution. The average kurtosis of the proposed value is very large at 31.9140. This suggests that only the main



**Figure 5.** Log-scaled spectrum (actualDOA :  $28^\circ$ ,  $8.5^\circ$ ,  $-11^\circ$ ,  $-30.5^\circ$ ): (a)–(d) New method results, (e)–(h) Bartlett results, (i)–(l) Capon results, and (m)–(p) MUSIC results.



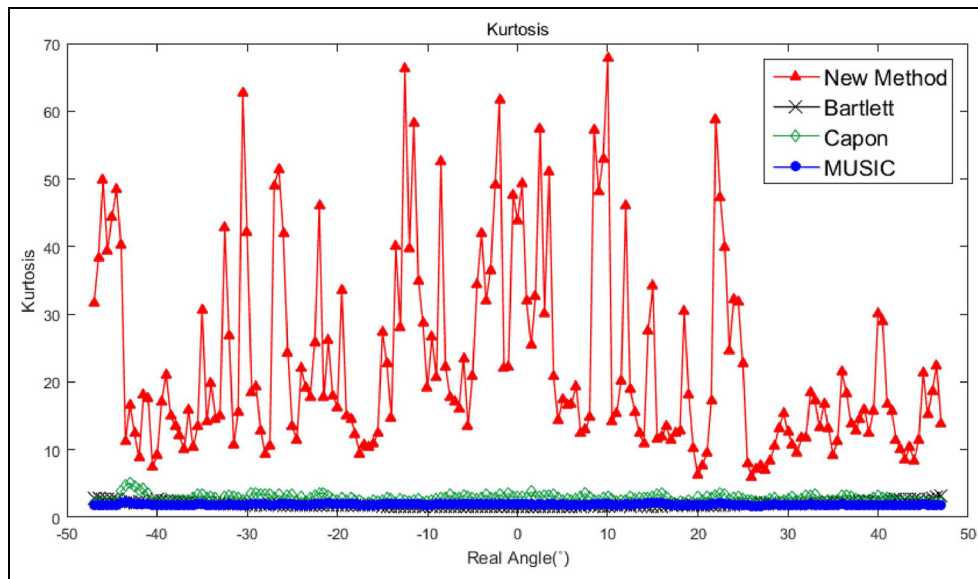


**Figure 6.** Average and standard deviation of DOA estimation error ( $-47^\circ \sim 47^\circ$ ).

**Table 1.** Average error and average standard deviation in 1952 scan data ( $-47^\circ \sim 47^\circ$ ) (mean  $\pm$  SD).

	New method	Bartlett	Capon	MUSIC
No noise ( $^\circ$ )	$0.8582 \pm 0.1909$	$1.0733 \pm 0.1886$	$1.6470 \pm 1.4637$	$1.0733 \pm 0.1886$

MUSIC: multiple signal classification.



**Figure 7.** Average kurtosis in each angle.

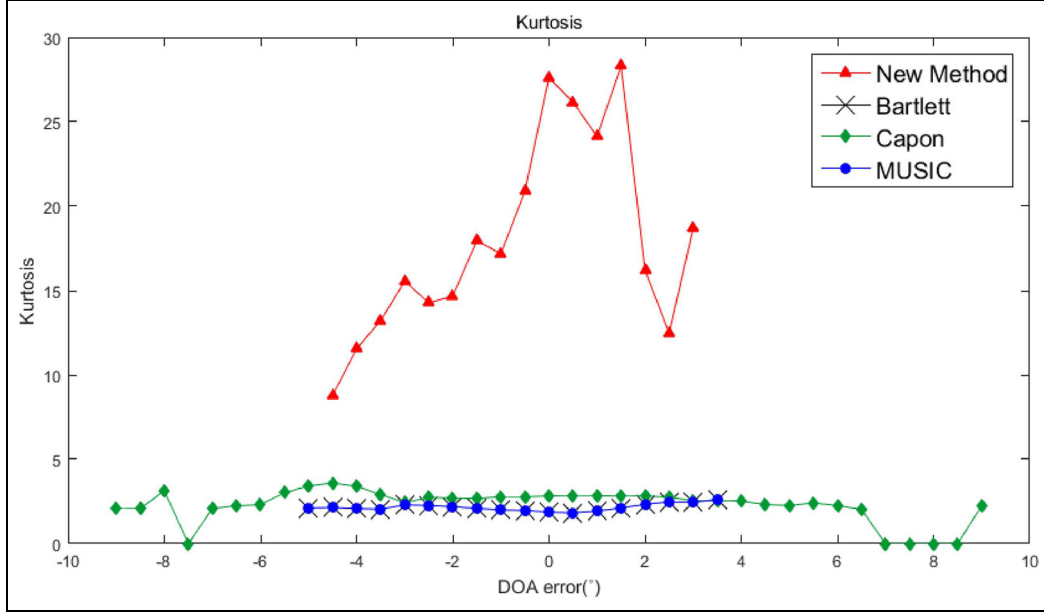
lobes with a peak value have large values and side lobes have small values, resulting in a biased distribution of the spectrum. This result confirms that the new method of DOA estimation has the best performance with the highest resolution.

The above results verified that the new method shows better performance in the accuracy of DOA estimation compared to the conventional algorithms.

Figure 9 shows the normalized spectrum of the new method, Bartlett, Capon, and MUSIC algorithms with

respect to actual DOA of  $28^\circ$  when SNR is set by 20, 10, 5, and 0 dB. As SNR decreases, we can observe that spectrum of the side lobes increases. Also Figure 10 provides the average DOA estimation error for various SNRs with respect to actual DOA from  $-47^\circ$  to  $47^\circ$ , and Table 3 summarizes the average and the standard deviation of the DOA estimation errors for the new and conventional methods. For all DOA estimation algorithms, DOA estimation error increases as SNR decreases. The average DOA estimation error and





**Figure 8.** Average kurtosis in each DOA estimation error.

**Table 2.** Average kurtosis in 1952 scan data ( $-47^\circ \sim 47^\circ$ ).

	New method	Bartlett	Capon	MUSIC
Kurtosis	31.9140	1.9947	2.7911	1.8776

MUSIC: multiple signal classification.

**Table 3.** Average DOA estimation error and standard deviation for various SNRs (mean  $\pm$  SD).

	20 dB	10 dB	5 dB	0 dB
New method	0.9754 $\pm$ 0.6321	1.8455 $\pm$ 2.0011	3.7953 $\pm$ 5.5305	8.2554 $\pm$ 13.5486
Bartlett	1.1752 $\pm$ 0.6185	1.8945 $\pm$ 1.8851	3.4987 $\pm$ 4.8217	7.3176 $\pm$ 12.1263
Capon	1.3814 $\pm$ 1.3165	2.1230 $\pm$ 2.5839	7.0986 $\pm$ 11.4869	16.0756 $\pm$ 22.9450
MUSIC	1.1752 $\pm$ 0.6185	1.8945 $\pm$ 1.8851	3.4987 $\pm$ 4.8217	7.3176 $\pm$ 12.1263

SNR: signal-to-noise ratio; SD: standard deviation; MUSIC: multiple signal classification.

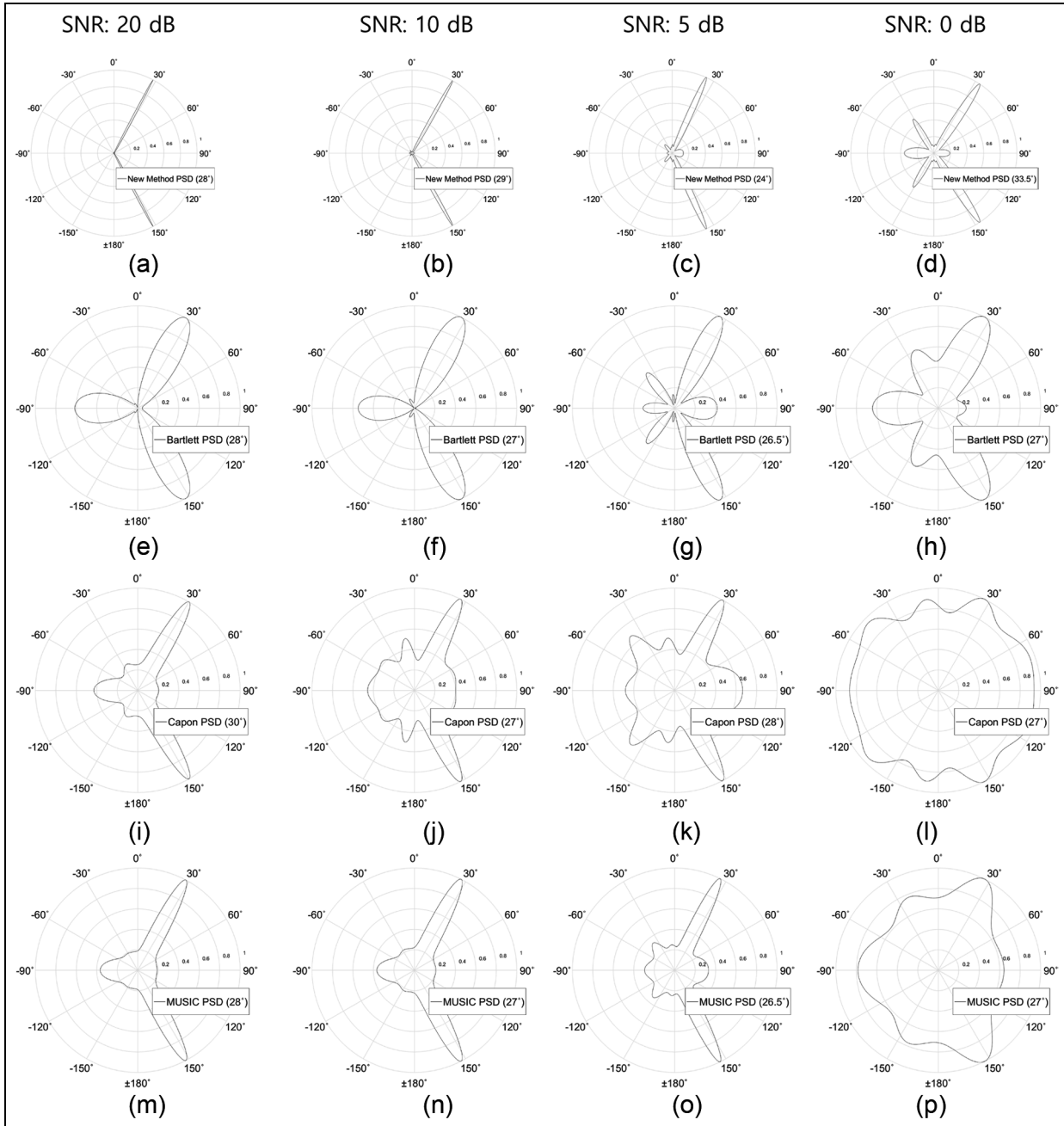
standard deviation of Capon is relatively large compared to Bartlett and MUSIC. Although there are some differences, the new method has the results that are almost similar to Bartlett and MUSIC results as SNR changes.

To verify the performance of the DOA estimation under multiple target conditions, we conducted the DOA estimation using the long-range radar (LRR) signal acquired from the chamber, normal road, and the parking lot. The number of sensors in an LRR sensor array was 8 and the sensor gap was  $1.8\lambda$ . In the case of the chamber data, the number of targets was set to two, and Figure 11(a) shows the DOA estimation results. Figure 11(b)–(e) shows the results for the real data

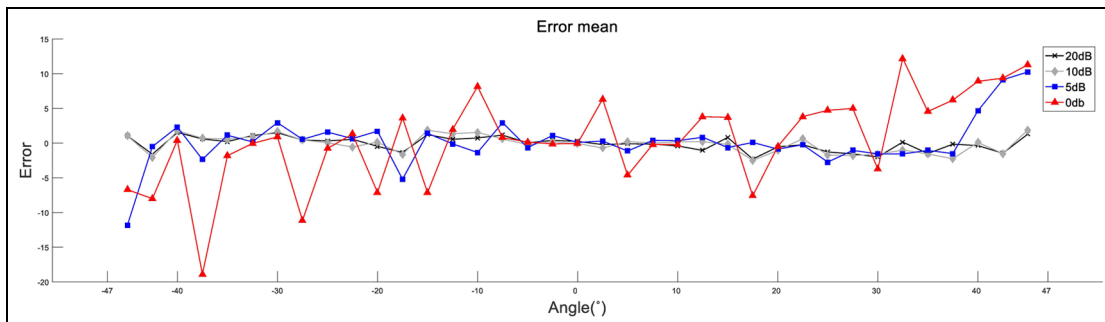
under multiple target conditions. The DOA estimation results of conventional algorithms show that spectrums of specific object are too small to be estimated, and the two close objects are not classified due to low resolution. However, the new method estimates the three targets at least or more and has better resolution than the conventional algorithms.

## Conclusion

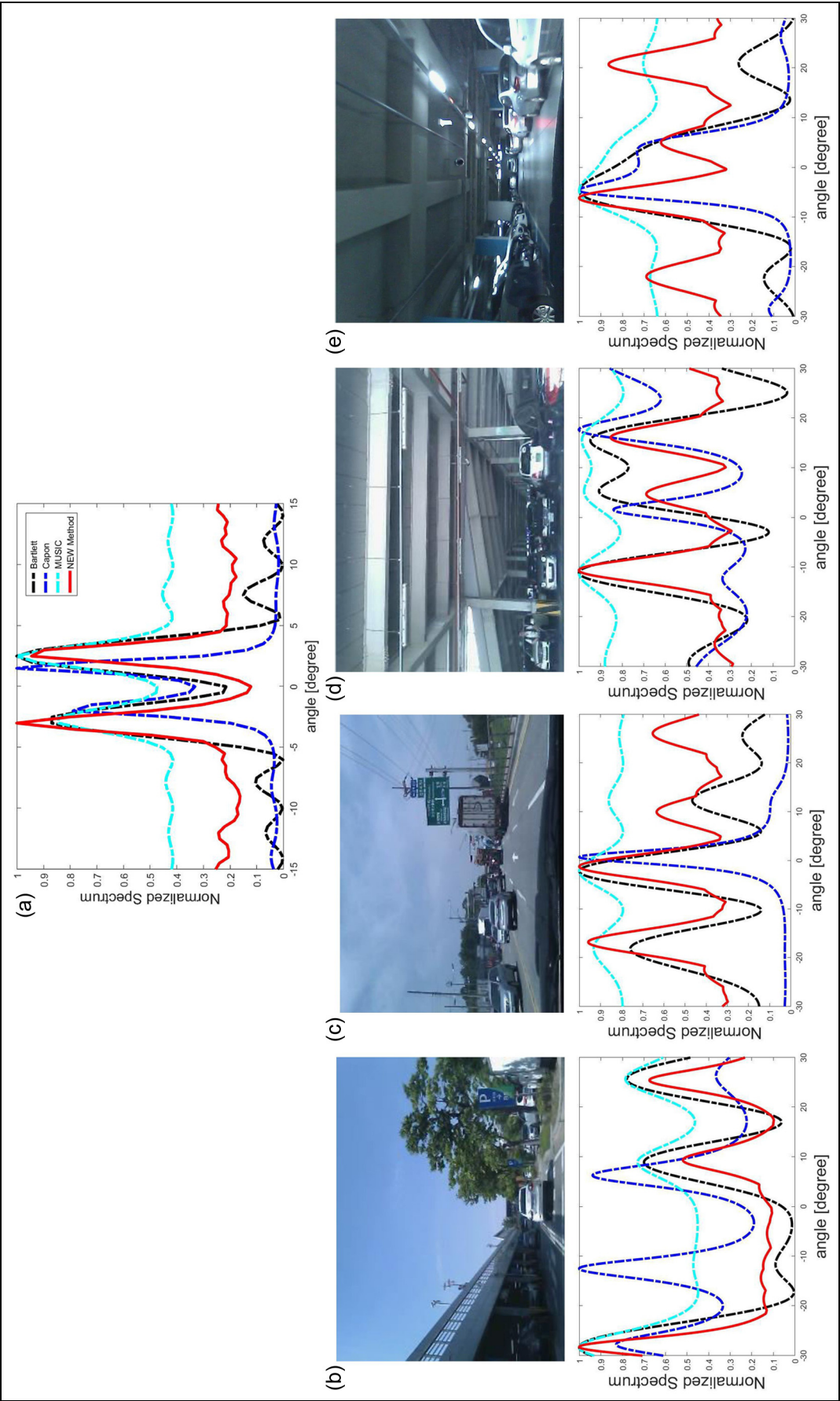
This article proposed a new DOA estimation method for more accurate estimation compared to the conventional DOA estimation algorithms. To compare the estimation error, the difference between the angle of



**Figure 9.** Normalized spectrum when SNR is set to 20, 10, 5, and 0 dB (actual DOA: 28°): (a)–(d) New method results, (e)–(h) Bartlett results, (i)–(l) Capon results, and (m)–(p) MUSIC results.



**Figure 10.** Average DOA estimation error of the new method when SNR is set to 20, 10, 5, and 0 dB (−47°~47°).



**Figure 11.** Normalized spectrum of multiple targets in chamber and actual road conditions: (a) result in chamber, and (b)–(e) results in actual conditions.

