Multiple and single snapshot compressive beamforming

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For a sound field observed on a sensor array, compressive sensing (CS) reconstructs the direction of arrival (DOA) of multiple sources using a sparsity constraint. The DOA estimation is posed as an underdetermined problem by expressing the acoustic pressure at each sensor as a phase-lagged superposition of source amplitudes at all hypothetical DOAs. Regularizing with an ℓ_1 -norm constraint renders the problem solvable with convex optimization, and promoting sparsity gives high-resolution DOA maps. Here the sparse source distribution is derived using maximum *a posteriori* estimates for both single and multiple snapshots. CS does not require inversion of the data covariance matrix and thus works well even for a single snapshot where it gives higher resolution than conventional beamforming. For multiple snapshots, CS outperforms conventional high-resolution methods even with coherent arrivals and at low signal-to-noise ratio. The superior resolution of CS is demonstrated with vertical array data from the SWellEx96 experiment for coherent multi-paths. © 2015 Acoustical Society of America. [http://dx.doi.org/10.1121/1.4929941]

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I. INTRODUCTION

Direction-of-arrival (DOA) estimation refers to the localization of several sources from noisy measurements of the wavefield with an array of sensors. DOA estimation can be expressed as a linear underdetermined problem with a sparsity constraint enforced on its solution. The compressive sensing^{1,2} (CS) framework asserts that this is solved efficiently with a convex optimization procedure that promotes sparse solutions.

In DOA estimation, CS achieves high-resolution acoustic imaging,^{3–5} outperforming traditional methods.⁶ Unlike the high-resolution subspace-based DOA estimators,^{7,8} DOA estimation via CS is reliable even with a single snapshot.^{9–11}

The least absolute shrinkage and selection operator $(LASSO)^{12}$ has been extended to multiple measurement vectors (here multiple snapshots).^{3,13} They modify the LASSO objective function by introducing a mixed-norm penalty term that promotes spatial sparsity. More specifically, the snapshots are combined with the ℓ_2 -norm, whereas the spatial samples are combined with the ℓ_1 -norm. Multiple-snapshot CS offers several benefits over other high-resolution DOA estimators,^{3,4,13} (1) it handles partially coherent arrivals; (2) it can be formulated with any number of snapshots, in contrast to, e.g., the minimum variance distortion-free response (MVDR) beamformer. (3) Its flexibility in formulation enables extensions to sequential

processing, and online algorithms.¹⁰ Here we show that CS achieves higher resolution than MUSIC and MVDR even in scenarios that favor these classical high-resolution methods.

In ocean acoustics, CS has found several applications in matched field processing^{14,15} and in coherent passive fathometry for inferring sediment interfaces depths and their number.¹⁶ Various wave propagation phenomena from a single source (refraction, diffraction, scattering, ducting, reflection) lead to multiple partially coherent arrivals received by the array. High-resolution beamformers cannot resolve these coherent arrivals.

CS for single snapshot has high-resolution capabilities and contrary to eigenvalue-based beamformers works for coherent arrivals.^{3,4,11} CS is limited by basis mismatch,¹⁷ which occurs when the DOAs do not coincide with the look directions of the angular spectrum and by basis coherence. Solutions to basis mismatch involve, for example, using atomic norm and solving the dual problem,^{5,18,19} which are not addressed here. Grid refinement alleviates basis mismatch for high signal-to-noise ratio (SNR) at the expense of increased computational complexity. A denser grid causes increased coherence among the steering vectors (basis coherence); this translates to bias and spread in the DOA estimates as demonstrated here. This is especially true in large two- or three-dimensional geo-acoustic inversion problems as, e.g., seismic imaging.^{20–22}

We use least squares optimization with an ℓ_1 -norm regularization term, also known as the LASSO,¹² to formulate the DOA estimation problem for single and multiple

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snapshots. The LASSO formulation complies with statistical models as it provides a maximum *a posteriori* (MAP) estimate, assuming a Gaussian data likelihood and a Laplacian prior distribution for the source acoustic pressure^{23,24} for both single (Sec. II B) and multiple snapshots¹³ (Sec. III). The LASSO is known to be a convex minimization problem and solved efficiently by interior point methods. In the LASSO formulation, Sec. IV A, the reconstruction accuracy depends on the choice of the regularization parameter that controls the balance between the data fit and the sparsity of the solution. We indicate that the regularization parameter can be found from the properties of the LASSO path,^{25,26} i.e., the evolution of the LASSO solution versus the regularization parameter.

The main focus of the paper is on performance evaluation for single and multiple snap-shots using both simulated (Sec. V) and real data (Sec. VI). Other excellent papers¹¹ have already performed performance evaluation for single snapshot, consistent with our simulations. We are not aware of performance evaluation for multiple snapshots.

In the following, the ℓ_p -norm of a vector $\mathbf{x} \in \mathbb{C}^N$ is defined as $\|\mathbf{x}\|_p = (\sum_{n=1}^N |x_n|^p)^{1/p}$. By extension, the ℓ_0 -norm is defined as $\|\mathbf{x}\|_0 = \sum_{n=1}^N 1_{x_n \neq 0}$ and the ℓ_∞ -norm as $\|\mathbf{x}\|_{\infty} = \max_{1 \le n \le N} |x_n|$. For a matrix $\mathbf{F} \in \mathbb{C}^{M \times L}$ the Frobenius norm $\|\cdot\|_{\mathcal{F}}$ is defined as $\|\mathbf{F}\|_{\mathcal{F}}^2 = \sum_{i=1}^M \sum_{j=1}^L |f_{i,j}|^2$.

II. SINGLE SNAPSHOT DOA ESTIMATION

We assume plane wave propagation and narrowband processing with a known sound speed. We consider the onedimensional problem with a uniform linear array (ULA) of sensors with the source location characterized by the DOA of the associated plane wave, $\theta \in [-90^\circ, 90^\circ]$, with respect to the array axis. The propagation delay from the *i*th potential source to each of the *M* array sensors is described by the steering (or replica) vector,

$$\mathbf{a}(\theta_i) = \frac{1}{\sqrt{M}} \left[1, e^{j(2\pi d/\lambda) 1 \sin \theta_i}, \dots, e^{j(2\pi d/\lambda)(M-1) \sin \theta_i} \right]^T,$$
(1)

where λ is the wavelength and d the sensor spacing.

Discretizing the half-space of interest, $\theta \in [-90^\circ, 90^\circ]$, into *N* angular directions the DOA estimation problem can be expressed as a source reconstruction problem with the linear model,

$$\mathbf{y} = \mathbf{A}\mathbf{x} + \mathbf{n},\tag{2}$$

where $\mathbf{y} \in \mathbb{C}^M$ is the complex-valued data vector from the measurements at the *M* sensors, $\mathbf{x} \in \mathbb{C}^N$ is the unknown vector of the complex source amplitudes at all *N* directions on the angular grid of interest, and $\mathbf{n} \in \mathbb{C}^M$ is the additive noise vector. The sensing matrix,

$$\mathbf{A} = [\mathbf{a}(\theta_1), \dots, \mathbf{a}(\theta_N)], \tag{3}$$

maps the hypothetical sources x to the observations y and has as columns the steering vectors, Eq. (1), at all look directions.

In the following, the noise is generated as independent and identically distributed (iid) complex Gaussian. The array signal-to-noise ratio (SNR) is defined as

SNR =
$$10 \log_{10} \frac{E\{\|\mathbf{A}\mathbf{x}\|_{2}^{2}\}}{E\{\|\mathbf{n}\|_{2}^{2}\}} (dB).$$
 (4)

A. Sparse reconstruction with CS

The problem of DOA estimation is to recover the set of non-zero components in the source vector $\mathbf{x} \in \mathbb{C}^N$, given the sensing matrix $\mathbf{A}_{M \times N}$ and an observation vector $\mathbf{y} \in \mathbb{C}^M$. Even though there are only a few sources K < M generating the acoustic field, we are interested in a fine resolution on the angular grid to achieve precise localization such that $M \ll N$ and the problem in Eq. (2) is underdetermined. A way to solve this ill-posed inverse problem is constraining the possible solutions with prior information.

Traditional methods solve the problem in Eq. (2) by seeking the solution with the minimum ℓ_2 -norm, which provides the best data fit (ℓ_2 -norm regularized least squares),

$$\hat{\mathbf{x}}_{\ell_2}(\mu) = \operatorname*{argmin}_{\mathbf{x}\in\mathbb{C}^N} \|\mathbf{y} - \mathbf{A}\mathbf{x}\|_2^2 + \mu \|\mathbf{x}\|_2^2.$$
(5)

The regularization parameter, $\mu \ge 0$, controls the relative importance between the data fit and the ℓ_2 -norm of the solution. The minimization problem in Eq. (5) is convex with analytic solution, $\hat{\mathbf{x}}_{\ell_2}(\mu) = \mathbf{A}^H (\mathbf{A}\mathbf{A}^H + \mu \mathbf{I}_M)^{-1} \mathbf{y}$, where \mathbf{I}_M is the $M \times M$ identity matrix. However, it aims to minimize the energy of the source \mathbf{x} through the ℓ_2 -norm regularization term rather than its sparsity, hence the resulting solution is non-sparse.

Conventional beamforming (CBF)⁷ is related to the ℓ_2 solution for large μ . From Eq. (5),

$$\hat{\mathbf{x}}_{\text{CBF}} = \lim_{\mu \to \infty} (\mu \hat{\mathbf{x}}_{\ell_2}(\mu)) = \mathbf{A}^H \mathbf{y}.$$
(6)

In principle, CBF combines the sensor outputs coherently to enhance the source signal at a specific look direction from the ubiquitous noise. CBF is robust to noise but suffers from low resolution and the presence of sidelobes.

Because **x** is sparse (there are only $K \ll N$ sources), it is appropriate to seek for the solution with the minimum ℓ_0 norm, which counts the number of non-zero entries in the vector, to find a sparse solution. However, the ℓ_0 -norm minimization problem is a non-convex combinatorial problem that becomes computationally intractable even for moderate dimensions. The breakthrough of CS^{1,2} came with the proof that for sufficiently sparse signals, $K \ll N$, and sensing matrices with sufficiently incoherent columns, the ℓ_0 -norm minimization problem is equivalent (at least in the noiseless case) to its convex relaxation, the ℓ_1 -norm minimization problem.^{27,28} By replacing the ℓ_0 -norm with the convex

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 ℓ_1 -norm, the problem can be solved efficiently with convex optimization even for large dimensions.^{29–31}

For noisy measurements, Eq. (2), the ℓ_1 -norm minimization problem is formulated as

$$\hat{\mathbf{x}}_{\ell_1}(\epsilon) = \underset{\mathbf{x} \in \mathbb{C}^N}{\operatorname{argmin}} \|\mathbf{x}\|_1 \text{ subject to } \|\mathbf{y} - \mathbf{A}\mathbf{x}\|_2 \le \epsilon,$$
(7)

where ϵ is the noise floor. The estimate $\hat{\mathbf{x}}_{\ell_1}(\epsilon)$ has the minimum ℓ_1 -norm while it fits the data up to the noise level. The problem in Eq. (7) can be equivalently written in an unconstrained form with the use of the regularizer $\mu \ge 0$,

$$\hat{\mathbf{x}}_{\ell_1}(\mu) = \operatorname*{argmin}_{\mathbf{x}\in\mathbb{C}^N} \|\mathbf{y} - \mathbf{A}\mathbf{x}\|_2^2 + \mu \|\mathbf{x}\|_1.$$
(8)

The sparse source reconstruction problem in Eq. (8) is a least squares optimization method regularized with the ℓ_1 -norm of the solution **x** and provides the best data fit (ℓ_2 -norm term) for the sparsity level determined by the regularization parameter μ . The optimization problem in Eq. (8) is also known as the least absolute shrinkage and selection operator (LASSO) because the ℓ_1 regularizer shrinks the coefficients of **x** toward zero as the regularization parameter μ increases.¹² This is illustrated in Fig. 1. For every ϵ , there exists a μ so that the estimates in Eqs. (7) and (8) are equal.

Once the active DOAs are recovered, by solving Eq. (7) or equivalently Eq. (8), the unbiased complex source amplitudes are determined from

$$\hat{\mathbf{x}}_{\rm CS} = \mathbf{A}_{\mathcal{M}}^+ \mathbf{y},\tag{9}$$

where $\mathbf{A}_{\mathcal{M}} \in \mathbb{C}^{N \times K}$ contains only the "active" steering vectors associated with non-zero components in the solution $\hat{\mathbf{x}}_{\ell_1}(\mu)$ and $\mathbf{A}_{\mathcal{M}}^+$ is its Moore–Penrose pseudoinverse.

For a given sparsity level *K* and corresponding set of active indexes \mathcal{M} , i.e., $|\mathcal{M}| = K$, Eq. (9) finds the best data fit. Thus if the active sensing matrix $\mathbf{A}_{\mathcal{M}}$ has sufficiently incoherent columns it represents the solution to the ℓ_0 problem

$$\hat{\mathbf{x}}_{\ell_0}(K) = \underset{\mathbf{x}\in\mathbb{C}^N}{\operatorname{argmin}} \|\mathbf{y} - \mathbf{A}\mathbf{x}\|_2 \text{ subject to } \|\mathbf{x}\|_0 = K.$$
(10)

B. MAP estimate via LASSO

We use the LASSO formulation, Eq. (8), to solve the DOA estimation problem in favor of sparse solutions. The

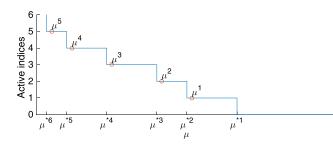


FIG. 1. (Color online) Illustration of the LASSO path: Number of active indices versus the regularization parameter μ . Increments in the active set occur at μ^{*p} .

choice of the (unconstrained) LASSO formulation over the constrained formulation, Eq. (7), allows the sparse reconstruction method to be interpreted in a statistical Bayesian setting, where the unknowns \mathbf{x} and the observations \mathbf{y} are both treated as stochastic (random) processes by imposing a prior distribution on the solution \mathbf{x} , which promotes sparsity.^{12,23,24}

Bayes theorem connects the posterior distribution $p(\mathbf{x}|\mathbf{y})$ of the model parameters \mathbf{x} conditioned on the data \mathbf{y} with the data likelihood $p(\mathbf{y}|\mathbf{x})$, the prior distribution of the model parameters $p(\mathbf{x})$ and the marginal distribution of the data $p(\mathbf{y})$,

$$p(\mathbf{x}|\mathbf{y}) = \frac{p(\mathbf{y}|\mathbf{x})p(\mathbf{x})}{p(\mathbf{y})}.$$
(11)

Then the maximum a posteriori (MAP) estimate is

$$\hat{\mathbf{x}}_{\text{MAP}} = \underset{\mathbf{x}}{\operatorname{argmax}} \ln p(\mathbf{x}|\mathbf{y})$$

$$= \underset{\mathbf{x}}{\operatorname{argmax}} [\ln p(\mathbf{y}|\mathbf{x}) + \ln p(\mathbf{x})]$$

$$= \underset{\mathbf{x}}{\operatorname{argmin}} [-\ln p(\mathbf{y}|\mathbf{x}) - \ln p(\mathbf{x})], \quad (12)$$

where the marginal distribution of the data $p(\mathbf{y})$ is omitted because it is independent of the model \mathbf{x} .

Based on a complex Gaussian noise model with independent and identically distributed (iid) real and imaginary parts, $\mathbf{n} \sim C\mathcal{N}(\mathbf{0}, \sigma^2 \mathbf{I})$, the likelihood of the data is also complex Gaussian distributed $p(\mathbf{y}|\mathbf{x}) \sim C\mathcal{N}(\mathbf{A}\mathbf{x}, \sigma^2 \mathbf{I})$,

$$p(\mathbf{y}|\mathbf{x}) = \pi^{-N} \sigma^{-2N} \mathrm{e}^{-\|\mathbf{y} - \mathbf{A}\mathbf{x}\|_2^2 / \sigma^2}.$$
(13)

Following Ref. 32, we assume that the coefficients of the solution \mathbf{x} are iid and follow a Laplacian-like distribution (for complex random variables). Such a prior has been shown to encourage sparsity in many situations because of the heavy tails and sharp peak at zero. The corresponding prior is

$$p(\mathbf{x}) \propto \prod_{i=1}^{N} e^{-\sqrt{(\operatorname{Rex}_i)^2 + (\operatorname{Imx}_i)^2}/\nu} = e^{-\|\mathbf{x}\|_1/\nu}.$$
 (14)

The LASSO estimate, Eq. (8), can be interpreted as the MAP estimate,

$$\hat{\mathbf{x}}_{\text{MAP}} = \underset{\mathbf{x}}{\operatorname{argmin}} [\|\mathbf{y} - \mathbf{A}\mathbf{x}\|_{2}^{2} + \mu \|\mathbf{x}\|_{1}] = \hat{\mathbf{x}}_{\ell_{1}}(\mu), \quad (15)$$

where $\mu = \sigma^2 / \nu$.

Equation (14) imposes no restriction on the source phases. Here, the phase is assumed uniformly $[0, 2\pi)$ distributed.

III. MULTIPLE-SNAPSHOT DOA ESTIMATION

Even though for moving sources it befits to solve one optimization problem for each snapshot sequentially,¹⁰ for stationary scenarios, the sensor data statistics can be

aggregated across snapshots to provide a more stable estimate. Multiple snapshots are referred to as multiple measurement vectors, and the recovery might have better performance than single measurement vectors.³³ Potentially the recovery can be made more robust by using a likelihood function Eq. (13) with colored noise (full covariance matrix) or based on the Huber norm.³⁴ For the multiple-snapshot case, all snapshot are collected into one matrix,

$$\mathbf{Y} = \mathbf{A}\mathbf{X} + \mathbf{N},\tag{16}$$

where, for *L* snapshots, $\mathbf{Y} = [\mathbf{y}(1), ..., \mathbf{y}(L)]$ and $\mathbf{N} = [\mathbf{n}(1), ..., \mathbf{n}(L)]$ are $M \times L$ matrices with the measurement and noise vectors per snapshot as columns, respectively, and \mathbf{X} is the $N \times L$ signal with the complex source amplitudes at the *N* look directions per snapshot as columns. For stationary sources, the matrix $\mathbf{X} = [\mathbf{x}(1), ..., \mathbf{x}(L)]$ exhibits row sparsity, i.e., it has a constant sparsity profile for every column because the few existing sources are associated with the same DOA for all snapshots. As the sources are stationary, it makes sense to sum the source energy across all snapshots, giving the row norm \mathbf{x}^{ℓ_2} ,

$$\mathbf{x}^{\ell_2} = \left(\sum_{l=1}^L |\mathbf{X}_{\cdot l}|^2\right)^{1/2}.$$
 (17)

This quantity is sparse and in analogy with the single snapshot case we impose a Laplacian-like prior

$$p(\mathbf{X}) = p(\mathbf{x}^{\ell_2}) \propto \exp\left(-\|\mathbf{x}^{\ell_2}\|_1/\nu\right),\tag{18}$$

with no phase assumption. Similar to Eq. (14), we assume the phase is uniformly iid distributed on $[0, 2\pi)$.

We assume an iid complex Gaussian distribution for the data likelihood

$$p(\mathbf{Y}|\mathbf{X}) \propto \exp\left(-\|\mathbf{Y} - \mathbf{A}\mathbf{X}\|_{\mathcal{F}}^2/\sigma^2\right).$$
(19)

Using Bayes theorem, the MAP solution is then

$$\mathbf{X} = \operatorname{argmax}_{p(\mathbf{Y}|\mathbf{X})p(\mathbf{X})}$$

=
$$\operatorname{argmin}_{\mathbf{X} \in \mathbb{C}^{N \times L}} \|\mathbf{Y} - \mathbf{A}\mathbf{X}\|_{\mathcal{F}}^{2} + \mu \|\mathbf{x}^{\ell_{2}}\|_{1}.$$
 (20)

In this formulation, we search for a sparse solution via the ℓ_1 constraint. The source amplitude can, however, vary across snapshot. This is in contrast to covariance-matrix based beamforming that just inverts for the average source power. The processing performance can be improved by doing an eigenvalue decomposition of **X** and retaining just the largest eigenvalues; see Refs. 3 and 4. The smaller eigenvalues contain mostly noise so this improves processing. However, this eigenvalue decomposition is not done here as this has features similar to forming a sample covariance matrix.

Once the active steering vectors have been recovered, the unbiased source amplitudes are estimated for each snapshot, similar to the single snapshot case, Eq. (9),

$$\hat{\mathbf{X}}_{\rm CS} = \mathbf{A}_{\alpha}^{+} \mathbf{Y}.$$
 (21)

If desired, an average power estimate $\mathbf{x}_{CS}^{\ell_2}$ can be obtained from the ℓ_2 -norm of the rows of $\hat{\mathbf{X}}_{CS}$, with the *i*th element squared of $\mathbf{x}_{CS}^{\ell_2}$ being the source power estimate at θ_i .

For reference, the CBF, MVDR, and MUSIC use the data sample covariance matrix,

$$\mathbf{C} = \frac{1}{L} \mathbf{Y} \mathbf{Y}^{H}.$$
 (22)

The beamformer power for CBF and MVDR, respectively, is then,

$$P_{\rm CBF}(\theta) = \mathbf{w}_{\rm CBF}^{H}(\theta) \mathbf{C} \mathbf{w}_{\rm CBF}(\theta), \qquad (23)$$

$$P_{\text{MVDR}}(\theta) = \mathbf{w}_{\text{MVDR}}^{H}(\theta) \mathbf{C} \mathbf{w}_{\text{MVDR}}(\theta), \qquad (24)$$

where the corresponding weight vectors are given by

$$\mathbf{w}_{\rm CBF}(\theta) = \mathbf{a}(\theta),\tag{25}$$

$$\mathbf{w}_{\mathrm{MVDR}}(\theta) = \frac{\mathbf{C}^{-1}\mathbf{a}(\theta)}{\mathbf{a}^{H}(\theta)\mathbf{C}^{-1}\mathbf{a}(\theta)}.$$
 (26)

The CBF can also be based directly on snapshots as the single snapshot CBF Eq. (6) can be generalized to multiple snapshots, $\hat{\mathbf{X}}_{\text{CBF}} = \mathbf{A}^H \mathbf{Y}$. The power estimates $P_{\text{CBF}}(\theta_i)$, $P_{\text{MVDR}}(\theta_i)$, and the corresponding *i*th squared component of $\mathbf{x}_{\text{CS}}^{\ell_2}$ are thus comparable. Note that because the MVDR weights in Eq. (26) involve the inverse of the sample covariance matrix, MVDR requires a full rank C, i.e., $L \ge M$ snapshots.

The MUSIC (Ref. 7) is based on the eigendecomposition of the data sample covariance matrix Eq. (22) and the separation of the signal and the noise subspace,

$$\mathbf{C} = \mathbf{U}_s \mathbf{\Lambda}_s \mathbf{U}_s^H + \mathbf{U}_n \mathbf{\Lambda}_n \mathbf{U}_n^H.$$
⁽²⁷⁾

The signal eigenvectors \mathbf{U}_s corresponding to the largest eigenvalues, $\mathbf{\Lambda}_s$, are at the same subspace as the steering vectors, Eq. (1), while the noise eigenvectors \mathbf{U}_n are orthogonal to the subspace of the steering vectors, thus $\mathbf{a}(\theta)^H \mathbf{U}_n = \mathbf{0}$. MUSIC uses the orthogonality between the signal and the noise subspace to locate the maxima in the spectrum,

$$P_{\text{MUSIC}}(\theta) = \frac{1}{\mathbf{a}(\theta)^{H} \mathbf{U}_{n} \mathbf{U}_{n}^{H} \mathbf{a}(\theta)}.$$
(28)

Both MVDR and MUSIC overcome the resolution limit of the conventional beamformer by exploiting signal information conveyed by the data sample matrix. However, their performance depends on the eigenvalues of the data sample matrix thus it degrades with few snapshots, when the data sample matrix is rank deficient, and in the presence of coherent sources, when the signal subspace is reduced (Chap. 9 in Ref. 7). CS does not have these limitations as it utilizes directly the measured pressure **Y**.

IV. REGULARIZATION PARAMETER SELECTION

The choice of the regularization parameter μ in Eq. (8), also called the LASSO shrinkage parameter, is crucial as it

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controls the balance between the sparsity of the estimated solution and the data fit determining the quality of the reconstruction.

For large μ , the solution is very sparse (with small ℓ_1 -norm), but the data fit is poor as indicated in Fig. 1. As μ decreases toward zero, the data fit is gradually improved because the corresponding solutions become less sparse. Note that for $\mu = 0$, the solution in Eq. (8) becomes the unconstrained least squares solution. Because the LASSO path is derived and demonstrated for a single observation, the statistics of the source signal or noise is irrelevant.

A. The LASSO path

As the regularization parameter μ evolves from ∞ to 0, the LASSO solution in Eq. (8) changes continuously following a piecewise smooth trajectory referred to as the solution path or the LASSO path.^{25,26} In this section, we show that the singularity points in the LASSO path are associated with a change in the sparsity of the solution and can be used to indicate an adequate choice for μ .

We obtain the full solution path using convex optimization to solve Eq. (8) iteratively for different values of μ . We use the CVX toolbox for disciplined convex optimization that is available in the MATLAB environment. It uses interior point solvers to obtain the global solution of a well-defined optimization problem.^{29–31}

Let $L(\mathbf{x}, \mu)$ denote the objective function in Eq. (8),

$$L(\mathbf{x},\mu) = \|\mathbf{y} - \mathbf{A}\mathbf{x}\|_{2}^{2} + \mu\|\mathbf{x}\|_{1}.$$
 (29)

The value $\hat{\mathbf{x}}$ minimizing Eq. (29) is found from its subderivative,

$$\partial_{\mathbf{x}} L(\mathbf{x}, \mu) = 2\mathbf{A}^{H}(\mathbf{A}\mathbf{x} - \mathbf{y}) + \mu \partial_{\mathbf{x}} \|\mathbf{x}\|_{1},$$
(30)

where the subdifferential operator ∂_x is a generalization of the partial differential operator for functions that are not differentiable everywhere (Ref. 31, p. 338). The subgradient for the ℓ_1 -norm is the set of vectors,

$$\partial_{\mathbf{x}} \|\mathbf{x}\|_{1} = \{\mathbf{s} : \|\mathbf{s}\|_{\infty} \le 1, \mathbf{s}^{H}\mathbf{x} = \|\mathbf{x}\|_{1}\},$$
(31)

which implies,

$$s_i = \frac{x_i}{|x_i|}, \quad x_i \neq 0,$$

 $|s_i| \le 1, \quad x_i = 0,$ (32)

i.e., for every active element $x_i \neq 0$ of the vector $\mathbf{x} \in \mathbb{C}^N$, the corresponding element of the subgradient is a unit vector in the direction of x_i . For every null element $x_i = 0$ the corresponding element of the subgradient has magnitude less than or equal to one. Thus the magnitude of the subgradient is uniformly bounded by unity, $\|\mathbf{s}\|_{\infty} \leq 1$.

Denote

$$\mathbf{r} = 2\mathbf{A}^{H}(\mathbf{y} - \mathbf{A}\hat{\mathbf{x}}),\tag{33}$$

the beamformed residual vector for the estimated solution $\hat{\mathbf{x}}$. Because Eq. (29) is convex, the global minimum is attained if $\mathbf{0} \in \partial_{\mathbf{x}} L(\mathbf{x}, \mu)$, which leads to the necessary and sufficient condition

$$\mu^{-1}\mathbf{r} \in \partial_{\mathbf{x}} \|\mathbf{x}\|_{1}. \tag{34}$$

Then from Eqs. (32) and (34), the coefficients $r_i = 2\mathbf{a}_i^H(\mathbf{y} - \mathbf{A}\hat{\mathbf{x}})$ of the beamformed residual vector $\mathbf{r} \in \mathbb{C}^N$ have amplitude such that

$$|r_i| = \mu, \quad \hat{x}_i \neq 0,$$

 $|r_i| \le \mu, \quad \hat{x}_i = 0,$
(35)

i.e., whenever a component of $\hat{\mathbf{x}}$ becomes non-zero, the corresponding element of the beam-formed residual hits the boundary identified with the regularization parameter, $\|\mathbf{r}\|_{\infty} \leq \mu$.

For multiple snapshots, with the $\hat{\mathbf{X}}$ determined from Eq. (20), the beamformed residuals become

$$\mathbf{R} = 2\mathbf{A}^{H}(\mathbf{Y} - \mathbf{A}\hat{\mathbf{X}}), \quad \mathbf{r}_{i} = \sqrt{\sum_{j=1}^{L} |\mathbf{R}_{ij}|^{2}}.$$
 (36)

The values of μ when changes in sparsity appear are obtained similarly to the single snapshot case.

B. Algorithm for the LASSO path

Although many algorithms exist for solving the LASSO problem, we have good experience with the new algorithm for compressive beamforming in Table I as it is reasonably fast and accurate. Section IV A motivates the formulation of the new algorithm where the values of μ for different sparsity levels are indicated by the residual vector **r**, solving the dual problem.³⁵ For large μ , the solution $\hat{\mathbf{x}} = \mathbf{0}$ is trivial and $\mathbf{r} = 2\mathbf{A}^H \mathbf{y}$ in Eq. (33). Decreasing μ , a first element of **x** becomes active when the corresponding element of **r** hits the

TABLE I. Fast iterative algorithm to solve the LASSO problem (8) for a desired sparsity level K and estimating the unbiased complex source amplitudes (9).

1: In	nitialize $i = 0$, $\mathbf{x}_{\ell_{i}}^{0} = 0$, $\mathbf{r}^{0} = 2\mathbf{A}^{H}\mathbf{y}$
2: w	while $ \mathcal{M}_i < K$
	i = i + 1
3:	$\mu^{i} = (1 - F) \text{ peak } (\mathbf{r}^{i-1}, K) + F \text{ peak}(\mathbf{r}^{i-1}, K+1)$
4a:	$\mathbf{x}_{\ell_1}^i =$ solution to Eq. (8) for \mathbf{A} , \mathbf{y} , $\mu = \mu^i$
4b:	$\mathbf{r}^{i} = 2\mathbf{A}^{H}(\mathbf{y} - \mathbf{A}\mathbf{x}_{\ell_{1}}^{i})$
5:	$\mathcal{M}_i = \{m \mathbf{x}_{\ell_1,m}^i > \delta_i\}, \delta_i = \epsilon \ \mathbf{x}_{\ell_1}^i \ _\infty$
er	nd
6: if	$F\left \mathcal{M}_{i} ight >K$
7:	$\mathcal{M}_i = \{m x_{\ell_1,m}^i > \delta_i\}, \delta_i = \text{peak}(x_{\ell_1}^i , K)$
er	nd
Л	$\mathcal{A} = \mathcal{M}_i$
8: $\hat{\mathbf{x}}_i$	$_{\ell_1}(\mu^i)=oldsymbol{x}^i_{\ell_1}$
9: \hat{x}_{0}	$_{\rm CS} = A_{\mathcal{M}}^+ y$
10: O	Dutput: $\mu^i, \hat{\mathbf{x}}_{\ell_1}(\mu^i), \hat{\mathbf{x}}_{CS}, \mathcal{M}.$

J. Acoust. Soc. Am. 138 (4), October 2015

Gerstoft et al. 2007

boundary, $\mu = 2 \|\mathbf{A}^{H}\mathbf{y}\|_{\infty}$. Assuming perfectly incoherent columns of **A** and solving for the second peak of **r** hitting the boundary μ indicates the value of μ for which a second element of **x** becomes active. This way we follow the LASSO path in Fig. 1 towards less sparse solutions by decreasing μ as detailed in Ref. 35.

Starting from Eq. (33) with $x_{\ell_1}(\mu^i)$ corresponding to regularization μ^i for the set of active indexes \mathcal{M}_i , the residual for the *n*th steering vector is now found,

$$r_{n}(\mu^{i}) = 2\boldsymbol{a}_{n}^{H}(\boldsymbol{y} - \boldsymbol{A}\boldsymbol{x}_{\ell_{1}}(\mu^{i}))$$

$$= 2\boldsymbol{a}_{n}^{H}\left(\boldsymbol{y} - \sum_{m \in \mathcal{M}_{i}} \boldsymbol{a}_{m}\boldsymbol{x}_{\ell_{1},m}(\mu^{i})\right)$$

$$\approx 2\boldsymbol{a}_{n}^{H}\left(\boldsymbol{y} - \sum_{m \in \mathcal{M}_{i-1}} \boldsymbol{a}_{m}\boldsymbol{x}_{\ell_{1},m}(\mu^{i-1})\right)$$
(37)

$$\approx 2a_n^H y.$$
 (38)

The two progressively stronger approximations, in Eqs. (37)–(38), in the preceding text are valid if the steering vectors corresponding to the final active set is sufficiently incoherent $|a_n^H a_m| \approx 0$. Equation (38) actually corresponds to the conventional beamformer $A^H y$ for a single snapshot. The preceding equation is used for the selection of μ so it does not mean that the peaks in the conventional beamformer corresponds to the CS solution.

The procedure is given in Table I, where peak(\mathbf{r} , k) is the *k*th peak of \mathbf{r} . We choose F = 0.9.

The dual method has been used to estimate the solution path of the real-valued²⁵ and the complex-valued³⁵ generalized LASSO problems. The generalized LASSO uses the ℓ_1 -norm to enforce structural or geometric constraints on the solution by replacing the sparsity constraint $||\mathbf{x}||_1$ with $||\mathbf{Dx}||_1$ for a structured matrix **D**. The generalized formulation performs well in certain applications, e.g., recovery of continuous sources by promoting block sparsity³⁶ and DOA tracking for moving sources by an adaptive update of a diagonal weighting matrix **D** which reflects the evolution of the source probability distribution.¹⁰

C. Regularization parameter selection via the LASSO path

The LASSO performance in DOA estimation is evaluated by simulations starting with a large μ and subsequently decreasing its value. We consider an ULA with M = 20sensors and spacing $d = \lambda/2$. Three sources are at DOAs $[-5, 0, 20]^{\circ}$ with corresponding magnitudes [1, 0.6, 0.2](linear) or [0, -4, -14] dB. The sensing matrix **A** in Eq. (3) is defined on a coarse angular grid $[-90^{\circ}:5^{\circ}:90^{\circ}]$ (Figs. 2–4) and a denser grid $[-90^{\circ}:1^{\circ}:90^{\circ}]$ (Fig. 5). The noise variance in Eq. (4) is chosen such that SNR = 20 dB.

The trade-off between regularization term $\|\hat{\mathbf{x}}\|_1$ and the data fit $\|\mathbf{y} - \mathbf{A}\hat{\mathbf{x}}\|_2^2$ in the LASSO estimate, Eq. (8), for a range of values of μ is depicted in Fig. 2. The relevant values of μ for the LASSO path are found between the two dots in Fig. 2(b), i.e., $1.54 > \mu > 0.02$. For these values of μ , the importance shifts from favoring sparser solutions for large μ toward diminishing the model residual's ℓ_2 -norm for smaller μ . From inspecting Fig. 2(b), it is difficult inferring the value

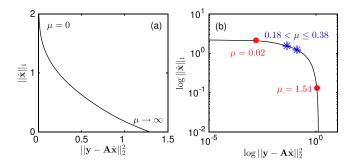


FIG. 2. (Color online) The data error $\|\mathbf{y} - \mathbf{A}\hat{\mathbf{x}}\|_2^2$ describing the goodness of fit, versus the ℓ_1 -norm in (a) linear scale and (b) log-log scale for the estimated solution $\hat{\mathbf{x}}$ for different values of the regularization parameter μ in the LASSO problem Eq. (8) for sparse DOA estimation.

of μ which results in the desired sparsity level. The LASSO path offers a more insightful method to determine the range of good values of μ [contained within the asterisks in Fig. 2(b)] as explained in the following text.

Figure 3 shows (a) the sparsity level $\|\hat{\mathbf{x}}\|_0$ of the LASSO solution, (b) the properties of the LASSO path and (c) the corresponding residual vector versus the regularization parameter μ . Because the interest is on sparse solutions $\hat{\mathbf{x}}$, it is natural inspecting the LASSO path for decreasing values of μ , i.e., interpreting Fig. 3 from right to left.

For large values of μ (e.g., $\mu = 2$), the problem in Eq. (8) is over-regularized, forcing the trivial solution $\hat{\mathbf{x}} = \mathbf{0}$ [Fig. 3(b)], thus $\|\hat{\mathbf{x}}\|_0 = 0$ [Fig. 3(a)]. In this case, the slopes for all components $|r_i|$ are zero [Fig. 3(c)] because $|r_i| = |2\mathbf{a}_i^H \mathbf{y}| < \mu$ for all $i \in [0, ..., N]$, which is independent of μ .

The first non-zero component of \hat{x} appears at $\mu = 2 \|\mathbf{A}^H \mathbf{y}\|_{\infty} = 1.76$ and remains active for $\mu \le 1.76$ [Fig. 3(b)], increasing the sparsity level to $\|\hat{\mathbf{x}}\|_0 = 1$ [Fig. 3(a)]. The corresponding component $|r_i| = |2\mathbf{a}_i^H(\mathbf{y} - \mathbf{a}_i \hat{x}_i)|$ [Fig. 3(c)] is equal to μ for $\mu \le 1.76$. The other components r_j change slope at the singular point $\mu = 1.76$ because now $|r_j|$

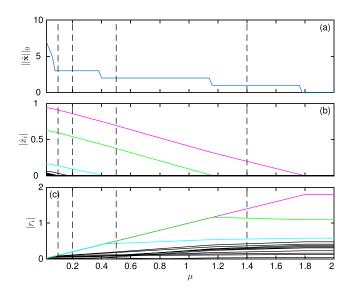


FIG. 3. (Color online) The LASSO path versus μ for three sources and SNR = 20 dB. (a) Sparsity level of the estimate $\hat{\mathbf{x}}$. (b) Paths for each component of the solution $\hat{\mathbf{x}}$. (c) Paths for each component of the beamformed residual $|\mathbf{r}| = 2|\mathbf{A}^{H}(\mathbf{y} - \mathbf{A}\hat{\mathbf{x}})|$. The vertical dashed lines indicates values of μ used in Figs. 4 and 5.

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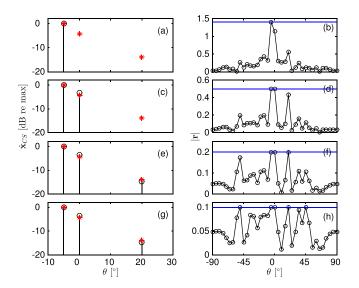


FIG. 4. (Color online) The unbiased estimate $\hat{\mathbf{x}}_{CS}(\circ)$ for the true source \mathbf{x} (*) and the corresponding beamformed residual vector for a coarse angular grid $[-90:5:90]^{\circ}$. (a)–(b) $\mu = 1.4$, (c)–(d) $\mu = 0.5$, (e)–(f) $\mu = 0.2$, and (g)–(h) $\mu = 0.1$ (corresponding to dashed lines in Fig. 3). The horizontal line in the residual plot (right) indicates the value of μ .

= $|2\mathbf{a}_{j}^{H}(\mathbf{y} - \mathbf{a}_{i}\hat{x}_{i})| < \mu$ for all $j \in [0, ..., N], j \neq i$. For $\mu \leq 1.14$, $\|\hat{\mathbf{x}}\|_{0} = 2$ [Fig. 3(a)] as $\hat{\mathbf{x}}$ acquires a second nonzero component [Fig. 3(b)] and the corresponding component $|r_{i}|$ becomes equal to μ [(Fig. 3(c)]. Similarly, the estimated solution has a third non-zero component for $\mu \leq 0.38$.

For $\mu \le 0.18$, $\hat{\mathbf{x}}$ has many non-zero components [Figs. 3(b) and 3(c)], and its sparsity level increases abruptly [Fig. 3(a)]. For such low values of μ , the importance shifts to the data fitting term (ℓ_2 -norm term) in the regularized problem, Eq. (8), and $\hat{\mathbf{x}}$ includes many non-zero noisy components gradually reducing the data error.

The specific values of μ at which an element of $\hat{\mathbf{x}}$ becomes active are denoted as the singular points in the piecewise smooth LASSO path. At a singular point, some component of \mathbf{r} hits the boundary μ , i.e., $|r_n| = \mu$ for some index *n*. Thus the properties of the LASSO path indicate the

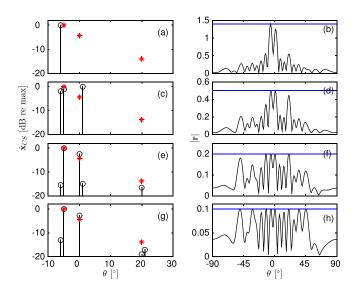


FIG. 5. (Color online) As in Fig. 4 but for the denser angular grid $[-90{:}1{:}90]^{\circ}.$

selection of the regularization parameter μ . For example, for a predefined sparsity level *K* a good choice of μ is found by decreasing μ until the *K*th singular point at the LASSO path.

Owing to the piecewise smooth nature of the LASSO path, there is a range of μ that gives the same sparsity level (i.e., between two singular points). In principle, the lowest μ in this range is desired as it gives the best data fit. Although any value of μ that achieves the desired sparsity suffices as once the active DOAs are recovered, the unbiased amplitudes are determined from Eq. (9).

Figure 4 shows the unbiased solution, Eq. (9), along with the corresponding beamformed residual for four sparsity levels of μ . Notice how the residuals decrease in value as μ is reduced. For $\mu = 0.1$, Fig. 4 shows that five potential source locations exists as they have hit the boundary, so that components of $|\mathbf{r}|$ becomes equal to μ . Solving Eq. (9) shows that two sources are weak and are not shown.

To increase precision in the LASSO reconstruction, a finer angular grid is required. However, angular grid refinement also causes higher coherence among steering vectors, Eq. (1), and the problem in Eq. (2) becomes increasingly underdetermined. Then when solving the LASSO minimization, Eq. (8) might not exhibit the desired sparsity. Due to basis coherence and as μ decreases, components in the estimate $\hat{\mathbf{x}}$ can be either activated (become non-zero) or annihilated. Similarly the residual components can hit or leave the boundary²⁵ [where components of |**r**| is equal to μ , see Eq. (35)].

In Fig. 5, a denser angular grid with spacing 1° and setup as Fig. 4 is used. For $\mu = 1.4$, there is just one active component [Fig. 5(a)] at -6° which is 1° away from the strongest DOA. This offset is mainly due to basis coherence, the correct location is not yet recovered. As μ is decreased, the correct bin is eventually obtained [Fig. 5(e)]. Thus when searching for a *K* sparse solution, it is often advantageous to search initially for more than *K* peaks and then limit the final solution to the *K* most powerful elements.

The residual **r** is systematically decreased as μ is reduced. All the active component can be seen as where components of $|\mathbf{r}|$ becomes equal to μ in the right panel of Fig. 5.

V. DOA ESTIMATION ERROR EVALUATION

If the source DOAs are well separated with not too different magnitude, the DOA estimation for multiple sources using CBF and CS turns out to behave similarly. They differ, however, in their behavior whenever two sources are closely spaced. The same applies for MVDR under the additional assumptions of incoherent arrivals and sufficient number of snapshots, $L \ge M$. The details are of course scenario dependent.

For the purpose of a quantitative performance evaluation with synthetic data, the estimated, $\hat{\theta}_k$, and the true, θ_k^{true} , DOAs are paired with each other such that the root mean squared DOA error is minimized in each single realization. After this pairing, the ensemble root-mean-squared error is computed,

$$\text{RMSE} = \sqrt{\text{E}\left[\frac{1}{K}\sum_{k=1}^{K} \left(\hat{\theta}_{k} - \theta_{k}^{\text{true}}\right)^{2}\right]}.$$
 (39)

J. Acoust. Soc. Am. 138 (4), October 2015

Gerstoft et al. 2009

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The data are generated to have a fixed SNR Eq. (4). The source phases of each x component is uniformly distributed on $[0, 2\pi)$ to generate a sample covariance matrix from which MVDR/MUSIC can resolve incoherent sources.

CBF suffers from low-resolution and the effect of sidelobes for both single and multiple data snapshots, thus the simple peak search used here is too simple. These problems are reduced in MVDR and MUSIC for multiple snapshots and they do not arise with CS.

The optimal performance for *K* sources is found by searching over all combinations of steering vectors for the maximum likelihood solution, Eq. (13), i.e., the best fitting source vector using Eq. (9). This is a NP-hard combinatorial problem that for *N* look directions requires evaluation of N!/K!(N-K)! solutions. For N = 361 and either K = 2 or K = 3, this gives 77 000 or 7700 000 combinations to be evaluated. This makes the exhaustive search approach impractical for larger *K*.

In the following simulation, we consider an array with M = 20 elements with spacing $d = \lambda/2$. The DOAs are assumed to be on a fine angular grid $[-90^{\circ}:0:5^{\circ}:90^{\circ}]$, i.e., $\mathbf{A} \in \mathbb{C}^{20\times361}$. The regularization parameter μ is chosen to correspond to the K + 2 largest peak of the residual in Eq. (33) using the procedure in Table I and retaining only the K largest source powers. We require the peaks of the CS to be at least 4 bins apart. Thus the exhaustive and the CS do not solve the identical problem, as the CS solves a smaller

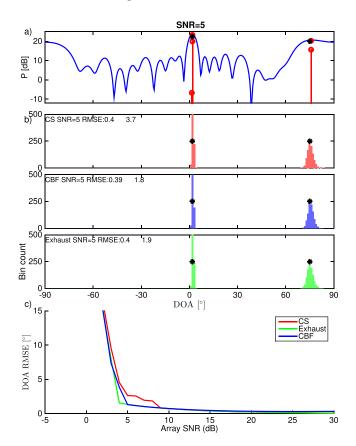


FIG. 6. (Color online) Single snapshot example for two sources at DOAs $[2,75]^{\circ}$ and magnitudes [22,20] dB. At SNR = 5 dB (a) spectra for CBF, CS (o) and unbiased CS (o, higher levels), and (b) CS, CBF and exhaustive-search histogram based on 1000 Monte Carlo simulations, and (c) CS, CBF and exhaustive-search performance versus SNR. The true source positions (*) are indicated in (a) and (b).

problem. Note that panel (c) in Figs. 6–9 shows the simulation results versus array SNR defined in Eq. (4).

A. Single snapshot

In the first scenario, we consider a single snapshot case with additive noise with K=2 well-separated DOAs at [2,75]° with magnitudes [22,20] dB, see Fig. 6. In the second scenario, a third weak source is included very close to the first source: thus K=3 and the source DOAs are [-3, 2, 75]° with magnitudes [12,22,20] dB, see Fig. 7. The synthetic data are generated according to Eq. (2).

For the first scenario, the CS diagrams in Fig. 6(a) show DOA estimation with small variance but indicate a bias toward endfire, as for the true DOA 75°, the CS estimate is 76°. Toward endfire, the main beam becomes broader and absorbs more noise power, The CBF spectra Fig. 6(a) are characterized by a high sidelobe level but for the two well-separated similar-magnitude sources this is a minor problem.

Using Monte Carlo simulations, we repeat the CS inversion for 1000 realizations of the noise in Fig. 6(b). The RMSE increases toward the endfire directions. This is to be expected as the main beam becomes wider and this results in a lower DOA resolution.⁴ Because the sources are wellseparated in this scenario, CS, CBF, and exhaustive search perform similarly with respect to RMSE.

Repeating the Monte Carlo simulations at several SNRs gives the RMSE performance of CS and CBF in Fig. 6(c). Their performance is about the same because the DOAs are well-separated.

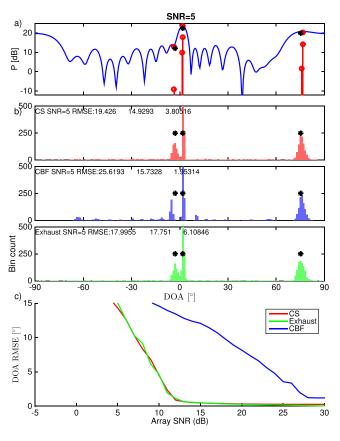


FIG. 7. (Color online) As Fig. 6 but for three sources at DOAs $[-3, 2, 75]^{\circ}$ and magnitudes [12,22,20] dB.

In the second scenario, the CBF cannot resolve the two closely spaced sources with DOAs $[-3, 2]^{\circ}$. They are less than a beamwidth apart as indicated in Fig. 7(a). Sidelobes cause a few DOA estimation errors at -65° in the CBF histogram, Fig. 7(b). Because CS obtains high-resolution even for a single snapshot, it performs much better than CBF, Fig. 7(c).

B. Multiple snapshot

In the multiple-snapshot scenario, MVDR and MUSIC use the data sample covariance matrix Eq. (22), whereas CBF and CS works directly on the observations **X** Eq. (20). The sample covariance matrix is formed by averaging *L* synthetic data snapshots. The source magnitude is considered invariant across snapshots. The source phase is for each snapshot sampled from a uniform distribution on $[0, 2\pi)$.

Due to the weak performance of MVDR in scenarios with coherent arrivals,⁷ we assume incoherent arrivals in the simulations although not needed for CS. For CS, we use Eq. (20) with a similar choice of regularization parameter μ as for the single snapshot case.

Using the same setup as in Fig. 7, but estimating the source DOAs based on L = 50 snapshots gives the results in Fig. 8. At SNR = 0 dB, the diagrams in Fig. 8(a) show that CS localizes the sources well in contrast to the CBF and MVDR that is also indicated in the histograms in Fig. 8(b).

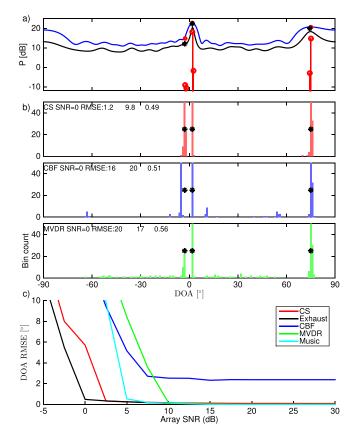


FIG. 8. (Color online) Multiple L = 50 snapshot example for three sources at DOAs $[-3, 2, 75]^{\circ}$ with magnitudes [12,22,20] dB. At SNR = 0 dB (a) spectra for CBF, MVDR, and CS (o) and unbiased CS (o, higher levels), and (b) CS, CBF and MVDR histogram based on 100 Monte Carlo simulations, and (c) CS, exhaustive-search, CBF, MVDR, and MUSIC performance versus SNR. The true source positions (*) are indicated in (a) and (b).

The RMSE in Fig. 8(c) shows that CBF does not give the required resolution even for high SNR. MVDR performs well for SNR > 10 dB, whereas CS performs well for SNRs down to 2.5 dB.

In a third scenario, the weak broadside sources are moved closer with DOAs defined as $[-2, 1, 75]^{\circ}$. Figure 9 gives about the same DOA estimates for CBF as it is already at its maximum performance even for high array SNR, confirming its low resolution. MVDR fails for SNR < 20 dB, which is 10 dB higher than the corresponding value in Fig. 8(c) (MUSIC fails also at a level 10 dB higher). Contrarily, CS fails only for SNR < 5 dB which is 2.5 dB higher [Figs. 8(c) and 9(c)]. Note how MVDR completely misses the weak source at -2° in Fig. 9(c), but CS localizes it with a larger spread. Thus as the weak source moves closer to the strong source, CS degrades slower than MVDR in terms of RMSE. This is a good indication of its high-resolution capabilities.

Figure 10 shows the estimated power at the one realization in Fig. 8(a) of L = 50 snapshots inverted simultaneously. We emphasize the scale of the problem. Equation (16) has $20 \times 50 = 1000$ equations to determine $361 \times 50 = 18050$ complex-valued variables at 361 azimuths and 50 snapshots observed on 20 sensors. The sparsity constraint is crucial here.

The CS (and especially the exhaustive-search) requires several orders of magnitude more CPU-time than the beamforming methods.

Many other simulations could be performed, for example, colored noise, no assumptions on number of sources,

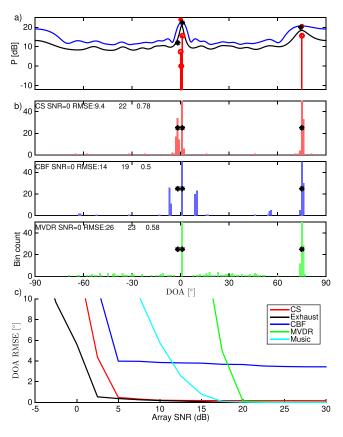


FIG. 9. (Color online) As Fig. 8 but with closer spaced sources $[-2, 1, 75]^{\circ}$.

J. Acoust. Soc. Am. 138 (4), October 2015

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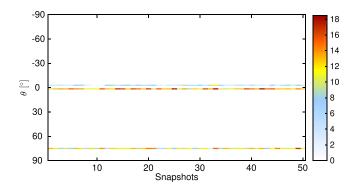


FIG. 10. (Color online) Power (linear) for the multiple snapshot case across azimuths and snapshots for one noise realization at SNR = 0 dB for the scenario with DOAs at $[-3, 2, 75]^{\circ}$.

and random source locations. From initial exploration of these, it is our impression that CS will perform well, though more simulations are required.

VI. EXPERIMENTAL RESULTS

The high-resolution performance of CS both in singleand multiple-snapshot cases is validated with experimental data in a complex multi-path shallow-water environment, and it is compared with conventional methods, namely, CBF and MVDR.

The data set is from the shallow water evaluation cell experiment 1996 (SWellEx-96) Event S5 (Refs. 37 and 38) collected on a 64-element vertical linear array. The array has uniform inter-sensor spacing 1.875 m and was deployed at waterdepth 16.5 m spanning 94.125–212.25 m. During Event S5, from 23:15–00:30 on 10–11 May 1996 west of Point Loma, CA, two sources, a shallow and a deep, were towed simultaneously from 9 km southwest to 3 km northeast of the array at a speed of 5 kn (2.5 m/s). Each source was transmitting a unique set of tones.

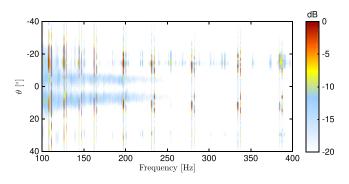


FIG. 11. (Color online) Spatial CBF spectrum across frequency at the source's closest point of approach to the array.

Here we are interested in the deep source towed at 54 m depth while at the vicinity of the closest point of approach (CPA) which was 900 m from the array and occurred around 00:15, 60 min into the event. The deep-towed source signal submitted a set of nine frequencies [112,130,148,166,201, 235,283,338,388] Hz at approximately 158 dB re 1 μ Pa. The processed recording has duration of 1.5 min (covering 0.5 min before and 1 min after the CPA) sampled at 1500 Hz. It was split into 87 snapshots of 2¹² samples (2.7 s) duration, i.e., with 63% overlap.

Figure 11 shows the multiple-snapshot CBF spatial spectrum, Eq. (24), over the 50–400 Hz frequency range. Arrivals are detected not only at the transmitted tonal frequencies of the deep towed source but also at several other frequencies corresponding to the shallow-towed source tonal frequencies, weaker deep source frequencies, and the acoustic signature of the tow-ship.

Single-snapshot processing with CBF and CS at the deep source tonal set, contour plots in Fig. 12, indicates the presence of several multipath arrivals, which are adequately stationary along the snapshots at the CPA. Due to the significant sound speed variation, it is not straightforward to associate the reconstructed DOAs with specific reflections. The CBF map

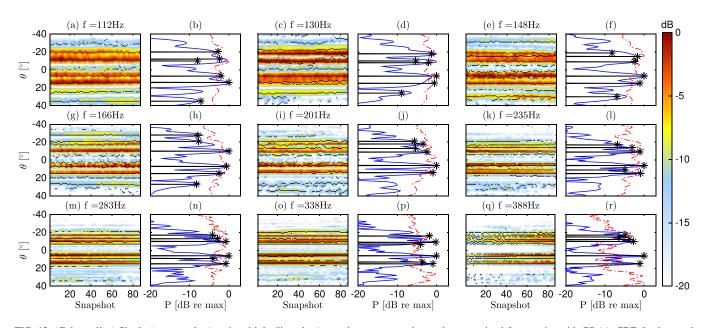


FIG. 12. (Color online) Single (contour plots) and multiple (line plots) snapshot reconstruction at the transmitted frequencies with CS (\star), CBF (background color, solid) and MVDR (dashed). For the single snapshot, we have assumed K = 10 sources while for the multiple snapshot K = 6.

2012 J. Acoust. Soc. Am. 138 (4), October 2015

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comprises six significant peaks but suffers from low resolution and artifacts due to sidelobes and noise. To choose the regularization parameter in the LASSO formulation for CS reconstruction, we solve iteratively Eq. (8) as described in Table I with initial value $\mu = 2 \|\mathbf{A}^H \mathbf{y}\|_{\infty}$, until the obtained estimate has a sparsity level of 10. The CS reconstruction results in improved resolution due to the sparsity constraint and significant reduction of artifacts in the map.

Combining the data from all the snapshots and processing with CBF, MVDR, and CS, as in Sec. VB, reveals that MVDR fails to detect the coherent multipath arrivals; see line plots in Fig. 12. Again the peaks of CBF and CS are consistent, but CS offers improved resolution.

We have here used higher sparsity for the singlesnapshot processing to allow for identifying non-stationary paths. The non-stationary path can be seen in several of the contour plots, most prominently at 112, 130, and 201 Hz. When performing multiple-snapshot processing where the solution is constrained to remain active at one azimuth (but with varying power), the stationary paths are most likely to contribute to the CS solution.

VII. CONCLUSION

The estimation of multiple DOA is formulated as a sparse source reconstruction problem. This is efficiently solvable using CS as a least squares problem regularized with a sparsity promoting constraint. The resulting solution is the MAP estimate for both the single- and multiplesnapshot formulations. The regularization parameter balances the data fit and the solution's sparsity. It is selected so that the solution is sufficiently sparse providing highresolution DOA estimates. A procedure to find an adequate choice for the regularization parameter is described whereby the DOAs are obtained.

CS provides high-resolution acoustic imaging both with single and multiple snapshot. The performance evaluation shows that for single snapshot data, CS gives higher resolution than CBF. For multiple snapshots, CS provides higher resolution than MVDR/MUSIC, and the relative performance improves as the source DOAs move closer together.

The real data example indicates that CS is capable of resolving multiple coherent wave arrivals, e.g., stemming from multipath propagation.

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