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Non-Parametric High-Resolution SAR Imaging

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Abstract—The development of high-resolution two-dimensional spectral estimation techniques is of notable interest in synthetic aperture radar (SAR) imaging. Typically, data-independent techniques are exploited to form the SAR images, although such approaches will suffer from limited resolution and high sidelobe levels. Recent work on data-adaptive approaches have shown that both the iterative adaptive approach (IAA) and the sparse learning via iterative minimization (SLIM) algorithm offer excellent performance with high-resolution and low side lobe levels for both complete and incomplete data sets. Regrettably, both algorithms are computationally intensive if applied directly to the phase history data to form the SAR images. To help alleviate this, efficient implementations have also been proposed. In this paper, we further this work, proposing yet further improved implementation strategies, including approaches using the segmented IAA approach and the approximative quasi-Newton technique. Furthermore, we introduce a combined IAA-MAP algorithm as well as a hybrid IAA- and SLIM-based estimation scheme for SAR imaging. The effectiveness of the SAR imaging algorithms and the computational complexities of their fast implementations are demonstrated using the simulated Slidy data set and the experimentally measured GOTCHA data set.

Index Terms—Spectral estimation, synthetic aperture radar imaging, data adaptive techniques, efficient algorithms.

I. INTRODUCTION

SYNTHETIC aperture radar (SAR) systems find applicability in a wide variety of commercial and governmental applications, including monitoring, mapping, and reconnaissance systems, and offers notable benefits due to such systems' ability to image in all weather conditions and times of the day. The measured SAR images are generally processed via various pre- and post-processing techniques, such as data-independent Fourier transforms and back-projections, estimating the scene reflectivity intensity to form an intensity image. The Fourier methods exploit the relationship between the signal phase history measurements and the scene reflectivity, but generally suffers from limited resolution and/or sidelobe artifacts, as well as the speckle phenomena [1]–[5]. To reduce these effects, one is typically forced to include various forms of smoothing and filtering, although this will result in further reduced image resolution. These drawbacks have led to an interest in finding improved processing techniques to form the two-dimensional (2-D) spectral estimate required to form the SAR image [5]–[10]. Of the presented approaches, the data-dependent Capon and APES algorithms [7], [10]–[12] seem particularly promising, although both methods generally require multiple snapshots (which is hard to satisfy due to the platform motion) or the use of sub-apertures (which would lead to lower resolution) to form the required sample covariance matrix estimate. The use of sparse signal recovery methods have also been investigated (see, e.g., [13], [14]) as such systems are well able to recover sparse radar images with high resolution, without requiring multiple snapshots to do so. Regrettably, this form of approaches is sensitive to the choice of the various user parameters, which are typically difficult to select in practice, as well as lack robustness to varying noise levels [15]. Recently, the iterative adaptive approach (IAA) [16] and the sparse learning via iterative minimization (SLIM) algorithm [17] have also been investigated for high-resolution spectral estimation (see, e.g., [18]–[22]), with both algorithms showing excellent performance for both complete or incomplete data sets. Regrettably, both algorithms are computationally intensive, and there has as a result been several works on how to form computationally efficient 1-D and 2-D implementations of these estimates for uniformly and non-uniformly sampled data sequences [22]–[26]. These implementations are formed exploiting the methods' inherent low displacement ranks, together with the development of suitable Gohberg-Semencul (GS) representations, as well as making use of Levinson-style and/or (possibly preconditioned) conjugate gradient (CG) solvers of the resulting linear systems of equations. In this paper, we further this work by combining and improving on the earlier presented implementations, including extending the segmented IAA (SIAA) algorithm introduced in [21] and the approximative quasi-Newton preconditioning CG algorithm developed for 1-D data sequences in [26] to 2-D data sets. Furthermore, we introduce a combined IAA-MAP algorithm as well as a hybrid IAA- and SLIM-based estimation scheme.

The remainder of this paper is organized as follows. In the following section, we briefly review the 2-D IAA and SLIM algorithms, respectively. Then, in Section III, we present improved efficient implementations of the IAA algorithm, followed in Section IV with similar improved techniques for the SLIM algorithm. In Section V, we then introduce several hybrid estimation schemes, whereas Section VI presents the performance of the proposed implementations on both simulated and measured SAR data sets. Finally, the paper is concluded in Section VII.
II. THE 2-D IAA AND SLIM ALGORITHMS

Let \( y(n_1, n_2) \) denote the 2-D phase history data of interest, and organize the data in a column-wise form, introducing

\[
\begin{align}
Y_{N_1, N_2} &= \begin{bmatrix} y_{N_1}(0) & \ldots & y_{N_1}(N_2 - 1) \end{bmatrix} \tag{1} \\
y_{N_1}(n_2) &= \begin{bmatrix} y(0, n_2) & \ldots & y(N_1 - 1, n_2) \end{bmatrix}^T \tag{2}
\end{align}
\]

where \( n_1 = 0, 1, \ldots, N_1 - 1 \) and \( n_2 = 0, 1, \ldots, N_2 - 1 \). Moreover, let \( y_{N_1, N_2} = \text{vec}(Y_{N_1, N_2}) \) and \( Y_{N_1, N_2} = \text{mat}\{y_{N_1, N_2}\} \), where \( \text{vec}\{\cdot\} \) denotes column-wise vectorization, and \( \text{mat}\{\cdot\} \) the inverse operation, recreating the matrix from the vectorized matrix. Furthermore, define the 2-D frequency vector

\[
f_{N_1, N_2}(\omega_1, \omega_2) \triangleq f_{N_2}(\omega_2) \otimes f_{N_1}(\omega_1), \tag{3}
\]

where \( \otimes \) denotes the Kronecker product, and \( f_{N}(\omega) \triangleq [1 \ e^{j\omega} \ \ldots \ e^{j(N-1)\omega}]^T \). The 2-D data model can then be written as

\[
y_{N_1, N_2} = F_{N_1, N_2, K_1, K_2} \alpha_{K_1, K_2} + e_{N_1, N_2} \tag{4}
\]

where

\[
F_{N_1, N_2, K_1, K_2} \triangleq \begin{bmatrix} f_{N_1, N_2}(\omega_1, \omega_0) & \ldots & f_{N_1, N_2}(\omega_{K_1-1}, \omega_{K_2-1}) \end{bmatrix}
\]

is composed by the 2-D frequency vectors of interest, and

\[
\alpha_{K_1, K_2} \triangleq [\alpha(\omega_0, \omega_0) \ \ldots \ \alpha(\omega_{K_1-1}, \omega_{K_2-1})]^T \tag{5}
\]

contains the complex amplitudes associated with each 2-D frequency pair \((\omega_k, \omega_k)\), whereas \( e_{N_1, N_2} \) denotes an additive noise. As detailed in [16], the IAA estimate of \( \alpha_{K_1, K_2} \) is formed as the estimate minimizing

\[
\min_{\alpha_{K_1, K_2}} \left\| y_{N_1, N_2} - \tilde{y}_{N_1, N_2}(\omega_k, \omega_k) \right\|^2 \left( R_{N_1, N_2}(\omega_k, \omega_k) \right)^{-1} \tag{6}
\]

over \( \alpha(\omega_k, \omega_k) \), with \( (\omega_k, \omega_k) \), for \( k = 0, 1, \ldots, K_1 - 1 \) and \( k = 0, 1, \ldots, K_2 - 1 \), denoting the 2-D frequency grid of interest, typically with \( K_1 > N_1 \) and \( K_2 > N_2 \), where

\[
\tilde{y}_{N_1, N_2}(\omega_k, \omega_k) \triangleq \alpha(\omega_k, \omega_k) f_{N_1, N_2}(\omega_k, \omega_k) \tag{7}
\]

and

\[
R_{N_1, N_2}(\omega_k, \omega_k) \triangleq R_{N_1, N_2} - \tilde{y}_{N_1, N_2}(\omega_k, \omega_k) \tilde{y}_{N_1, N_2}^H(\omega_k, \omega_k) \tag{8}
\]

is the noise plus interference covariance matrix, whereas the data covariance matrix is estimated as

\[
R_{N_1, N_2} \triangleq F_{N_1, N_2, K_1, K_2}^H D_{K_1, K_2} F_{N_1, N_2, K_1, K_2} = \sum_{k_1=0}^{K_1-1} \sum_{k_2=0}^{K_2-1} \tilde{y}_{N_1, N_2}(\omega_k, \omega_k) \tilde{y}_{N_1, N_2}^H(\omega_k, \omega_k) \tag{9}
\]

with \( D_{K_1, K_2} = \text{diag}\{ \left| \alpha(\omega_0, \omega_0) \right|^2, \ldots, \left| \alpha(\omega_{K_1-1}, \omega_{K_2-1}) \right|^2 \} \), \( |x|^2 \triangleq x^H Ax \), and \( (\cdot)^H \) denote the conjugate transpose. Minimizing (6) with respect to \( (\alpha(\omega_k, \omega_k)) \) yields

\[
\alpha(\omega_k, \omega_k) = \frac{f_{N_1, N_2}^H(\omega_k, \omega_k) \left[ R_{N_1, N_2}^*(\omega_k, \omega_k) \right]^{-1} y_{N_1, N_2}}{f_{N_1, N_2}^H(\omega_k, \omega_k) \left[ R_{N_1, N_2}^*(\omega_k, \omega_k) \right]^{-1} f_{N_1, N_2}(\omega_k, \omega_k)} \tag{10}
\]

The 2-D IAA algorithm is then formed by iterating

\[
\alpha(\omega_k, \omega_k) = \frac{f_{N_1, N_2}^H(\omega_k, \omega_k) R_{N_1, N_2}^{-1} y_{N_1, N_2}}{f_{N_1, N_2}^H(\omega_k, \omega_k) R_{N_1, N_2}^{-1} f_{N_1, N_2}(\omega_k, \omega_k)} \triangleq \frac{\psi(\omega_k, \omega_k)}{\varphi(\omega_k, \omega_k)} \tag{11}
\]

until convergence, where (9) has been derived from (8) using the matrix inversion lemma, bypassing the need of computing \( R_{N_1, N_2}^{-1}(\omega_k, \omega_k) \) for each 2-D frequency pair \((\omega_k, \omega_k)\).

Typically, \( R_{N_1, N_2} \) is initialized to the identity matrix \( I_{N_1, N_2} \).

The computational cost of the resulting 2-D IAA algorithm using brute force is approximately \( N_1^2 N_2^2 + (2N_1^2 N_2^2 + N_1 N_2)K_1 K_2 \) operations.

The SLIM algorithm introduced in [17] is instead formed by minimizing the regularized cost function

\[
N_1 N_2 \log(\eta) + \frac{1}{\eta} |y_{N_1, N_2} - \tilde{y}_{N_1, N_2}|^2 + \frac{2}{\eta} \sum_{k_1=0}^{K_1-1} \sum_{k_2=0}^{K_2-1} (|\beta(\omega_k, \omega_k)|^q - 1)^2 \tag{12}
\]

wrt the system parameters \( \beta_{N_1, N_2} \), formed as

\[
\beta_{K_1, K_2} \triangleq [\beta(\omega_0, \omega_0), \ldots, \beta(\omega_{K_1-1}, K_2-1)]^T, \tag{13}
\]

and the covariance noise variable \( \eta \) of the underline data model

\[
\begin{align}
\eta_{N_1, N_2} &= y_{N_1, N_2} + e_{N_1, N_2} \tag{14} \\
\eta_{N_1, N_2} &\triangleq F_{N_1, N_2, K_1, K_2} \beta_{K_1, K_2} \tag{15}
\end{align}
\]

and with \( 0 < q \leq 1 \). Summarizing, the 2-D SLIM algorithm is formed by iterating [17]

\[
\beta_{K_1, K_2} = \mathbf{P}_{K_1, K_2} F_{N_1, N_2, K_1, K_2}^H \mathbf{S}_{N_1, N_2}^{-1} y_{N_1, N_2} \tag{16}
\]

\[
\mathbf{S}_{N_1, N_2} = F_{N_1, N_2, K_1, K_2} \mathbf{P}_{K_1, K_2} F_{N_1, N_2, K_1, K_2}^H + \eta I_{N_1, N_2} \tag{17}
\]

until practical convergence, with

\[
\mathbf{P}_{K_1, K_2} \triangleq \text{diag}\{|\beta(\omega_0, \omega_0)|^{q-2}, \ldots, |\beta(\omega_{K_1-1}, \omega_{K_2-1})|^{q-2}\} \tag{18}
\]

where typically, \( \beta_{K_1, K_2} \) is initialized by the 2-D DFT of \( y_{N_1, N_2} \), i.e.,

\[
\beta_{K_1, K_2} = F_{N_1, N_2, K_1, K_2} y_{N_1, N_2} \tag{19}
\]

whereas \( \eta \) is initialized as

\[
\eta = \lambda \frac{1}{N_1 N_2} |y_{N_1, N_2} - F_{N_1, N_2, K_1, K_2} \beta_{K_1, K_2}|^2 \tag{20}
\]

where \( \lambda \) is a positive scaling factor.
III. Fast Computation of the 2-D IAA Algorithm

Given the high complexity required to form the 2-D IAA estimate, we proceed to examine ways to form computationally efficient implementations of the algorithm. We first briefly review the efficient 2-D IAA implementation recently developed independently in [24], [25], since this implementation will be the backbone of the schemes then presented herein. In this implementation, the computational reduction is achieved by making use of the inherently low displacement rank of the underlying data covariance matrix, whose Toeplitz block Toeplitz (TBT) structure allows for fast matrix inversion and matrix vector multiplication, which together with the fast computation of the relevant dependent trigonometric polynomials (see also [27]) result in computational schemes several orders of magnitude faster than that of the direct (brute force) implementation. We then proceed to note that further computational savings can be achieved by applying a segmented data procedure, wherein the original image is first segmented into several possibly overlapped parts, each of which is subsequently processed by the IAA algorithm. Finally, reminiscent to the fast approximative implementation of the 1-D IAA algorithm presented in [26], where the data covariance matrix is approximated by a lower order representation formed using a lower order autoregressive (AR) model, an approximative 2-D IAA algorithm and its efficient implementation are also presented. As the 2-D frequency vector (3) is defined over a uniformly spaced grid of frequencies \( (\omega_1, \omega_2) \triangleq (2\pi k_1/K_1, 2\pi k_2/K_2) \), where \( k_1 = 0, 1, \ldots, K_1 - 1 \) and \( k_2 = 0, 1, \ldots, K_2 - 1 \), the 2-D covariance matrix \( R_{N_1 N_2} \) is defined by (10), a TBT matrix of the form

\[
R_{N_1 N_2} = \begin{bmatrix}
R_{0}^{1} & R_{1}^{0}H & \ldots & R_{N_2 - 2}^{1H} \\
R_{1}^{1} & R_{0}^{2} \quad \ldots & R_{N_2 - 2}^{1H} \\
\vdots & \vdots \quad \ddots & \vdots \\
R_{N_2}^{1} & R_{N_2 - 1}^{1} & \ldots & R_{N_2}^{N_1 - 1}
\end{bmatrix}
\]

where the matrix entries \( R_{\ell}^{\ell} \), for \( \ell = 0, 1, \ldots, N_2 - 1 \), are Toeplitz matrices of size \( N_1 \times N_1 \). As shown in [23]–[25], \( R_{N_1 N_2} \) may be extracted from a circulant block circulant (CBC) matrix of higher dimensions as (see also [28])

\[
S_{K_1 K_2} \begin{bmatrix} R_{N_1 N_2} \times \times \end{bmatrix} = W_{K_1 K_2} = D_{K_1 K_2} W_{K_1 K_2},
\]

where \( W_{K_1 K_2} \) denotes the 2-D discrete Fourier Transform (DFT) matrix and \( S_{K_1 K_2} \) is a suitable permutation matrix. The TBT structure of \( R_{N_1 N_2} \) allows for a low displacement rank representation which results in efficient matrix inversion and fast matrix vector multiplication used in the sequel for solving the linear system of equations that appears in the numerator of (9) and defined for further use as \( d_{N_1 N_2} \triangleq R_{N_1 N_2}^{-1} \), as well as for the efficient computation of the coefficients of the 2-D polynomial \( \varphi(\omega_1, \omega_2) \) that appears in the denominator of (9). Since (10) is a TBT matrix, it may be partitioned as

\[
R_{N_1 N_2} = \begin{bmatrix}
R_{N_1}^{N_1(N_2 - 1)} & R_{N_2 - 1}^{b} \\
R_{N_2 - 1}^{H} & R_{N_1}^{N_1(N_2 - 1)}
\end{bmatrix} = \begin{bmatrix}
R_{N_1}^{0} & R_{N_1}^{N_1(N_2 - 1)} \\
R_{N_2}^{H} & R_{N_1}^{N_1(N_2 - 1)}
\end{bmatrix}
\]

where \( R_{N_2 - 1}^{-1} \) and \( R_{N_2 - 1}^{-1} \) denote block matrices of dimensions \( N_1(N_2 - 1) \times N_1 \). Applying the matrix inversion lemma for partitioned matrices to (21) yields (see, e.g., [29])

\[
R_{N_1 N_2}^{-1} = \begin{bmatrix}
R_{N_1}^{N_1(N_2 - 1)} & 0 \\
0 & 0 \\
0 & 0
\end{bmatrix} + B_{N_2} B_{N_2}^{H}
\]

\[
B_{N_2} = \begin{bmatrix} B_{N_2 - 1} & 0 \end{bmatrix}
\]

\[
\tilde{A}_{N_2} = \begin{bmatrix} A_{N_2 - 1} \end{bmatrix}
\]

\[
\tilde{A}_{N_2} = \begin{bmatrix} 0 \\
0 & 0 \\
0 & 0
\end{bmatrix} + \tilde{A}_{N_2} \tilde{A}_{N_2}^{H}
\]

where \( B_{N_2} \) and \( \tilde{A}_{N_2} \) are block matrices of dimensions \( N_1 N_2 \times N_1 \) defined by

\[
B_{N_2} = \begin{bmatrix} B_{N_2 - 1} & 0 \end{bmatrix}
\]

\[
\tilde{A}_{N_2} = \begin{bmatrix} A_{N_2 - 1} \end{bmatrix}
\]

\[
A_{N_2 - 1} = -R_{N_1(N_2 - 1)} B_{N_2 - 1}^{f}
\]

\[
A_{N_2 - 1} = -R_{N_1(N_2 - 1)} B_{N_2 - 1}^{f}
\]

\[
A_{N_2} = R_{N_2}^{0} + R_{N_2}^{(H)} A_{N_2 - 1}
\]

\[
A_{N_2} = R_{N_2}^{0} + R_{N_2}^{(H)} A_{N_2 - 1}
\]

with \( A_{N_2}^{f} \) and \( A_{N_2}^{f} \) denoting the Cholesky factors of \( A_{N_2} \) and \( A_{N_2} \), respectively. Define a block shifting matrix, \( Z_{N_1 N_2} \), as \( Z_{N_1 N_2} = Z_{N_2} \otimes I_{N_1} \), where

\[
Z_{N_2} = \begin{bmatrix} 0 \times \times \end{bmatrix}
\]

Clearly, \( Z_{N_1 N_2} \) is a block digital representation with respect to the lower shifting block matrices \( Z_{N_1 N_2} \) and \( Z_{N_1 N_2} \) takes the form [30]

\[
\nabla_{Z_{N_1 N_2}} (R_{N_1 N_2}^{-1}) \triangleq A_{N_2} A_{N_2}^{H} - Z_{N_1 N_2} B_{N_2} B_{N_2}^{H} Z_{N_1 N_2}^{-1}
\]

resulting a suitable GS factorization of \( R_{N_1 N_2}^{-1} \) of the form

\[
R_{N_1 N_2}^{-1} = \sum_{i=1}^{2} \sigma_{i} L_{(\nabla_{N_1 N_2})}^{2} L_{(\nabla_{N_1 N_2})}
\]

where \( \sigma_{i} = 1, \sigma_{2} = -1, T_{N_1 N_2}^{1} \triangleq A_{N_2}, \) and \( T_{N_2}^{2} \triangleq Z_{N_2}^{2} B_{N_2}^{-H} Z_{N_1 N_2}^{-1} \) denoting a block lower Toeplitz matrix of block dimensions \( N_2 \times N_2 \), having block entries of size \( N_1 \times N_1 \) each. Thus, the matrix-vector or matrix-matrix products involving (32) can be organized using the block DFT, and its fast implementation computed using the FFT [31], provided that the generator block matrices \( T_{N_2}^{2} \) and \( T_{N_2}^{2} \) are available. The latter can be efficiently estimated using the celebrated Levinson-Whittle-Wiggins-Robinson (LWWR) algorithm (see, e.g., [32]). The computational complexity of the LWWR algorithm is approximately \( C_{LWWR} = 1.5N_2^2 N_2^2 + 4N_2^2 N_2 \). Moreover, due to the persymmetric property that each Toeplitz matrix entry possesses, i.e., as \( J_{N_1 N_2} R_{N_1 N_2} J_{N_1 N_2} = R_{N_2}^{H} \), where the block exchange matrix, \( J_{N_1 N_2} \), is defined as a block anti-diagonal matrix with the exchange matrix \( J_{N_1 N_2} \) along the (block) anti-diagonal, it holds that [33] \( A_{N_2}^{-1} = J_{N_1 N_2} B_{N_2}^{-H} J_{N_1 N_2} \) and \( A_{N_2}^{-1} = J_{N_1 N_2} B_{N_2}^{-H} J_{N_1 N_2} \), which may be used for further reduction of the computational cost (although the resulting scheme may then be more sensitive to the numerical implementation). Using the LWWR algorithm
for the computation of the displacement representation of (32), \( \mathbf{d}_{N_1 N_2} \) may be computed at a cost of approximately 2\( N_1^2 \phi(2N_2) + 8N_1^2 N_2 \) operations, where \( \phi(2N_2) \) denotes the cost of performing a 1-D FFT (IFFT) of size \( 2N_2 \times 1 \), since the GS factorization (32) involves products of block lower Toeplitz matrices, thus allowing for a fast matrix vector multiplication using the FFT. Finally, the coefficients of the 2-D trigonometric polynomial \( \varphi(\omega_{k_1}, \omega_{k_2}) \) that appears in the denominator of (9) are computed using the GS representation in (32) and the fast scheme developed in [27]. The resulting fast IAA (FIAA) implementation, presented independently in [24], [25], and for the reader's convenience summarized in Table I, requires roughly

\[
C^{\text{FIAA}} \approx m_I \left[ 1.5N_2^2 N_3^3 + 4N_1^2 \phi(2N_2) + 8N_1^2 N_2 + 5N_1 \phi(2N_1, 2N_2) + 3\phi(K_1, K_2) \right],
\]

(33) 

operations, where \( m_I \) is the number of 2-D IAA iterations and \( \phi(K_1, K_2) \) denotes the cost of performing a 2-D FFT (IFFT) of size \( K_1 \times K_2 \).

Motivated by the segmented IAA (SIAA) algorithm introduced in [21], which allows for a trade-off between variance and bias of the estimate, we proceed to develop a 2-D SIAA algorithm, wherein the 2-D data set is divided into \( L \) possible overlapping segments of size \( N_1 \times N_2 \) each, with \( 0 < N_1, N_2 < N_2 \). The 2-D SIAA algorithm is then formed by iterating

\[
\begin{align*}
\alpha^{(\ell)}(\omega_{k_1}, \omega_{k_2}) & = \frac{\mathbf{f}_{S_1 S_2}^{H}(\omega_{k_1}, \omega_{k_2}) \mathbf{R}_{S_1 S_2}^{-1}(\ell) \mathbf{y}_{S_1 S_2}(\ell)}{\mathbf{f}_{S_1 S_2}^{H}(\omega_{k_1}, \omega_{k_2}) \mathbf{R}_{S_1 S_2}^{-1}(\ell) \mathbf{f}_{S_1 S_2}(\omega_{k_1}, \omega_{k_2})} \\
& \triangleq \psi^{(\ell)}(\omega_{k_1}, \omega_{k_2}), \quad \ell = 1, 2, \ldots L \\
\mathbf{R}_{S_1 S_2} & = \mathbf{F}_{S_1 S_2 K_1 K_2} \mathbf{D}_{S_1 S_2 K_1 K_2}^{H} \mathbf{F}_{S_1 S_2 K_1 K_2}^{H}
\end{align*}
\]

(34) 

until practical convergence, with

\[
\mathbf{D}_{S_1 S_2 K_1 K_2} \triangleq \text{diag} \left\{ \Phi^{0}(\omega_{0}, \omega_{0}) \ldots \Phi^{(L)}(\omega_{K_1-1}, \omega_{K_2-1}) \right\},
\]

where

\[
\Phi^{(L)}(\omega_{k_1}, \omega_{k_2}) = \frac{1}{L} \sum_{\ell=1}^{L} |\alpha^{(\ell)}(\omega_{k_1}, \omega_{k_2})|^2
\]

(36) 

is the averaged spectra at \( (\omega_{k_1}, \omega_{k_2}) \) over all segments, with \( y_{S_1 S_2}(\ell) \triangleq \text{vec} \left\{ \mathbf{y}_{S_1 S_2}(\ell) \right\} \) denoting the vectorized data corresponding to the \( \ell \)-th image segment \( \mathbf{y}_{S_1 S_2}(\ell) \) of size \( S_1 \times S_2 \). The 2-D SIAA can be efficiently implemented using a similar approach as in the case of the 2-D FIAA algorithm, although in this case, the vectors \( \mathbf{d}_{S_1 S_2}(\ell) \triangleq \mathbf{R}_{S_1 S_2}^{-1}(\ell) \mathbf{y}_{S_1 S_2}(\ell) \), for \( \ell = 1, 2, \ldots L \), that appear in the numerator of (34) are computed using the GS factorization of \( \mathbf{R}_{S_1 S_2} \). Thus, the overall complexity of the 2-D fast SIAA (FSIAA) algorithm, including the dominant factors only, is given by

\[
C^{\text{FSIAA}} = m_{S_1} \left[ 1.5S_2^2 N_3^3 + 5S_1 \phi(2S_1, 2S_2) + 4S_2^2 \phi(2S_2) \right] 6LS_1 \phi(2S_2) + (2L + 1) \phi(K_1, K_2),
\]

(37) 

where \( m_{S_1} \) is the number of 2-D SIAA iterations.

The above discussed 2-D FIAA and FSIAA implementations form exact implementation of the corresponding brute force algorithms, and although substantially faster than the brute force implementations, these can still be computationally prohibitive for some applications. For this reason, we recently introduced a fast approximative CG-based 1-D IAA algorithm in [23]. This algorithm has also been extended to a block-recurse (1-D) formulation applied to blood velocity estimation in ultrasound imaging [34]. Here, we further extend on this work, allowing the algorithm to also handle 2-D data sets. The resulting implementation is substantially more efficient than even the above fast implementations, without having more than a marginal effect in the accuracy of the estimated parameters. The implementation is motivated by the Quasi-Newton (QN) algorithm formulated in [35], and then further developed in [36]–[39], wherein an efficient implementation scheme of approximate recursive least squares algorithms is formed by imposing a low order AR approximation on the input signal of the adaptive algorithm. In the spectral estimation case, this concept may be exploited by constructing a 2-D QN approximation of the covariance matrix by extrapolating a lower order, incomplete, solution of the 2-D linear system for the full size data matrix under consideration. The thus proposed 2-D QN-IAA algorithm implicitly estimates the approximative covariance matrix \( \mathbf{Q}_{N_1 N_2} \) in place of \( \mathbf{R}_{N_1 N_2} \), such that

\[
\mathbf{Q}_{N_1 N_2}^{-1} = \begin{bmatrix} 0 & 0^T \\ 0 & \mathbf{R}_{N_1 M_2}^{-1} \end{bmatrix} + \mathbf{A}_{2D} \mathbf{A}_{2D}^H
\]

(37) 

where \( \mathbf{A}_{2D} \triangleq [\mathbf{A}_{N_1}^{Q} \mathbf{Z}_{N_2} \mathbf{A}_{N_1}^{Q} \ldots \mathbf{Z}_{N_2-M_2} \mathbf{A}_{N_1}^{Q}] \) with \( \mathbf{A}_{N_1}^{Q} = [\mathbf{A}_{M_2}^{Q} \mathbf{0}_{(N_2-M_2) \times M_2}^T] \), where \( \mathbf{A}_{2D} \) is a block Toeplitz matrix of block size \( N_2 \times (N_2-M_2) \) with matrix entries of size \( N_1 \times N_1 \). Here, (37) results from an incomplete 2-D LWWR algorithm where, by construction, the 2-D forward and backward matrix valued reflection coefficient are set equal to zero, \( \mathbf{K}_{N_1}^{(L)} = 0 \) and \( \mathbf{K}_{N_1}^{(L)} = 0 \), for \( \ell = M_2 + 1, M_2 + 2, \ldots, N_2 \). Thus, an approximate 2-D IAA algorithm can be derived by the direct use of the matrix \( \mathbf{Q}_{N_1 N_2}^{-1} \) in place of \( \mathbf{R}_{N_1 N_2} \) that appears in (9). Since the inverse \( \mathbf{Q}_{N_1 N_2}^{-1} \) is already available, no further computations are required for this purpose. Thus, the resulting
approximate 2-D IAA algorithm is formed by iterating

$$
\tilde{\alpha}(\omega_1, \omega_2) = \frac{f_{N_1N_2}^H(\omega_1, \omega_2)Q_{N_1N_2}^{-1}Y_{N_1N_2}}{f_{N_1N_2}^H(\omega_1, \omega_2)Q_{N_1N_2}^{-1}f_{N_1N_2}(\omega_1, \omega_2)}
\triangleq \hat{\varphi}(\omega_1, \omega_2)
$$

(38)

$$
R_{N_1M_2} = F_{N_1M_2,K_1K_2}D_{K_1K_2}F_{N_1M_2,K_1K_2}^H
$$

(39)

until practical convergence, where

$$
\tilde{D}_{K_1K_2} \triangleq \text{diag}\left\{ |\tilde{\alpha}(\omega_0, \omega_0)|^2 \ldots |\tilde{\alpha}(\omega_{K_1-1}, \omega_{K_2-1})|^2 \right\} .
$$

The resulting 2D QN-IAA algorithm can be implemented efficiently using the techniques developed above, although the variable $\tilde{d}_{N_1N_2} \triangleq Q_{N_1N_2}^{-1}y_{N_1N_2}$ that appears in the numerator of (38) is now computed using (37) as

$$
\hat{d}_{N_1N_2} = \begin{bmatrix} 0 & 0^T \\ 0 & R_{N_1M_2}^{-1} \end{bmatrix} y_{N_1N_2} + A_{2D}A_{2D}^H y_{N_1N_2}
$$

(40)

which can efficiently be implemented using the GS factorization of $R_{N_1M_2}^{-1}$ and the fact that the matrix $A_{2D}$ is block Toeplitz, at a cost of $2N_1^2\phi(2M_2) + 8N_1^2M_2 + N_2^2\phi(2M_2) + 2N_1^2N_2$ operations. Moreover, $\hat{\varphi}(\omega_1, \omega_2)$ that appears in the denominator of (38) can be expressed as

$$
\hat{\varphi}(\omega_1, \omega_2) = \frac{f_{N_1M_2}^H(\omega_1, \omega_2)Q_{N_1M_2}^{-1}f_{N_1M_2}(\omega_1, \omega_2)}{(N_2 - M_2) |f_{N_1M_2}^H(\omega_1, \omega_2)\hat{A}_{M_2}|^2}
$$

(41)

allowing for a reduction in the computational cost for the estimation of the coefficients of the 2D trigonometric polynomial. The overall computational complexity of the 2-D QN-FIAA spectral estimation algorithm is given by

$$
C_{QN-FIAA} \approx m \left[ 1.5N_1^3M_2^2 + 2N_1^2\phi(2M_2) + 8N_1^2M_2 + N_1^2\phi(2M_2) + 2N_1^2N_2 + 2N_1^2\phi(2M_2) + 2N_1^2\phi(N_1) + 5N_1\phi(2N_1, 2M_2) + 3\phi(K_1, K_2) \right],
$$

(42)

where $m$ is the number of 2-D QN-IAA iterations. Further computational savings can be achieved by noting that, in most situations $M_2 \ll N_1$, implying that the application of suitable permutations allows us to construct an equivalent matrix defined as $R_{M_2N_1} = S R_{N_1M_2} S^T$, where $S$ is a permutation matrix and $R_{M_2N_1}$ is a block Toeplitz matrix with block size $N_1 \times N_1$, having Toeplitz matrix entries of size $M_2 \times M_2$. Thus, $R_{N_1M_2}^{-1} = S^T \hat{R}_{M_2N_1}^{-1} S$, implying that the generators of the inverse matrix may be obtained using the LWWR algorithm applied to block matrices with entries of size $M_2 \times M_2$. The new generators are thus “thinner” than the previous ones, having width $M_2$ instead of $N_2$, and thus allowing for a more efficient implementation, requiring only $1.5M_2^2N_1\phi$ operations, which is notably less than the earlier required $1.5N_1^2M_2^2$ operations, especially when $M_2 \ll N_1$. The block GS representation in (32) is then restructured accordingly to accommodate thinner block matrices, implying that the first product in (40) can be organized as

$$
\begin{bmatrix} 0 & 0^T \\ 0 & S^T \hat{R}_{M_2N_1}^{-1} S \end{bmatrix} y_{N_1M_2}
$$

resulting in a cost of $2M_2^2\phi(2N_1) + 8M_2^2N_1\phi$ operations. Finally, we examine how the forward predictor required in the construction of $\tilde{A}_{M_2}$ can be computed. Recall that

$$
R_{N_1M_2} \tilde{A}_{M_2} = \begin{bmatrix} I_{N_1} \\ 0 \end{bmatrix}
$$

(43)

or

$$
\tilde{R}_{M_2N_1} \hat{S} \tilde{A}_{M_2} = \hat{S} \begin{bmatrix} I_{N_1} \\ 0 \end{bmatrix}
$$

(44)

which implies

$$
\tilde{A}_{M_2} = S^T \hat{R}_{M_2N_1}^{-1} \hat{S} \begin{bmatrix} I_{N_1} \\ 0 \end{bmatrix}
$$

(45)

Given the GS representation of $\tilde{R}_{M_2N_1}$, this step can be accomplished in $4N_1^2M_2^2 + 3M_2N_1\phi(2N_1)$ operations, and finally $\hat{A}_{M_2} = \hat{A}_{M_2}[A_{M_2}^{-1}]^{1/2}$, where $[A_{M_2}^{-1}]^{1/2}$ denotes the Cholesky factor of $A_{M_2}$. Then, in total, the computational complexity of this alternative 2-D QN IAA implementation is

$$
C_{QNFIAA} \approx m \left[ 1.5M_2^2N_1^2 + N_1^3M_2^2 + 4N_1^2M_2^2 + 3M_2N_1\phi(2N_1) + 2N_1^2\phi(N_1) + 5N_1\phi(2N_1, 2M_2) + 3\phi(K_1, K_2) \right].
$$

Summarizing our analysis, the fast implementation of the 2-D IAA algorithms discussed so far may be computed at a cost of approximately

$$
C_{QNFIAA} \approx 1.5N_1^3N_2^2 + 1.5K_1K_2 \log_2(K_1K_2)
$$

$$
C_{QNFIAA-I} \approx 1.5N_1^3M_2^2 + 1.5K_1K_2 \log_2(K_1K_2)
$$

$$
C_{QNFIAA-II} \approx 1.5N_1^3M_2^2 + 1.5K_1K_2 \log_2(K_1K_2)
$$

operations per IAA iteration. To get some insight into the above expressions, consider the special, yet common, case where $N \triangleq N_1 = N_2$, $K \triangleq K_1 = K_2$, and let $M_2 = N_1/3$, which result in $C_{QNFIAA} \approx 1.5N_1^5 + 3K_2^2 \log_2(K)$, $C_{QNFIAA-I} \approx 1/6N_1^5 + 3K^2 \log_2(K)$, and $C_{QNFIAA-II} \approx 1/18N_1^5 + 3K^2 \log_2(K)$, respectively. On the other hand, the computational complexity of the 2-D SFI AA algorithm with $L = 5$ overlapped segments, each of size equal to the one fourth of original image, i.e., $S_1 = S_2 = N/2$, reduces to $C_{SFI AA} \approx 3N_1^5/64 + 11K^2 \log_2(K)$. It is worth noting that the steps of the algorithm can to a large extent be parallelized. The algorithm essentially requires two basic processing components that solves the 2-D TBT linear system and computes the 2-D FFT. The latter can be implemented in parallel using a bank of 1-D FFT units, while the former, which is by far the most computational demanding unit of the algorithm, allows for a parallel implementation using a Schur-type implementation [40]–[44]. Such an implementation avoids the inner product computations inherently involved in the LWWR recursions, thereby allowing for a parallel implementation with locally recursive algorithms via a transformation using a canonical mapping methodology [45]–[47], allowing for an efficient array implementation via a systolic or a waveform architecture. The reader is referred to [40]–[49] for a further discussion on these aspects.
TABLE II
SUMMARY OF THE FAST 2-D SLIM ALGORITHM

\[
\begin{align*}
\mathbf{P}_{K_1K_2} & = \text{diag} \left\{ |\beta(\omega_0, \omega_0)|^2, \ldots, |\beta(\omega_{K_1-1}, \omega_{K_2-1})|^2 \right\} \\
\mathbf{C}_{K_1K_2}^H & = \mathbf{W}_{K_1K_2}^H \mathbf{P}_{K_1K_2} \mathbf{W}_{K_1K_2} \\
\mathbf{C}_{K_1K_2} & = \mathbf{S}_{K_1K_2} ^ \times \left[ \Sigma_{N_1N_2}^1 \times \Sigma_{N_1N_2}^2 \right] \mathbf{S}_{K_1K_2} ^ \times \\
\Sigma_{N_1N_2} & = \Sigma_{N_1N_2} + \eta \mathbf{I}_{N_1N_2} \\
\mathbf{r}_{N_1N_2} & = \mathbf{y}_{N_1N_2} - \mathbf{\Sigma}_{N_1N_2} \mathbf{d}_{N_1N_2} \\
\rho_0 & = |\mathbf{r}_{N_1N_2}|^2, \quad k = 1 \\
\text{while } \sqrt{\mathbf{r}_k^H \mathbf{r}_k} > \epsilon |\mathbf{y}_{N_1N_2}|^2 \quad \text{and } \kappa < \kappa_{\text{max}} \\
\gamma & = \rho_{k-1} / \rho_{k-2} \\
\mathbf{P}_{N_1N_2} & = \mathbf{r}_{N_1N_2} + \gamma \mathbf{P}_{N_1N_2} \\
\mathbf{w}_{N_1N_2} & = \mathbf{\Sigma}_{N_1N_2} \mathbf{P}_{N_1N_2} \\
\delta & = \rho_{k-1} / (|\mathbf{r}_{N_1N_2}|^2 |\mathbf{w}_{N_1N_2}|^2) \\
\mathbf{d}_{N_1N_2} & = \mathbf{d}_{N_1N_2} + \delta \mathbf{P}_{N_1N_2} \\
\mathbf{r}_{N_1N_2} & = \mathbf{r}_{N_1N_2} - \delta \mathbf{w}_{N_1N_2} \\
\rho_k & = |\mathbf{r}_{N_1N_2}|^2, \quad k = k + 1 \\
\mathbf{P}_{K_1K_2} & = \mathbf{P}_{K_1K_2} F_{N_1N_2,K_1K_2}^H \mathbf{d}_{N_1N_2} \\
\eta & = \frac{1}{N_1N_2} |\mathbf{y}_{N_1N_2} - \mathbf{F}_{N_1N_2,K_1K_2}^H \mathbf{d}_{N_1N_2}|^2 \\
\end{align*}
\]

IV. FAST COMPUTATION OF THE 2-D SLIM ALGORITHM

Brute force implementation of the SLIM algorithm described by (15)-(17) requires approximately \( N_1^3 N_2^2 + N_1^2 N_2^2 K_1 K_2 \) operations per iteration, with usually no more than 10-15 iterations necessary for convergence. Fortunately, this often prohibitive computational burden may be substantially reduced when the parameters sought correspond to a unitary \( \mathbf{U} \mathbf{W} \mathbf{V} \mathbf{U}^H \mathbf{V}^H \) solution. The computational cost per iteration can be reduced to approximately

\[
C_{\text{SLIM}} \approx 1.5N_1^3 N_2^2 + 1.5K_1 K_2 \log_2(K_1 K_2)
\]

operations. As it has been pointed out in [17], [22], this figure can be further reduced by using an iterative CG-based linear solver instead of the LWWR algorithm for the solution of the TBT linear system involved in (15). This is particularly suitable as the SLIM algorithm only requires the solution of the TBT linear system (in contrast to the IAA case where in addition the displacement representation of the TBT matrix is also needed) and as the CG algorithm may be expected to converge faster than it is anticipated by the dimensionality of the TBT matrix, since, due to the assumptions of the line spectral model under consideration, the rank of (16) can be expected to be relatively small. The resulting 2-D CG-SLIM algorithm is summarized in Table II, having a computational complexity of approximately

\[
C_{\text{SLIM-CG}} \approx \kappa_{\text{CG}} (5N_1 N_2 + 2\phi(2N_1, 2N_2)) + 3\phi(K_1, K_2) + \phi(2N_1, 2N_2) \approx \kappa_{\text{CG}} 4N_1 N_2 \log_2(4N_1 N_2)
\]

operations per SLIM iteration, provided that the TBT vector multiplications are computed using circular embedding and the 2-D FFT, with \( \kappa_{\text{CG}} \) denoting the number of CG iterations required for convergence. Compared to the fast CG 2-D SLIM implementation presented in [17], [22] and where fast TBT computations are organized using circular embedding of size equal to the size of the 2-D frequency grid, resulting in a complexity approximately estimated as \( \kappa_{\text{CG}} K_1 K_2 \log_2(K_1 K_2) \), the proposed approach is faster, yet mathematically equivalent. For a typical 2-D spectral estimation scenario, where \( N = N_1 = N_2 \) and \( K = K_1 = K_2 \), and where \( K = 5N \), the proposed implementation requires about 6 times less computations than that of the previously presented approach; the larger \( K \) is compared to \( N \), the larger the gain is, with the overall complexity of the fast 2-D SLIM implementations being about

\[
C_{\text{SLIM-GS}} \approx 1.5N^5 + 1.5K^2 \log_2(K) \\
C_{\text{SLIM-CG}} \approx \kappa_{\text{CG}} 8N^2 \log_2(4N) + 3K^2 \log_2(K)
\]

per SLIM iteration.

V. HYBRID SPECTRAL ESTIMATION SCHEMES

The aforementioned spectral estimation methods possess various merits and limitations. Below, we consider several hybrid methods that take advantages of these merits while overcoming the limitations of the separate methods. First, we note that the 2-D IAA algorithms will provide a non-parametric, robust, and user parameter free algorithm, which has also been found to be more accurate than the corresponding SLIM estimates, although with a notably higher sidelobe level (see also [22]). In order to achieve sidelobe levels comparable to those of SLIM, one may instead form a combined approach that first apply the 2-D IAA estimate to compute a dense spectral estimate, which is then, upon convergence, followed by a refinement stage formed as

\[
\hat{\alpha}(\omega_{k_1}, k_2) \triangleq |\alpha(\omega_{k_1}, \omega_{k_2})|^2 F_{K_1K_2}^H (\omega_{k_1}, \omega_{k_2}) R_{N_1N_2}^{-1} y_{N_1N_2} \\
\hat{\alpha}(\omega_{k_1}, k_2) \triangleq |\hat{\alpha}(\omega_{k_1}, \omega_{k_2})|^2 F_{K_1K_2}^H (\omega_{k_1}, \omega_{k_2}) Q_{N_1N_2}^{-1} y_{N_1N_2} \\
\]

Since SLIM achieves sparsity based on solving a hierarchical Bayesian model through maximizing the a posteriori probability density function (MAP), this ad hoc step is referred to as a MAP step, and the resulting algorithm as the 2-D IAA-MAP algorithm. Using similar arguments, one may similarly form the 2-D QN-IAA-MAP algorithm by instead using (38)-(39) followed by the MAP step

\[
\hat{\alpha}(\omega_{k_1}, k_2) \triangleq |\hat{\alpha}(\omega_{k_1}, \omega_{k_2})|^2 F_{K_1K_2}^H (\omega_{k_1}, \omega_{k_2}) Q_{N_1N_2}^{-1} y_{N_1N_2} \\
\hat{\alpha}(\omega_{k_1}, k_2) \triangleq |\hat{\alpha}(\omega_{k_1}, \omega_{k_2})|^2 F_{K_1K_2}^H (\omega_{k_1}, \omega_{k_2}) Q_{N_1N_2}^{-1} y_{N_1N_2} \\
\]

The computational effort of performing the MAP step in both cases is negligible, and the resulting algorithms can thus be implemented at cost given by (33) and (42), respectively.
Alternatively, one may note that the 2-D IAA algorithms, being initialized by setting the data covariance matrix $R_{N_1,N_2} = I_{N_1,N_2}$ will result in an initial spectrum which is identical to the (scaled and zero padded) 2-D DFT of the data vector $Y_{N_1,N_2}$. The resulting low resolution and high sidelobe levels may slow down the 2-D IAA algorithm, necessitating several iterations to achieve convergence, while at the same time requiring unnecessary high level of computations in the earlier iterations. This drawback can be circumvented if a more accurate and relatively cheap spectral estimate is in place of the 2-D DFT during the initialization of the IAA. Since both the 2-D SFIAA and the 2-D QN-FIAA are less expensive than the 2-D FIAA algorithm, while at the same time capable of producing spectra of higher quality than that of the 2-D DFT method, a reasonable hybrid method may be formed by exploiting these cheaper and somewhat less accurate estimators, followed by more expensive and accurate estimators at the latter iterations. We term the method combing $m_{si}$ iterations of the 2-D SFIAA algorithm followed by $m_i$ iterations of the 2-D FIAA algorithm the 2-D H-SFIAA($m_{si}$)FIAA($m_i$) scheme. Similarly, the 2-D H-QNFIAAA($m_{qi}$)FIAA($m_i$) scheme may be formed by instead using the 2-D QN-FIAA during the earlier $m_{qi}$ iterations. Finally, we note that the 2-D FIAA, the 2-D SFIAA or the 2-D QN-FIAA can be used for the initialization of the 2-D SLIM recursions in (15)-(17), in place of (18), resulting in similar hybrid IAA/SLIM spectral estimation schemes.

VI. NUMERICAL AND EXPERIMENTAL EXAMPLES

We proceed to examine the performance of the discussed estimators on the simulated Slicy data set and the experimentally measured GOTCHA data set. We begin by examining the 2-D phase-history Slicy data generated at $0^\circ$ azimuth angle using XPATCH [50], a high frequency electromagnetic scattering prediction code for complex 3-D objects. A photo of the Slicy object taken at $45^\circ$ azimuth angle and a SAR image benchmark obtained via FFT from a complete $288 \times 288$ data matrix are shown in Figures 1(a) and 1(b), respectively. In the following, we examine a lower dimensional subset formed using only the $N_1 = N_2 = 80$ center block of the phase-history data, with $K_1 = K_2 = 400$ uniformly spaced 2-D frequency points. Figure 2 shows the SAR images obtained by the aforementioned spectral estimation techniques, including the FFT-based estimate, using 10 SLIM or IAA iterations for the respective methods, or 9 SFIAA (QN-FIAA) iterations followed by one FIAA iteration for the hybrid IAA schemes, as well as 3 additional SLIM-0 iterations at the conclusion of the 10 iterations of the various algorithms for the hybrid SLIM variants1. Furthermore, $L = 5$, $M_2 = 32$, and $\epsilon = 10^{-6}$ for the SFIAA, QN-FIAA and CG-SLIM algorithms, and their corresponding hybrid schemes, respectively. As shown in Figure 2, the FFT is, as expected, found to yield low resolution and high sidelobes, whereas the IAA and SLIM based estimates can be found to result in significantly higher resolution and lower sidelobe levels. As is clear from the figure, the hybrid methods allow for notably sparser estimates, with both the (hybrid) IAA-MAP and hybrid SLIM variants satisfactorily balancing the tradeoff between the image resolution and detail preservation as compared to SLIM-0 and SLIM-1. Table III summarizes the computation times needed by the aforementioned algorithms to form the $K_1 \times K_2$ SAR image from the Slicy data on an ordinary workstation (Intel Xeon E5506 processor 2.13G Hz, 12GB RAM, Windows 7 64-bit, and MATLAB R2010b). As previously mentioned, the hybrid IAA schemes reduce the computation cost significantly with only a slight performance degradation as compared to their FIAA counterpart. It is worth noting that by performing several iterations of SLIM-0 at the conclusion of various algorithms including the (hybrid) IAA algorithms and SLIM-1, the hybrid SLIM schemes can greatly suppress the sidelobe levels without drastically increasing the computation complexities.

We proceed to examine the methods’ performance for the GOTCHA Air Force Research Laboratory data set. The GOTCHA volumetric SAR data set, Version 1.0, consists of SAR phase history data collected at X-band with a 640 MHz bandwidth with full azimuth coverage at eight different elevation angles with full polarization [51]. The imaging scene consists of numerous civilian vehicles and calibration targets, as shown in Figure 3. Here, we examine the performance on the phase history data with full azimuth coverage collected at the first pass for a HH polarization channel of a Chevrolet Malibu, parked in the upper corner of the parking lot as shown in Figure 3. We use $4^\circ$ subapertures from $0^\circ$ to $360^\circ$ with no overlap, which results in a total of 90 subapertures. For each subaperture, one 2-D spatial image is formed by using a 2-D FFT on the corresponding phase history (k-space) data. An $N_1 \times N_2 = 80 \times 80$ block of the spatial data centered about the Chevrolet Malibu is then chipped out and transformed back into k-space using an inverse FFT (IFFT) operation. The discussed spectral estimation techniques are then applied to the so-obtained $N_1 \times N_2$ phase history data to get one $K_1 \times K_2 = 400 \times 400$ image for each subaperture. By using the auxiliary information provided by the GOTCHA data set (e.g., the antenna locations, range to scene center, azimuth and elevation angles), the image is then projected onto the ground plane and interpolated to form a 2-D ground image. The resulting 90 2-D ground images are then combined using the non-coherent max magnitude operator to yield the recon-

1In the examined examples, no significant further improvement was achieved after the specified number of iterations.
Fig. 2. Modulus of the SAR images of the Slicy object obtained from an 80 × 80 data matrix via FFT, (hybrid) IAA (-MAP), and (hybrid) SLIM variants.

Structured Malibu image, whose dimensions are [5, 15] × [−10, 0] meters with grid size 0.05 meters in both dimensions (i.e., forming a 201 × 201 image). Figure 4 illustrates the resulting images for the discussed methods, using the above introduced parameter settings, clearly indicating the superior performance of the introduced algorithms as compared to the FFT based approach. As before, the SLIM-0 estimate can be seen to be too sparse to preserve certain vehicle features, whereas the hybrid methods are found to again provide high resolution images with low sidelobe levels. Table III summarizes the computation times needed by the discussed methods to form the reconstructed Malibu images (the running times start from applying the various algorithms to the 80 × 80 phase history data for each subaperture until the so-obtained 90 subimages are fused to form the final images as shown in Figure 4). Table III illustrates that the computational time ratios of the Gotcha to Slicy data sets for the various algorithms are mostly near or slightly above 90, which is reasonable since, for the Slicy data, one process one 80 × 80 data matrix whereas for the Gotcha data, one process 90 80 × 80 data matrices. The exceptions are the ratios for FFT, SLIM-0, SLIM-1 and H-SLIM1-SLIM0. For FFT, the reason for a ratio much higher than 90 is that the time consumed in the image fusion process dominates the overall running time for Gotcha. For the CG-based SLIM variants, the reason for ratios lower than 90 is that the running time depends largely on the number of total
CG iterations required for convergence for a given accuracy controlled by the error threshold. Take SLIM-0 for example. The ratio of the running time is around 40 and the total number of CG iterations required by SLIM-0 for Gotcha and Slicy is 24774 and 1171, respectively. This approximate ratio of 21 is of CG iterations required by SLIM-0 for Gotcha and Slicy.

### VII. Conclusions

In this work, we have presented fast implementations of the 2-D IAA and SLIM algorithms, exploring the rich internal structure of the estimators. The proposed implementations are found to offer a significantly reduced computational complexity, with the proposed approximative implementations offering even further computational reductions, at the cost of only slight performance degradation. By including a sparsity promoting final step at the conclusion of the iterations, notable sidelobe level reductions are achieved, allowing for a satisfactorily balance between the image resolution and detail preservation. The effectiveness of the algorithms have been verified using both simulated and experimentally measured data sets.

### References


Fig. 4. Comparison of the reconstructed Malibu images obtained by FFT, (hybrid) IAA (-MAP), and (hybrid) SLIM variants.


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