Beamforming when the sound velocity is not precisely known

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Beamforming is an integral part of most signal processing systems in active or passive sonars. The delays used to generate a beam are functions of the sound velocity, which depends on temperature, salinity, and pressure. There is a loss in array gain if the delays are incorrectly set. This will occur when the sound velocity in the water surrounding the hydrophones is different from the velocity that was used to set the delays. This paper makes two points: (1) fixed delay line sonars suffer a loss in gain when the true sound speed in the water is different from the velocity that is used to set the delays, and (2) there are signal processing techniques for two- or three-dimensional arrays that yield source bearings that are independent of the true sound velocity. These techniques require variable time delays, which can be realized using digital processing.

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INTRODUCTION

Beamforming is an integral part of most signal processing systems in active or passive sonars, and is part of the processing used to estimate target bearings. The delays used to generate a beam are functions of the sound velocity, which depends on temperature, salinity, and pressure. Beamforming filters the received signals in "velocity space".¹ The filter is detuned, and the beam pattern distorted, if the delays are incorrectly set.² This will occur when the sound velocity in the water surrounding the hydrophones is different from the velocity that was used to set the delays. Current fixed delay line sonars do not adapt to changes in sound velocity. Modern hybrid electronic beamformers, on the other hand, can be made to adapt to changes in velocity if temperature and salinity are monitored. This paper makes two points: (1) fixed delay line sonars suffer a loss in gain when the true sound speed in the water is different from the velocity that is used to set the delays, and (2) there are signal processing techniques for twoor three-dimensional arrays that yield source bearings that are independent of the true sound velocity. These techniques require variable time delays, which can be realized using digital processing. The first part of this paper presents some insight to the distortion that is caused when the delays are not correctly set as a function of the true velocity. The second part presents a signal processing method for bearing estimation that is independent of the true sound velocity.

I. THE RESPONSE OF A DETUNED BEAMFORMER

To simplify exposition, consider a horizontal planar array of K hydrophones. Let (x_k, y_k) denote the coordinates of the kth sensor (k = 1, ..., K) with respect to a coordinate system whose origin is at the array's center. Let θ_0 denote the bearing of a distant source with respect to the x axis (Fig. 1). Assume that the source is at the same depth as the array, so that the source from the source propagates across the array as a plane wave with velocity c_0 , the sound velocity in the water. To compute the response, let the source be a single frequency tone with unit amplitude. If the delays are set using a velocity c, the beamformer's power response for a look angle θ is given by

$$P(\theta_0, \theta) = \left| \sum_{k=1}^{K} \exp i\omega_0 \right| \times \left(\frac{x_k \cos \theta}{c} + \frac{y_k \sin \theta}{c} - \frac{x_k \cos \theta_0}{c_0} - \frac{y_k \sin \theta_0}{c_0} \right) \right|^2, \quad (1)$$

where ω_0 is the angular frequency of the source.

If $c = c_0$, then $P(\theta_0, \theta_0) = K^2$. If $c \neq c_0$, it follows from (1) that $P(\theta_0, \theta_0) < K^2$, i.e., the response of a detuned beamformer for $\theta = \theta_0$ is less than the response of a tuned one.

The loss of gain for the array geometry of one of our fleet sonars cannot be presented in an unclassified paper. To obtain some insight to the loss in gain when $c \neq c_0$, suppose that the array geometry is a square lattice with $K = N^2$ sensors whose interelement spacing is d, i.e., the sensor positions are $\{jd, ld\}$, where j, l = 0, $\pm 1, \ldots, \pm (N-1)/2$ (assuming N odd). From (1),

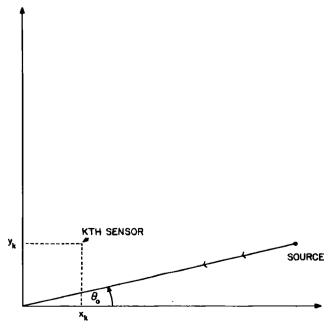
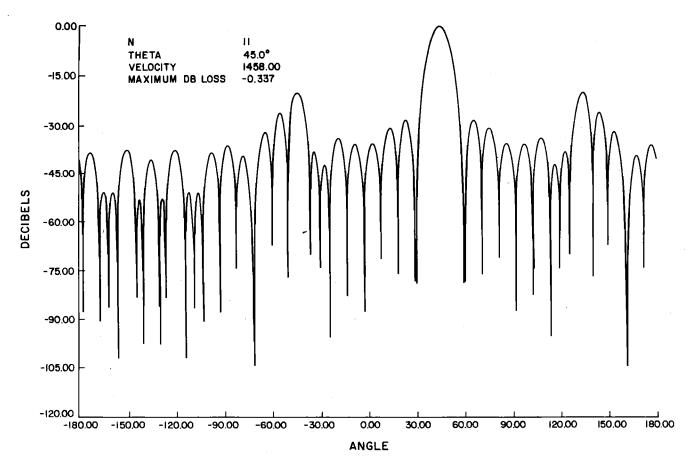
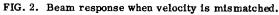


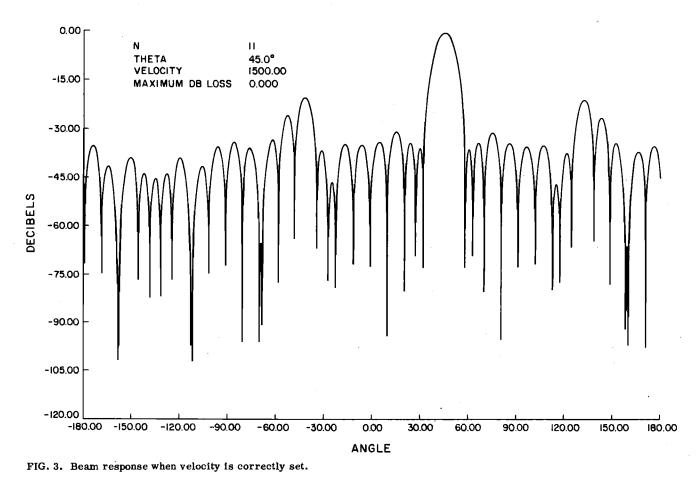
FIG. 1. Array coordinate system.

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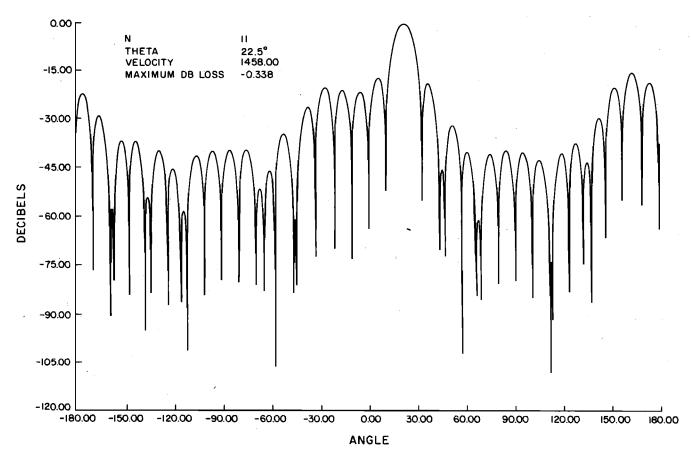


FIG. 4. Mismatched beam response for a direction of 22.5 degrees.

$$P(\theta_{0},\theta) = \frac{\sin^{2}[\omega_{0}(c^{-1}\cos\theta - c_{0}^{-1}\cos\theta_{0})(Nd/2)]}{\sin^{2}[\omega_{0}(c^{-1}\cos\theta - c_{0}^{-1}\cos\theta_{0})(d/2)]} \\ \times \frac{\sin^{2}[\omega_{0}(c^{-1}\sin\theta - c_{0}^{-1}\sin\theta_{0})(Nd/2)]}{\sin^{2}[\omega_{0}(c^{-1}\sin\theta - c_{0}^{-1}\sin\theta_{0})(d/2)]}, \quad (2)$$

$$= \frac{\sin^{2}[\pi(c^{-1}c_{0}\cos\theta - \cos\theta_{0})Nd\lambda_{0}^{-1}]}{\sin^{2}[\pi(c^{-1}c_{0}\cos\theta - \cos\theta_{0})Nd\lambda_{0}^{-1}]} \\ \times \frac{\sin^{2}[\pi(c^{-1}c_{0}\sin\theta - \sin\theta_{0})Nd\lambda_{0}^{-1}]}{\sin^{2}[\pi(c^{-1}c_{0}\sin\theta - \sin\theta_{0})A\lambda_{0}^{-1}]},$$

where λ_0 is the source wavelength. Thus

$$P(\theta_{0},\theta_{0}) = \frac{\sin^{2}[\pi(c^{-1}c_{0}-1)Nd\lambda_{0}^{-1}\cos\theta_{0}]}{\sin^{2}[\pi(c^{-1}c_{0}-1)d\lambda_{0}^{-1}\cos\theta_{0}]} \times \frac{\sin^{2}[\pi(c^{-1}c_{0}-1)Nd\lambda_{0}^{-1}\sin\theta_{0}]}{\sin^{2}[\pi(c^{-1}c_{0}-1)d\lambda_{0}^{-1}\sin\theta_{0}]}.$$
(3)

The larger the array aperture, the greater the loss in gain when $c \neq c_0$. This implies that a large aperture linear array can be badly detuned if c is not very close to c_0 .

As an example, let $d\lambda_0^{-1} = 1/2$, N = 19, and c = 1500 m/s. If the water temperature is 2°C, then $c_0 = 1458$ m/s.³ From (3), $P(\theta_0, \theta_0)/N^4 = 0.79$ for $\theta_0 = \pi/4$. Since $P(\theta_0, \theta_0) = N^4$ if $c = c_0$, the loss of power is 1 dB. The loss is also 1 dB for $\theta_0 = 0$, $\pi/2$ and is 1.1 dB for $\theta_0 = \pi/8$.

In addition to the loss in gain at $\theta = \theta_0$, the response is distorted. To illustrate the distortion, I plotted expression (2) as a function of θ_0 for $\theta = \pi/8$ and $\pi/4$ using N = 11, $c_0 = 1458$ m/s, c = 1500 m/s, and $d = \lambda_0/2$. The plot of $P(\theta_0, \pi/4)$ is given in Fig. 2. As a comparison, $P(\theta_0, \pi/4)$ for $c_0 = c = 1500$ m/s is given in Fig. 3. The sidelobes of the distorted response are higher than those of the undistorted response. The peak of the response is at $\theta_0 = 45^{\circ}$, but with a loss or gain of 0.34 dB. These results are consistent with the results of Quazi and Nuttall² for random phasing errors.

The distorted response for $\theta = \pi/8$ is shown in Fig. 4. The peak to sidelobe ratio is lower than that for $\theta = \pi/4$. The peak is at $\theta_0 = 22.5^\circ$. These results also hold when N = 19, where the loss of gain at $\theta_0 = \theta$ is higher.

II. ESTIMATING BEARING INDEPENDENTLY OF VELOCITY

Delay-and-sum processing of a planar array can be adapted to yield a bearing estimator that is independent of c_0 . Let $p_k(t)$ denote the output from the kth sensor. Assume that $p_k(t)$ is a plane wave $s(t - \alpha_0 x_k - \beta_0 y_k)$ plus stationary Gaussian noise, where $\alpha_0 = c_0^{-1} \cos\theta_0$ and β_0 $= c_0^{-1} \sin\theta_0$. Sampling the channels at times t_1, \ldots, t_N , define the least-squares measure

$$C(\alpha,\beta) = \sum_{j=1}^{N} \left| \sum_{k=1}^{K} p(t_j + \alpha x_k + \beta y_k) \right|^2, \qquad (4)$$

where α and β are time delays per unit distances. Let $\hat{\alpha}$ and $\hat{\beta}$ be values that jointly maximize $C(\alpha, \beta)$. Levin⁴ shows that, ignoring end effects, $\hat{\alpha}$ and $\hat{\beta}$ are maximum likelihood estimators of α_0 and β_0 . Using the frequency-wavenumber approach to array processing, Hinich and

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Shaman⁵ derive the statistical properties of maximum likelihood estimators.

Given $\hat{\alpha}$ and $\hat{\beta}$, the maximum likelihood estimator of θ_0 is $\hat{\theta} = \tan^{-1}(\hat{\beta}/\hat{\alpha})$.⁶ This estimator does not use c_0 . A related robust method is given by Bennett.⁷

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