

Distributed State Estimation With Dimension Reduction Preprocessing

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Abstract—System state estimation relies heavily on the measurements. With the advance of sensing technology, the ability to measure is no longer a bottleneck in many systems, and more and more researchers now focus on the rich-information setting, i.e., big data. However, although information never hurts, it does not help unconditionally. How to make the most of it depends on whether we can process the data efficiently. In some systems, the inherent constraint such as the bandwidth and cost makes it necessary to reduce the dimension of the measurement before further processing. The problem that the raw measurements are first preprocessed to reduce size and then used for estimation is addressed in this paper. It is shown that there is a lower bound on the size of the preprocessed data such that if the size is beyond the bound, there exists a closed-form estimator design that the linear minimum mean-square estimation can be obtained. Moreover, we propose an algorithm that is guaranteed to converge to a stationary point to design an estimator in the conditions that the lower bound cannot be reached. Besides convergence, the proposed algorithm guarantees bounded performance loss compared with the global optimal solution under some additional conditions. Finally, simulation results in three different applications are shown to demonstrate the effectiveness of the proposed algorithm.

Index Terms—Distributed state estimation, fusion, LMMSE, lossless compression, optimal lossy compression.

I. INTRODUCTION

THE sensing technology has enabled us to measure the “states” in our daily life in the past few decades, such as monitoring the water distribution [1], railway traffic [2] and power grid [3]. In recent years, the sensing technology has been greatly improved in terms of cost and resolution, due to which the wide deployment of sensors with/without high sampling rate becomes possible. Such improvements lead to the fact that people are able to obtain more and more measurements, i.e., a typical phenomenon in the big data trend. With the availability of more measurements, the performance of the estimators are expected to be improved. They would be able to provide more

precise information about the “states” even in non-ideal environments. However, measurement itself does not unconditionally help. Whether we can make the most of it depends on the ability to effectively process the measurements to extract the information we need.

The way we leverage the measurements must be compatible with the inherent constraints of the system. In some cases, the estimator is unable to directly utilize the massive measurements due to some limitations in the system. For example, in wireless sensor networks, the sensors are usually distantly distributed such that transmitting massive data within a small time interval is expensive, and sometimes even impossible considering the fact that sensors are always powered by battery and subject to stringent power and bandwidth limit [4]–[6]. Another example is the channel state estimation of communication systems. Since the terminal device is generally small and cheap, the memory is often limited and thus impossible to store the massive measurements. In these systems, preprocessing the measurements to make them compatible with the inherent constraint of the system is necessary.

Motivated by these facts, we aim to design a preprocessing method that could make the raw measurements concise enough while at the same time preserve as much information as possible for the subsequent estimation. There has been extensive research in similar topics, especially in the realm of wireless sensor network where stringent power and bandwidth constraint is always imposed. In order to save power and bandwidth, one of the approaches is to use quantized data for the state estimation [7]–[11]. This method is inherent with the discrete nature of the communication between the sensors and the fusion center (FC) in the FC-based network or between sensors in the *ad hoc* network. Instead of limiting the bits representing each measurement, another approach is to preprocess the data transmitted by each sensor and/or cluster to reduce the dimension of data that need to be transmitted [12]–[23]. This method is motivated by the fact that the raw data contains some redundancy, and the preprocessing could help to reduce the redundancy. Moreover, due to the block-wise and multi-rate [24] nature of most wireless sensor networks, reducing the dimension of the data is useful in matching the data from multiple sensors and/or clusters. Besides the wireless sensor network, other researches on compressing the raw data before estimation lie in different fields including image processing [25], power grid [26], [27] and robotic networks [28].

The authors in [20]–[22] provided methods that utilizing a subset of the raw measurements for state estimation in order to reduce the dimension of data. While using a subset of the raw

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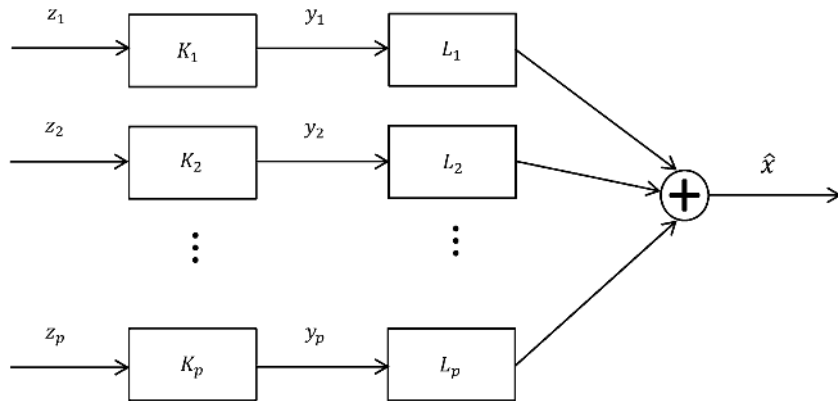


Fig. 1. The hierarchical system structure.

measurements is a good solution in some systems, especially in the wireless sensor networks where each sensor has limited computation power, there are certain cases that the raw measurements can be compressed in some more complicated ways to improve the performance. The authors in [16] presented a minimal bound of the dimension of the compressed sensor data, which could achieve the same estimation performance as using uncompressed sensor data. In other words, this bound is a “lossless” bound. The lossless compression problem was also investigated in [17]. In practice, it is not always possible and/or necessary to ensure the lossless performance. Sometimes, engineers are more concerned about the optimal compression given any system inherent constraint. This problem was studied in [19], however the results were argued to be suboptimal in [18], in which the authors further considered the fading and noisy channels in the sensor network. The algorithm provided in [18] was also suboptimal, which ensured the convergence to a stationary point. Moreover, the algorithm in [18] is computation expensive and sometimes unstable due to the matrix inversion required at each iteration. In [23], the authors addressed the measurement dimension reduction problem with knowledge of the network topology.

In this paper, we consider the block-wise preprocessing and estimation problem. We aim to jointly design the preprocessors and the subsequent estimator where the preprocessors associated with each measurement block is responsible for compressing the raw data within this block to meet the inherent constraints of the system and the estimator relies solely on the compressed data for estimation. We assume that all preprocessors and the estimator are linear. It is found that there is a bound for the dimension reduction. If the permitted dimension is beyond this bound, then it is possible to design the preprocessors and the estimator to make the estimation equivalent to the linear minimum mean square error (LMMSE) estimation, which is the best that can be achieved within the linear space. In other words, there is no performance degradation incurred by the compression. However, if the required dimension reduction is below this bound, then it is impossible to achieve the LMMSE. In this case, we propose an algorithm to design the preprocessor and the subsequent estimator to minimize the mean square error (MSE) between the real “state” and the estimation. Different from most existing algorithms that can only ensure convergence to a stationary point without

performance guarantee, the proposed estimator can bound the MSE by a constant factor times the global minimum value under some additional conditions. The effectiveness of the proposed method is illustrated in different applications. It is shown in some cases that the designed preprocessors are able to compress the raw data into vectors of dimension far below the “LMMSE-achieving” bound without incurring too much performance loss.

The rest of this paper is organized as follows. In Section II, the dimension reduction problem is formulated. The bound for the lossless dimension reduction is proposed in Section III and the algorithm for designing the estimator if the reduction requirement is below the bound is provided in Section IV. The performance bound of the proposed algorithm under some additional conditions is analyzed in Section V. The proposed algorithm is illustrated in three different applications in Section VI. Section VII concludes this paper.

II. PROBLEM FORMULATION

We consider a system where the linear measurements of states are contaminated by noise as follows

$$\mathbf{z} = \mathbf{H}\mathbf{x} + \mathbf{v}, \quad (1)$$

where $\mathbf{x} \in \mathbf{R}^n$ is the vector composed of the n states of the system, $\mathbf{z} \in \mathbf{R}^m$ is the measurement vector, \mathbf{H} is the m by n measurement matrix with $m \geq n$, and $\mathbf{v} \in \mathbf{R}^m$ is the vector composed of the m noise components. Without loss of generality, it is assumed that all the state and noise components are zero mean.

In this paper, we focus on the two-level distributed state estimation problem in this system where the measurement vector \mathbf{z} is partitioned into multiple vectors \mathbf{z}_i 's, and each of them is defined as a block. The measurements in one block would be preprocessed to a shorter vector before subsequent estimation. As shown in Fig. 1, each \mathbf{z}_i is a block and \mathbf{y}_i is the corresponding shorter vector by preprocessing. The subsequent state estimation will be purely based on the \mathbf{y}_i 's.

Assume that there is no shared measurement, i.e., each measurement is involved in only one block. Let \mathbf{z}_i denote the i -th block, and we have $\mathbf{z} = [\mathbf{z}_1^T \cdots \mathbf{z}_p^T]^T$, $\mathbf{z}_i \in \mathbf{R}^{m_i}$, where m_i is the number of measurements corresponding with the i -th block and \mathbf{z}^T denotes the transpose of \mathbf{z} . p is the total number

of blocks. The (1) could be re-written in the block-wise form as follows

$$\begin{pmatrix} \mathbf{z}_1 \\ \vdots \\ \mathbf{z}_p \end{pmatrix} = \begin{pmatrix} \mathbf{H}_1 \\ \vdots \\ \mathbf{H}_p \end{pmatrix} \mathbf{x} + \begin{pmatrix} \mathbf{v}_1 \\ \vdots \\ \mathbf{v}_p \end{pmatrix} \quad (2)$$

where \mathbf{H}_i and \mathbf{v}_i are the measurement matrix and noise vector associated with the block i , respectively.

We are interested in designing a distributed two-level linear estimator $\mathbf{K} = (\mathbf{K}_1 \mathbf{K}_2 \cdots \mathbf{K}_p)$, $\mathbf{L} = (\mathbf{L}_1 \mathbf{L}_2 \cdots \mathbf{L}_p)$, and $\mathbf{G}_i = \mathbf{L}_i \mathbf{K}_i$, $i = 1, 2, \dots, p$, where the local measurements related with block i are first locally processed using \mathbf{K}_i , and then further processed using \mathbf{L}_i as follows

$$\hat{\mathbf{x}} = \sum_{i=1}^p \mathbf{L}_i \mathbf{y}_i = \sum_{i=1}^p \mathbf{L}_i \mathbf{K}_i \mathbf{z}_i = \sum_{i=1}^p \mathbf{G}_i \mathbf{z}_i. \quad (3)$$

where $\mathbf{y}_i = \mathbf{K}_i \mathbf{z}_i$ is the block \mathbf{z}_i preprocessed by \mathbf{K}_i .

Our goal is to properly design \mathbf{K} and \mathbf{L} such that the mean square error (MSE) is minimized. According to the orthogonality principle [29], we can write the MSE as follows

$$E[(\hat{\mathbf{x}} - \mathbf{x})^2] = E[(\hat{\mathbf{x}}_{LMMSE} - \mathbf{x})^2] + E[(\hat{\mathbf{x}} - \hat{\mathbf{x}}_{LMMSE})^2], \quad (4)$$

where \mathbf{x} is the true state vector, $\hat{\mathbf{x}}_{LMMSE}$ is the output of the linear minimum mean square error (LMMSE) estimator, which is in the form of [29]

$$\begin{aligned} \hat{\mathbf{x}}_{LMMSE} &= \mathbf{W} \mathbf{z} \\ &= \left[\boldsymbol{\Sigma}_x - \boldsymbol{\Sigma}_x \mathbf{H}^T \boldsymbol{\Sigma}_e^{-1} \mathbf{H} \left(\boldsymbol{\Sigma}_x^{-1} + \mathbf{H}^T \boldsymbol{\Sigma}_e^{-1} \mathbf{H} \right)^{-1} \right] \\ &\quad \times \mathbf{H}^T \boldsymbol{\Sigma}_e^{-1} \mathbf{z}, \end{aligned} \quad (5)$$

where $\boldsymbol{\Sigma}_x$ and $\boldsymbol{\Sigma}_e$ denote the covariance matrix of \mathbf{x} and \mathbf{v} , respectively. Partitioning the corresponding matrices into sub-matrices for each block, we have $\mathbf{H} = [\mathbf{H}_1^T \cdots \mathbf{H}_p^T]^T$, $\mathbf{H}_i \in \mathbf{R}^{m_i \times n}$, and $\boldsymbol{\Sigma}_e$ is the block diagonal matrix composed of $\boldsymbol{\Sigma}_{e_1}, \dots, \boldsymbol{\Sigma}_{e_p}$, $\boldsymbol{\Sigma}_{e_i} \in \mathbf{R}^{m_i \times m_i}$.

Let us define

$$\mathbf{W}_i \triangleq \left[\boldsymbol{\Sigma}_x - \boldsymbol{\Sigma}_x \mathbf{H}^T \boldsymbol{\Sigma}_e^{-1} \mathbf{H} \left(\boldsymbol{\Sigma}_x^{-1} + \mathbf{H}^T \boldsymbol{\Sigma}_e^{-1} \mathbf{H} \right)^{-1} \right] \mathbf{H}_i^T \boldsymbol{\Sigma}_{e_i}^{-1}, \quad (6)$$

for $i = 1, \dots, p$ and then (5) can be re-written as

$$\hat{\mathbf{x}}_{LMMSE} = \sum_{i=1}^p \mathbf{W}_i \mathbf{z}_i. \quad (7)$$

In (4), $E[(\hat{\mathbf{x}}_{LMMSE} - \mathbf{x})^2]$ is a constant independent of the designed estimator. Thus, minimizing the MSE of the designed estimator is equivalent to minimizing $E[(\hat{\mathbf{x}} - \hat{\mathbf{x}}_{LMMSE})^2]$, which can be further expanded by substituting (3) and (7) as follows

$$E[(\hat{\mathbf{x}} - \hat{\mathbf{x}}_{LMMSE})^2] = Tr((\mathbf{G} - \mathbf{W}) \boldsymbol{\Sigma}_z (\mathbf{G} - \mathbf{W})^T) \quad (8)$$

where $\boldsymbol{\Sigma}_z = \mathbf{H} \boldsymbol{\Sigma}_x \mathbf{H}^T + \boldsymbol{\Sigma}_e$, $\mathbf{G} = (\mathbf{L}_1 \mathbf{K}_1 \mathbf{L}_2 \mathbf{K}_2 \cdots \mathbf{L}_p \mathbf{K}_p)$.

By (3), the rows of \mathbf{K}_i which are all zeros correspond to the zeros in \mathbf{y}_i and are not necessarily kept. Therefore, the constraint on the size of \mathbf{y}_i is equivalent to limiting the number of nonzero rows of \mathbf{K}_i . According to (4) and (8), the problem of minimizing the MSE can be formulated as follows,

$$\begin{aligned} \min_{\mathbf{L}_i, \mathbf{K}_i} & Tr((\mathbf{G} - \mathbf{W}) \boldsymbol{\Sigma}_z (\mathbf{G} - \mathbf{W})^T) \\ \text{s.t.} & \alpha_i \leq c_i, \forall i, \end{aligned} \quad (9)$$

where α_i is the number of nonzero rows of \mathbf{K}_i and c_i is the constraint on the size of \mathbf{y}_i . In other words, the designing purpose is to use \mathbf{K}_i to linearly combine the measurements in \mathbf{z}_i to reduce the size of data while preserving as much as possible useful information for the subsequent state estimation. Due to the linear measurement model $\mathbf{z} = \mathbf{H} \mathbf{x} + \mathbf{v}$, the measurements are not totally independent and thus contains some redundancy. The \mathbf{K}_i would serve to remove the redundancy by linearly combining them. However, the redundancy is limited, due to which there is a minimum size to represent the measurements in \mathbf{z}_i . It will be shown in the next section that if c_i is beyond this size, \mathbf{K}_i can remove the redundancy such that the constraints are satisfied and all the useful information is preserved. If c_i is below the minimum size, then not only the redundancy but also some useful information would be removed, incurring performance loss in the subsequent estimation.

III. MINIMUM DIMENSION REQUIREMENT FOR LOSSLESS COMPRESSION

In this section, a sufficient and necessary condition on the c_i 's for the system to achieve the LMMSE estimation will be shown. Before we show the condition, let us first establish an equivalent form of the problem in (9).

For any given \mathbf{G}_i , it is always possible to use the singular value decomposition (SVD) to re-write it as [30]

$$\mathbf{G}_i = \mathbf{U}_i \boldsymbol{\Sigma}_i \mathbf{V}_i' \quad (10)$$

where \mathbf{V}_i' denote the Hermitian conjugate of \mathbf{V}_i .

If let $\mathbf{L}_i = \mathbf{U}_i$ and $\mathbf{K}_i = \boldsymbol{\Sigma}_i \mathbf{V}_i'$, then $\alpha_i = \text{rank}(\mathbf{G}_i)$, which means that we can always reduce α_i to $\text{rank}(\mathbf{G}_i)$. Moreover, it is impossible to reduce α_i below $\text{rank}(\mathbf{G}_i)$ because if $\alpha_i < \text{rank}(\mathbf{G}_i)$, then $\text{rank}(\mathbf{K}_i) \leq \alpha_i < \text{rank}(\mathbf{G}_i)$. In this case, the controversy comes that $\text{rank}(\mathbf{G}_i) = \text{rank}(\mathbf{L}_i \mathbf{K}_i) \leq \text{rank}(\mathbf{K}_i) < \text{rank}(\mathbf{G}_i)$. In other words, for any given \mathbf{G}_i , we can always and at most reduce α_i to $\text{rank}(\mathbf{G}_i)$, i.e., limiting α_i is equivalent to limiting $\text{rank}(\mathbf{G}_i)$. The problem in (9) could be reformulated as

$$\begin{aligned} \min_{\mathbf{G}_i} & Tr((\mathbf{G} - \mathbf{W}) \boldsymbol{\Sigma}_z (\mathbf{G} - \mathbf{W})^T) \\ \text{s.t.} & \text{rank}(\mathbf{G}_i) \leq c_i, \forall i. \end{aligned} \quad (11)$$

Therefore, the problem of jointly designing \mathbf{L} and \mathbf{K} is equivalent to finding an optimal \mathbf{G} in the low-rank space constrained by c_i 's to minimize the MSE. From (11), we develop the sufficient and necessary condition on c_i 's for the system to achieve the LMMSE estimation. Please note that similar result with different assumptions had also been derived in Theorem 2.1 of [16].

Theorem 1: A sufficient and necessary condition on c_i 's for the estimation defined in (3) constrained by (11) to achieve the global LMMSE state estimation is $\forall i, c_i \geq \text{rank}(\mathbf{H}_i)$.

Proof: First notice from (6) that $\text{rank}(\mathbf{W}_i) = \text{rank}(\mathbf{H}_i)$. In the following, we would prove $c_i \geq \text{rank}(\mathbf{W}_i)$ as the sufficient and necessary conditions instead.

(proof of sufficiency) If $c_i \geq \text{rank}(\mathbf{W}_i)$, then \mathbf{W} is in the low-rank space constrained by c_i 's. Choose $\mathbf{G}_i = \mathbf{W}_i$ for all the blocks thus $\mathbf{G} = \mathbf{W}$, achieving the LMMSE.

(Proof of Necessity) The state estimation $\hat{\mathbf{x}}(\mathbf{z})$ in (3) as a function of \mathbf{z} achieves the LMMSE if $\hat{\mathbf{x}}(\mathbf{z}) = \hat{\mathbf{x}}_{\text{LMMSE}}(\mathbf{z})$ for any $\mathbf{z} \in \mathbf{R}^m$ where $\hat{\mathbf{x}}_{\text{LMMSE}}(\mathbf{z})$ is defined in (7). We will show the necessity by contradiction.

Suppose there exists a block indexed s such that $c_s < \text{rank}(\mathbf{W}_s)$. Let $\dim(\Omega)$ denote the dimension of the linear space Ω , since $\text{rank}(\mathbf{G}_s) \leq c_s < \text{rank}(\mathbf{W}_s)$, $\dim(\Theta_{\mathbf{G}_s}) > \dim(\Theta_{\mathbf{W}_s})$ where $\Theta_{\mathbf{G}_s}$ and $\Theta_{\mathbf{W}_s}$ are the null spaces of \mathbf{G}_s and \mathbf{W}_s , respectively. In other words, \exists a vector $\tilde{\mathbf{z}}_s$ such that $\mathbf{G}_s \tilde{\mathbf{z}}_s = 0$ while $\mathbf{W}_s \tilde{\mathbf{z}}_s \neq 0$. Therefore, there exists a vector $\tilde{\mathbf{z}} = (\mathbf{0} \ \tilde{\mathbf{z}}_s \ \mathbf{0})^T$ such that $\hat{\mathbf{x}}(\mathbf{z}) = \sum_{i=1}^p \mathbf{G}_i \mathbf{z}_i \neq \sum_{i=1}^p \mathbf{W}_i \mathbf{z}_i = \hat{\mathbf{x}}_{\text{LMMSE}}(\mathbf{z})$, where $\mathbf{0}$ is the vector with appropriate size composed of all zeros. By definition, the estimation $\hat{\mathbf{x}}(\mathbf{z})$ does not achieve the LMMSE. Therefore, $r_i \geq \text{rank}(\mathbf{H}_i)$ is a necessary condition for the estimator \mathbf{G} to achieve LMMSE. ■

From Theorem 1, we can see that the minimum c_i for the system to achieve LMMSE is $\text{rank}(\mathbf{H}_i)$. Taking expectation over \mathbf{v} in (2), we have $E[\mathbf{z}_i] = \mathbf{H}_i \mathbf{x}$, where we can see that the number of independent values in the mean of \mathbf{z}_i is determined by $\text{rank}(\mathbf{H}_i)$. Therefore, Theorem 1 could be interpreted that no information of the measurement \mathbf{z}_i is lost if c_i is at least the number of independent components in \mathbf{z}_i . If we view the linear combination as projection, in case that $c_i < \text{rank}(\mathbf{H}_i)$, one tries to project the vector \mathbf{z}_i into a space with insufficient bases, due to which some information is lost and the LMMSE estimation cannot be achieved.

IV. ESTIMATOR DESIGN FOR LOSSY COMPRESSION

In the previous section, we derive a lower bound for c_i 's to enable the LMMSE estimator, i.e., for any dimension constraint beyond the lower bound, we can achieve the same MSE performance as the LMMSE estimator. However, in some situations where high compression ratio is needed due to, for example, high communication cost, the required dimension may be even smaller than the lower bound. In such a case, the design of the estimators will be constrained in the low-rank space, as given in (11). Obviously, the objective function $\text{Tr}((\mathbf{G} - \mathbf{W})\boldsymbol{\Sigma}_z(\mathbf{G} - \mathbf{W})^T)$ is quadratic thus a convex function of \mathbf{G} , but the set $\{\mathbf{G} | \text{rank}(\mathbf{G}_i) \leq c_i\}$ is not a convex set. In this section, we proposed an algorithm to tackle this non-convex optimization problem that is guaranteed to converge.

To make (11) more tractable, we first decompose the covariance matrix as follows

$$\boldsymbol{\Sigma}_z = \mathbf{Q}\boldsymbol{\Lambda}\mathbf{Q}^T = \boldsymbol{\Sigma}_z^{\frac{1}{2}}\boldsymbol{\Lambda}\boldsymbol{\Sigma}_z^{\frac{1}{2}T} \quad (12)$$

where $\boldsymbol{\Sigma}_z^{\frac{1}{2}} = \mathbf{Q}\boldsymbol{\Lambda}^{\frac{1}{2}}$.

We then further re-write $\boldsymbol{\Sigma}_z^{\frac{1}{2}}$ in the block-wise form

$$\boldsymbol{\Sigma}_z^{\frac{1}{2}} = \begin{bmatrix} \mathbf{A}_1 \\ \mathbf{A}_2 \\ \vdots \\ \mathbf{A}_p \end{bmatrix} \triangleq \mathbf{A} \quad (13)$$

where $\mathbf{A}_i \in \mathbf{R}^{m_i \times n}$. Now we can re-write the objective function in (11) as follows

$$\begin{aligned} & \text{Tr}[(\mathbf{G} - \mathbf{W})\boldsymbol{\Sigma}_z(\mathbf{G} - \mathbf{W})^T] \\ &= \text{Tr}\left[(\mathbf{G} - \mathbf{W})\boldsymbol{\Sigma}_z^{\frac{1}{2}}\boldsymbol{\Sigma}_z^{\frac{1}{2}T}(\mathbf{G} - \mathbf{W})^T\right] \\ &= \left\| \mathbf{W}\boldsymbol{\Sigma}_z^{\frac{1}{2}} - \sum_{i=1}^p \mathbf{G}_i \mathbf{A}_i \right\|_F^2 \end{aligned} \quad (14)$$

where $\|\mathbf{X}\|_F$ denotes the Frobenius norm [30] of matrix \mathbf{X} .

Since the matrices \mathbf{W} and $\boldsymbol{\Sigma}_z$ are determined by the system parameter, they are independent of the designed estimator. Let us denote the matrix $\mathbf{W}\boldsymbol{\Sigma}_z^{\frac{1}{2}}$ by \mathbf{Y} . With $\mathbf{G} = [\mathbf{G}_1 \ \mathbf{G}_2 \ \cdots \ \mathbf{G}_p]$, the problem in (11) can be re-written as

$$\begin{aligned} & \min_{\mathbf{G}_i} \quad \|\mathbf{Y} - \mathbf{G}\mathbf{A}\|_F^2 \\ & \text{s.t.} \quad \text{rank}(\mathbf{G}_i) \leq c_i, \forall i. \end{aligned} \quad (15)$$

In (15), we can see that the problem boils down to finding estimator \mathbf{G} in the low rank space to minimize the distance between $\mathbf{G}\mathbf{A}$ and a point \mathbf{Y} , where \mathbf{A} and \mathbf{Y} are independent of \mathbf{G} . If \mathbf{A} is an identity matrix, the solution is trivial by discarding the most insignificant singular values of \mathbf{W}_i until $\text{rank}(\mathbf{G}_i) \leq c_i, \forall i$ [31]. However, for a general \mathbf{A} , there is no explicit solution.

Since the problem in (15) is hard to directly solve, we seek to tackle it by formulating another optimization problem which is an approximation to the original problem (15). By introducing a new matrix $\mathbf{D} \in \mathbf{R}^{n \times m}$, the objective function in (15) can be written as

$$\begin{aligned} \|\mathbf{Y} - \mathbf{G}\mathbf{A}\|_F^2 &= \|\mathbf{Y} - \mathbf{D}\mathbf{A} + \mathbf{D}\mathbf{A} - \mathbf{G}\mathbf{A}\|_F^2 \\ &\leq \|\mathbf{Y} - \mathbf{D}\mathbf{A}\|_F^2 + \|\mathbf{D}\mathbf{A} - \mathbf{G}\mathbf{A}\|_F^2 \\ &\leq \|\mathbf{Y} - \mathbf{D}\mathbf{A}\|_F^2 + \lambda_{(1)}^2(\mathbf{A})\|(\mathbf{D} - \mathbf{G})\|_F^2 \end{aligned} \quad (16)$$

where $\lambda_{(1)}(\mathbf{X})$ is the maximum absolute value of the eigenvalues of \mathbf{X} . Therefore, we try to solve the problem

$$\begin{aligned} & \min_{\mathbf{G}_i, \mathbf{D}_i} \quad \left\| \mathbf{Y} - \sum_{i=1}^p \mathbf{D}_i \mathbf{A}_i \right\|_F^2 + \lambda_{(1)}^2(\mathbf{A}) \left(\sum_{i=1}^p \|\mathbf{G}_i - \mathbf{D}_i\|_F^2 \right) \\ & \text{s.t.} \quad \text{rank}(\mathbf{G}_i) \leq c_i, \forall i \end{aligned} \quad (17)$$

which is to minimize an upper bound of the objective function in (15).

We further transform the problem in (17) using a similar idea in [32]. We replace $\lambda_{(1)}^2(\mathbf{A})$ by another parameter $\gamma > 0$. This

transformation gives the algorithm one more degree of flexibility in choosing γ to further improve the performance than fixing it to be $\lambda_{(1)}^2(\mathbf{A})$. The problem becomes

$$\begin{aligned} \min_{\mathbf{D}_i, \mathbf{G}_i} \quad & \left\| \mathbf{Y} - \sum_{i=1}^p \mathbf{D}_i \mathbf{A}_i \right\|_F^2 + \gamma \left(\sum_{i=1}^p \|\mathbf{G}_i - \mathbf{D}_i\|_F^2 \right) \\ \text{s.t.} \quad & \text{rank}(\mathbf{G}_i) \leq c_i, \forall i. \end{aligned} \quad (18)$$

where \mathbf{G}_i 's are the estimators in the low-rank constrained space and \mathbf{D}_i 's are the estimators in the unconstrained space. The first term in (18) is the distance between the appropriate estimation chosen in the non-constrained space and the point that we aim to approximate in the non-constrained space. The second term is the distance between the point chosen in the non-constrained space and its projection in the low rank space. From (18), we can see that the optimal estimator is found by simultaneously minimizing two different distances through a balance factor γ . In such a case, we will obtain an estimator \mathbf{D} in the unconstrained space that make $\|\mathbf{Y} - \sum_{i=1}^p \mathbf{D}_i \mathbf{A}_i\|_F^2$ small, while at the same time the corresponding projection on the low rank space will not introduce too much performance degradation. In the sequel, it will be discussed in details how to solve the optimization problem and derive the performance bound.

Since directly solving the optimization problem in (18) is difficult, we will solve it iteratively with two steps in each iteration: in the first step we fix \mathbf{G}_i and optimize \mathbf{D}_i , and then in the second step we optimize \mathbf{G}_i by fixing \mathbf{D}_i . For the first step, when \mathbf{G}_i is fixed, since the feasible set for \mathbf{D}_i is $\mathbf{R}^{m_i \times n}$, the problem is a convex optimization problem, and we can derive the solution using the first order condition as follows

$$(\mathbf{D}\mathbf{A} - \mathbf{Y})\mathbf{A}_i^T + \gamma(\mathbf{D}_i - \mathbf{G}_i) = \mathbf{0}, \forall i. \quad (19)$$

At the second step we fix \mathbf{D} in (18) and optimize it with respect to \mathbf{G}_i , which is equivalent to optimizing $\|\mathbf{G}_i - \mathbf{D}_i\|_F^2$ since $\|\mathbf{Y} - \sum_{i=1}^p \mathbf{D}_i \mathbf{A}_i\|_F^2$ is independent of \mathbf{G}_i . By Eckart-Young Theorem [31], the optimal \mathbf{G}_i for the problem

$$\begin{aligned} \min_{\mathbf{G}_i} \quad & \|\mathbf{G}_i - \mathbf{D}_i\|_F^2 \\ \text{s.t.} \quad & \text{rank}(\mathbf{G}_i) \leq c_i, \forall i \end{aligned} \quad (20)$$

is derived by applying low rank projection for each \mathbf{D}_i as follows

$$\mathbf{G}_i = \mathbf{T}_i \overline{\mathbf{\Delta}}_i \mathbf{S}_i' \quad (21)$$

where $\mathbf{D}_i = \mathbf{T}_i \mathbf{\Delta}_i \mathbf{S}_i'$ is the singular value decomposition of \mathbf{D}_i and $\overline{\mathbf{\Delta}}_i$ is the truncated singular value matrix where only the most significant c_i singular values are kept while all other singular values are assigned 0.

The (19) can be re-written as

$$\mathbf{D}\mathbf{A}\mathbf{A}_i^T + \gamma\mathbf{D}_i = \mathbf{Y}\mathbf{A}_i + \gamma\mathbf{G}_i, \forall i. \quad (22)$$

Combining (22) for all the blocks, we have

$$\mathbf{D} = (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{-1}(\mathbf{Y}\mathbf{A} + \gamma\mathbf{G}). \quad (23)$$

Combining (21) and (23), the iterative steps for solving (18) can be written as follows

$$\begin{aligned} \mathbf{G}_i^{(k)} &= \mathbf{T}_i^{(k)} \overline{\mathbf{\Delta}}_i^{(k)} \mathbf{S}_i^{(k)'} , \forall i \\ \mathbf{D}^{(k+1)} &= (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{-1} (\mathbf{Y}\mathbf{A} + \gamma\mathbf{G}^{(k)}) \end{aligned} \quad (24)$$

where

$$\begin{aligned} \mathbf{G}^{(k)} &= [\mathbf{G}_1^{(k)}, \mathbf{G}_2^{(k)}, \dots, \mathbf{G}_p^{(k)}] \\ \mathbf{D}^{(k)} &= [\mathbf{D}_1^{(k)}, \mathbf{D}_2^{(k)}, \dots, \mathbf{D}_p^{(k)}]. \end{aligned} \quad (25)$$

The initial point of the iteration could be easily set as $\mathbf{D}^{(0)} = \mathbf{W}$. The algorithm is terminated if $\text{MSE}^{(k+1)} - \text{MSE}^{(k)} < \epsilon$ where $\text{MSE}^{(k)}$ is the MSE achieved by the estimator at k iterations and ϵ is the pre-defined threshold. There is a tradeoff between runtime and performance in selecting ϵ .

It can be seen from (24) that the proposed algorithm iteratively fixes one of \mathbf{G} and \mathbf{D} and updates the other one to minimize the cost in (18) until the termination condition is reached. Thus, it can be easily seen that the proposed algorithm always converges since the cost is always nonincreasing during the iterations and the cost is lower bounded by the LMMSE. Note that the converged solution is a stationary point and it may not be a global optimum. However, one unique characteristic of the proposed algorithm is that under some additional conditions, the MSE of the local optimal \mathbf{G} designed by this algorithm is bounded by a constant C times the minimum MSE achievable by any low rank estimator \mathbf{G} satisfying the constraints in (11), which will be shown in next section.

V. PERFORMANCE ANALYSIS UNDER ADDITIONAL CONDITIONS

In this section, it will be shown that the convergence value of $\tilde{\mathbf{G}}$ would bound the MSE by a constant C times the MSE achieved by \mathbf{G}^* where \mathbf{G}^* is the estimator minimizing $\|\mathbf{Y} - \mathbf{G}\mathbf{A}\|_F^2$ over the whole low-rank space $\{\mathbf{G} | \text{rank}(\mathbf{G}_i) \leq c_i, \forall i\}$. We will also show that C is related with the restricted isometry constant [33] of the matrix \mathbf{A} . To show the condition, we first introduce the definition of restricted isometry constant.

Definition 1: The d -restricted isometry constant [33] of \mathbf{A} is defined as δ_d such that $\forall \mathbf{X} \in \mathbf{R}^{n \times m}$ satisfying $\text{rank}(\mathbf{X}) \leq d$,

$$(1 - \delta_d)\|\mathbf{X}\|_F^2 \leq \|\mathbf{X}\mathbf{A}\|_F^2 \leq (1 + \delta_d)\|\mathbf{X}\|_F^2. \quad (26)$$

With this definition, we are able establish a lemma that would facilitate the performance analysis of the proposed iterative algorithm.

Lemma 1: Suppose the eigenvalue decomposition of $\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I}$ is $\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I} = \mathbf{J}\mathbf{\Gamma}\mathbf{J}^T$, define $(\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{1/2} = \mathbf{J}\mathbf{\Gamma}^{1/2}$, then if \mathbf{A} has the d -restricted isometry constant δ_d where $d \leq n$, for any $\mathbf{X} \in \mathbf{R}^{n \times m}$ satisfying $\text{rank}(\mathbf{X}) \leq d$, we have

$$(1 - \delta_d + \gamma)\|\mathbf{X}\|_F^2 \leq \left\| \mathbf{X}(\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{\frac{1}{2}} \right\|_F^2 \leq (1 + \delta_d + \gamma)\|\mathbf{X}\|_F^2 \quad (27)$$

Proof: For any \mathbf{X} satisfying the conditions in Definition 1, we have

$$\begin{aligned} \left\| \mathbf{X}(\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{\frac{1}{2}} \right\|_F^2 &= \text{Tr} \left(\mathbf{X}(\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})\mathbf{X}^T \right) \\ &= \text{Tr}(\mathbf{X}\mathbf{A}\mathbf{A}^T\mathbf{X}^T) + \gamma\text{Tr}(\mathbf{X}\mathbf{X}^T) \\ &= \|\mathbf{X}\mathbf{A}\|_F^2 + \gamma\|\mathbf{X}\|_F^2 \end{aligned}$$

By applying the restricted isometry property (RIP), it can be further re-written as

$$(1 - \delta_d + \gamma)\|\mathbf{X}\|_F^2 \leq \|\mathbf{X}\mathbf{A}\|_F^2 + \gamma\|\mathbf{X}\|_F^2 \leq (1 + \delta_d + \gamma)\|\mathbf{X}\|_F^2 \quad (28)$$

To evaluate the estimator calculated by the proposed algorithm, we first define the functions

$$\phi(\mathbf{X}) = \frac{1}{2}\|\mathbf{Y} - \mathbf{X}\mathbf{A}\|_F^2 \quad (29)$$

and

$$\Psi(\mathbf{X}) = \|\mathbf{Y} - \mathbf{X}\mathbf{A}\|_F^2 + \mu \quad (30)$$

where $\mathbf{X} \in \mathbf{R}^{n \times m}$, and $\mu = E[(\hat{\mathbf{x}}_{LMMSE} - \mathbf{x})^2]$. According to (4), $\Psi(\mathbf{G})$ is the MSE achieved by the estimator \mathbf{G} . Define \mathbf{G}^* as

$$\begin{aligned} \mathbf{G}^* &= \arg \min_{\mathbf{G}} \phi(\mathbf{G}) \\ \text{s.t. } \text{rank}(\mathbf{G}_i) &\leq c_i. \end{aligned} \quad (31)$$

In other words, \mathbf{G}^* is the optimal solution for the minimization problem in (15). Then we have the following theorem:

Theorem 2: If \mathbf{A} has the n -restricted isometry constant δ_n satisfying $\delta_n < \sqrt{5} - 2$, then by the iterative steps in (24),

$$\lim_{k \rightarrow \infty} \frac{\Psi(\mathbf{G}^{(k)})}{\Psi(\mathbf{G}^*)} < C \quad (32)$$

where

$$\begin{aligned} C &= \frac{1 + l + \frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2}}{1 - l} \\ l &= \left(\frac{2\delta_n\gamma^2}{(1-\delta_n+\gamma)^2} + \frac{1}{2}\gamma \right) \cdot \frac{4}{1-\delta_n} + \frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2}. \end{aligned} \quad (33)$$

Proof: For any \mathbf{G} satisfying the constraints in (15), we have $\text{rank}(\mathbf{G}) \leq n$.

By Lemma (1), the following condition holds for any \mathbf{G} satisfying the constraints in (15)

$$(1 - \delta_n + \gamma)\|\mathbf{G}\|_F^2 \leq \left\| \mathbf{G}(\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{\frac{1}{2}} \right\|_F^2 \leq (1 + \delta_n + \gamma)\|\mathbf{G}\|_F^2. \quad (34)$$

Since the $\phi(\mathbf{G})$ is quadratic in \mathbf{G} , we could represent it by the Taylor series up to the second order as follows

$$\begin{aligned} \phi(\mathbf{G}^{(k+1)}) - \phi(\mathbf{G}^{(k)}) &= \left\langle (\mathbf{G}^{(k)}\mathbf{A} - \mathbf{Y})\mathbf{A}^T, \mathbf{G}^{(k+1)} - \mathbf{G}^{(k)} \right\rangle \\ &\quad + \frac{1}{2} \left\| (\mathbf{G}^{(k+1)} - \mathbf{G}^{(k)})\mathbf{A} \right\|_F^2 \end{aligned} \quad (35)$$

where $\langle \mathbf{M}, \mathbf{N} \rangle \triangleq \text{Trace}(\mathbf{M}\mathbf{N}^T)$. Recall that in (24)

$$\mathbf{D}^{(k+1)}(\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I}) = \mathbf{Y}\mathbf{A}^T + \gamma\mathbf{G}^{(k)}. \quad (36)$$

Substituting $\mathbf{Y}\mathbf{A}^T$ in the above equation into (35), we get

$$\begin{aligned} &\phi(\mathbf{G}^{(k+1)}) - \phi(\mathbf{G}^{(k)}) \\ &= \frac{1}{2} \left\| (\mathbf{G}^{(k+1)} - \mathbf{D}^{(k+1)}) (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{\frac{1}{2}} \right\|_F^2 \\ &\quad - \frac{1}{2} \left\| (\mathbf{G}^{(k)} - \mathbf{D}^{(k+1)}) (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{\frac{1}{2}} \right\|_F^2 \\ &\quad - \frac{1}{2}\gamma \left\| \mathbf{G}^{(k+1)} - \mathbf{G}^{(k)} \right\|_F^2 \\ &\leq \frac{1}{2}(1 + \delta_n + \gamma) \left\| \mathbf{G}^{(k+1)} - \mathbf{D}^{(k+1)} \right\|_F^2 \\ &\quad - \frac{1}{2}(1 - \delta_n + \gamma) \left\| \mathbf{G}^{(k)} - \mathbf{D}^{(k+1)} \right\|_F^2 \\ &\leq \frac{1}{2}(1 + \delta_n + \gamma) \left\| \mathbf{G}^* - \mathbf{D}^{(k+1)} \right\|_F^2 \\ &\quad - \frac{1}{2}(1 - \delta_n + \gamma) \left\| \mathbf{G}^{(k)} - \mathbf{D}^{(k+1)} \right\|_F^2 \end{aligned} \quad (37)$$

where we have applied Lemma (1) in the first inequality since $\text{rank}(\mathbf{G}^{(k+1)} - \mathbf{D}^{(k+1)}) \leq n$ and $\text{rank}(\mathbf{G}^{(k)} - \mathbf{D}^{(k+1)}) \leq n$. The last inequality is due to the fact that $\mathbf{G}^{(k+1)}$ is the global minimizer of $\|\mathbf{G} - \mathbf{D}^{(k+1)}\|_F$ in the block-wise low-rank subspace thus replacing it with \mathbf{G}^* would increase the value. On the other hand,

$$\begin{aligned} &\phi(\mathbf{G}^*) - \phi(\mathbf{G}^{(k)}) \\ &= \frac{1}{2} \left\| (\mathbf{G}^* - \mathbf{D}^{(k+1)}) (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{\frac{1}{2}} \right\|_F^2 \\ &\quad - \frac{1}{2} \left\| (\mathbf{G}^{(k)} - \mathbf{D}^{(k+1)}) (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{\frac{1}{2}} \right\|_F^2 \\ &\quad - \frac{1}{2}\gamma \left\| \mathbf{G}^* - \mathbf{G}^{(k)} \right\|_F^2 \\ &\geq \frac{1}{2}(1 - \delta_n + \gamma) \left\| \mathbf{G}^* - \mathbf{D}^{(k+1)} \right\|_F^2 \\ &\quad - \frac{1}{2}(1 + \delta_n + \gamma) \left\| \mathbf{G}^{(k)} - \mathbf{D}^{(k+1)} \right\|_F^2 \\ &\quad - \frac{1}{2}\gamma \left\| \mathbf{G}^* - \mathbf{G}^{(k)} \right\|_F^2 \end{aligned} \quad (38)$$

where Lemma (1) has been used.

Now subtracting (38) from (37), we get

$$\begin{aligned} &\phi(\mathbf{G}^{(k+1)}) - \phi(\mathbf{G}^*) \\ &\leq \delta_n \left\| \mathbf{G}^* - \mathbf{D}^{(k+1)} \right\|_F^2 + \delta_n \left\| \mathbf{G}^{(k)} - \mathbf{D}^{(k+1)} \right\|_F^2 \\ &\quad + \frac{1}{2}\gamma \left\| \mathbf{G}^* - \mathbf{G}^{(k)} \right\|_F^2 \\ &= \delta_n \left\| [(\mathbf{G}^*\mathbf{A} - \mathbf{Y})\mathbf{A}^T \right. \\ &\quad \left. + \gamma(\mathbf{G}^* - \mathbf{G}^{(k)})] (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{-1} \right\|_F^2 \\ &\quad + \delta_n \left\| (\mathbf{G}^{(k)}\mathbf{A}\mathbf{A}^T - \mathbf{Y}\mathbf{A}^T) (\mathbf{A}\mathbf{A}^T + \gamma\mathbf{I})^{-1} \right\|_F^2 \\ &\quad + \frac{1}{2}\gamma \left\| \mathbf{G}^* - \mathbf{G}^{(k)} \right\|_F^2 \end{aligned}$$

$$\begin{aligned}
&\leq \frac{2\delta_n}{(1-\delta_n+\gamma)^2} \left(\|(\mathbf{G}^* \mathbf{A} - \mathbf{Y}) \mathbf{A}^T\|_F^2 \right. \\
&\quad \left. + \gamma^2 \|\mathbf{G}^* - \mathbf{G}^{(k)}\|_F^2 \right) \\
&\quad + \frac{\delta_n}{(1-\delta_n+\gamma)^2} \left\| \left(\mathbf{G}^{(k)} \mathbf{A} - \mathbf{Y} \right) \mathbf{A}^T \right\|_F^2 \\
&\quad + \frac{1}{2} \gamma \|\mathbf{G}^* - \mathbf{G}^{(k)}\|_F^2 \\
&\leq \frac{4\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2} \phi(\mathbf{G}^*) + \frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2} \phi(\mathbf{G}^{(k)}) \\
&\quad + \left(\frac{2\delta_n\gamma^2}{(1-\delta_n+\gamma)^2} + \frac{1}{2}\gamma \right) \|\mathbf{G}^* - \mathbf{G}^{(k)}\|_F^2 \quad (39)
\end{aligned}$$

where the Lemma (1) and the Cauchy-Schwarz inequality $\|(\mathbf{G}^* \mathbf{A} - \mathbf{Y}) \mathbf{A}^T + \gamma(\mathbf{G}^* - \mathbf{G}^{(k)})\|_F^2 \leq 2(\|(\mathbf{G}^* \mathbf{A} - \mathbf{Y}) \mathbf{A}^T\|_F^2 + \gamma^2 \|\mathbf{G}^* - \mathbf{G}^{(k)}\|_F^2)$ have been used in the first inequality. The second inequality is due to the RIP in (26). By the same way, the last term is bounded by

$$\begin{aligned}
&\|\mathbf{G}^* - \mathbf{G}^{(k)}\|_F^2 \\
&\leq \frac{1}{1-\delta_n} \|\mathbf{G}^* \mathbf{A} - \mathbf{G}^{(k)} \mathbf{A}\|_F^2 \\
&\leq \frac{1}{1-\delta_n} \cdot 2 \left[\|\mathbf{G}^* \mathbf{A} - \mathbf{Y}\|_F^2 + \|\mathbf{G}^{(k)} \mathbf{A} - \mathbf{Y}\|_F^2 \right] \\
&= \frac{4}{1-\delta_n} \left(\phi(\mathbf{G}^*) + \phi(\mathbf{G}^{(k)}) \right). \quad (40)
\end{aligned}$$

Substituting (40) into (39) we can get

$$\begin{aligned}
&\phi(\mathbf{G}^{(k+1)}) - \phi(\mathbf{G}^*) \\
&\leq \frac{4\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2} \phi(\mathbf{G}^*) + \frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2} \phi(\mathbf{G}^{(k)}) \\
&\quad + \left(\frac{2\delta_n\gamma^2}{(1-\delta_n+\gamma)^2} + \frac{1}{2}\gamma \right) \cdot \frac{4}{1-\delta_n} \left(\phi(\mathbf{G}^*) + \phi(\mathbf{G}^{(k)}) \right) \quad (41)
\end{aligned}$$

By simplifying (41), we can get

$$\phi(\mathbf{G}^{(k+1)}) - C\phi(\mathbf{G}^*) \leq l \left(\phi(\mathbf{G}^{(k)}) - C\phi(\mathbf{G}^*) \right) \quad (42)$$

where

$$\begin{aligned}
l &= \left(\frac{2\delta_n\gamma^2}{(1-\delta_n+\gamma)^2} + \frac{1}{2}\gamma \right) \cdot \frac{4}{1-\delta_n} + \frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2} \\
C &= \frac{1+l+\frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2}}{1-l} \quad (43)
\end{aligned}$$

Since $\lim_{\gamma \rightarrow 0} l = \frac{2\delta_n(1+\delta_n)}{(1-\delta_n)^2} < 1$ if $\delta_n < \sqrt{5} - 2$, it is possible to find small enough γ such that $0 < l < 1$ making the iteration converge, and $\lim_{k \rightarrow \infty} \phi(\mathbf{G}^{(k)}) = C\phi(\mathbf{G}^*)$.

According to (29) and (30), the MSE achieved by the estimator $\mathbf{G}^{(k)}$ is

$$\Psi(\mathbf{G}^{(k)}) = 2\phi(\mathbf{G}^{(k)}) + \mu. \quad (44)$$

TABLE I
MINIMUM ACHIEVABLE VALUES OF C FOR SOME δ_n

δ_n	C	δ_n	C
0.02	1.1331	0.04	1.2977
0.06	1.5045	0.08	1.7696
0.10	2.1186	0.12	2.5949
0.14	3.2778	0.16	4.3301
0.18	6.1470	0.20	10.0000

Comparing it with $\Psi(\mathbf{G}^*)$

$$\lim_{k \rightarrow \infty} \frac{\Psi(\mathbf{G}^{(k)})}{\Psi(\mathbf{G}^*)} = \frac{2\phi(\mathbf{G}^{(k)}) + \mu}{2\phi(\mathbf{G}^*) + \mu} < \frac{2\phi(\mathbf{G}^{(k)})}{2\phi(\mathbf{G}^*)} = C \quad (45)$$

where the inequality is due to that fact that $\phi(\mathbf{G}^{(k)})$ is always greater than $\phi(\mathbf{G}^*)$ and $\mu > 0$. ■

It is shown in the above discussions that besides an estimator design, the proposed algorithm also provides some measure based on the system parameter δ_n to evaluate the performance of the designed estimator, which is different from the existing method that can only guaranteed convergence to a stationary point [18]. Please note that the condition $\delta_n < \sqrt{5} - 2$ is only a sufficient condition for the performance analysis and it should not limit the applicability of the proposed algorithm. The convergence of the algorithm is always guaranteed independent of this condition. Moreover, sometimes it is possible to find some γ to make $l < 1$ even if $\delta_n \geq \sqrt{5} - 2$.

From (33), since $l > 0$, C is always greater than 1, which means that the MSE of the designed estimator \mathbf{G} is always greater than that of the optimal low-rank estimator and C could provide a bound of the difference between them. This bound is related with γ and the n -restricted isometry constant δ_n . The minimum feasible values of C for some cases when $\delta_n < \sqrt{5} - 2$ are shown in Table I. For some cases, if δ_n is small, we can choose appropriate γ which would ensure both l and $\frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2}$ to be close to 0 thus C is close to 1. The MSE of the designed estimator will be close to that of the optimal estimator \mathbf{G}^* . However, if δ_n is relatively large, we need to choose small enough γ to ensure $l < 1$, which would make $\frac{2\delta_n(1+\delta_n)}{(1-\delta_n+\gamma)^2}$ relatively large and therefore C would be much greater than 1. In this case, the upper bound of performance loss provided by C is loose. Nevertheless, it does not necessarily indicate the performance of the designed estimator is poor, which will be shown by numerical examples in next section.

Before comparing the performance with that of the method in [18], let us first compare the computation complexity of these two methods. In Algorithm 1 of [18], it requires multiple matrix inversion at each iteration. There are two main disadvantages: first, sometimes the matrix that needs to be inverted is so close to singular that inverting it would incur a lot of inaccuracy due to which the convergence becomes an issue. Another disadvantage is that since the matrix inversion is of cubic complexity with respect to the number of measurements, this method might not be scalable. However, in the method proposed in this paper, the matrix inversion $(\mathbf{A}\mathbf{A}^T + \gamma I)^{-1}$ is independent of the iteration and thus does not need to be calculated each time. It makes the algorithm more robust without the risk of inverting

a matrix close to singular. Also, it reduces the complexity significantly and therefore makes it more suitable for the condition with massive measurements, especially in the case that estimators are re-designed periodically to meet the time-varying conditions.

VI. APPLICATIONS

There are a variety of problems in different fields using the linear measurement model in (1) where the proposed method can be applied such that the raw measurements are compressed to meet the system's inherent constraints. If the conditions in Theorem 1 are satisfied, then the distributed LMMSE estimator can be implemented. Moreover, we are more interested in the performance of the designed low-rank estimator if those conditions can not be satisfied. In this section, three state estimation applications with inherent constraints are discussed to illustrate the effectiveness of the proposed algorithm.

A. MIMO Channel Estimation

To fully achieve the spatial and time diversity that MIMO systems can offer, the accurate channel state information (CSI) is required at the transmitter and/or receiver [34]. For example, the beamforming requires the CSI at the transmitter [35] while the space-time code requires the CSI at both the transmitter and the receiver [36]. While there are multiple ways to do the CSI estimation, in this example, we consider the approach using training sequence. In a MIMO system with t transmitting and r receiving antennas, we use the training sequence to estimate the CSI at the receiver. Namely, we consider the state estimation problem at each receiving antenna, where t states are to be estimated. This estimation problem can be modeled using (1) where \mathbf{x} is the $t \times 1$ state vector of the channels from t transmitting antennas to the specific receiving antenna, \mathbf{z} is composed of the received signals from time 1 up to N by the receiving antenna, \mathbf{H} is the matrix composed of the training sequence from the t transmitting antennas from time 1 up to N and \mathbf{v} is the noise vector.

Generally, the training sequence length N is required to be no less than t to ensure that the system is determined. In some cases, N is much larger than t to make the estimation less vulnerable to abnormal noise and/or interference from other devices. While the accuracy of the estimation is expected to increase with the length of the training sequence, it also brings massive measurements that the estimator need to deal with. Traditionally, the receiver would store all the measurements before using them to do the estimation. However, in some systems, the receiver side is expected to be cheap and small thus there might be not enough space for the massive measurements. We use the block-wise preprocessing to solve this problem. Instead of waiting until the end of the training sequence, the receiver would periodically preprocess the latest measurement block by a matrix \mathbf{L}_i to make it a shorter vector \mathbf{y}_i to be stored where i is the index of the block. The subsequent estimation is done at the end of the training sequence when all the p blocks have been received and preprocessed. Recall in (3) that the estimation is purely dependent on the preprocessed data \mathbf{y}_i 's thus the raw measurement blocks can be discarded immediately after the preprocessing such that a lot of memory space can be saved.

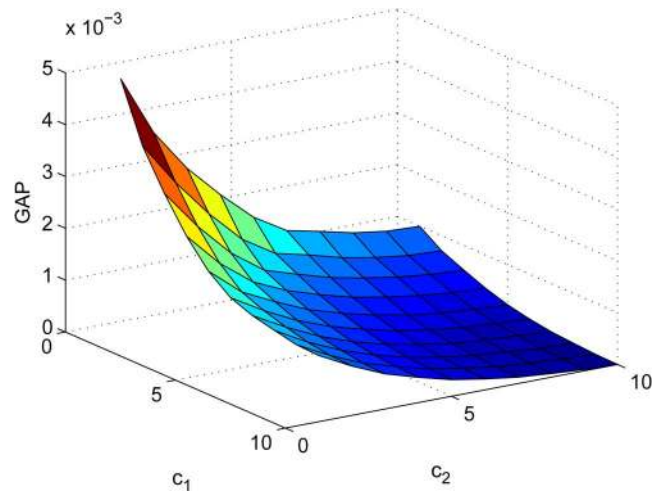


Fig. 2. Performance of lossy preprocessing and estimation with combinations of c_1 and c_2 .

In this example, we assume that the MIMO system is equipped with 10 transmitting antennas thus the dimension of the state is 10. To ensure the performance, the training sequence of length $N = 40$ is employed thus the total number of measurements is 40. The measurements are partitioned equally into two blocks of length 20 with the corresponding measurement matrices \mathbf{H}_1 and \mathbf{H}_2 where $\text{rank}(\mathbf{H}_1) = \text{rank}(\mathbf{H}_2) = 10$. Moreover, in this application, the $\Sigma_{\mathbf{x}}$ is small compared with $\Sigma_{\mathbf{v}}$, which corresponds with the case that the system is with high measurement noise. Assume that the measurement noises are i.i.d, and without loss of generality assume that $\Sigma_{\mathbf{v}} = \mathbf{I}$. Note that the δ_n of \mathbf{A} is bounded by $\delta_n \leq \lambda_{(1)}(\Sigma_{\mathbf{z}} - \mathbf{I}) = \lambda_{(1)}(\mathbf{H}\Sigma_{\mathbf{x}}\mathbf{H}') = 0.2$ in this example, which satisfies the sufficient condition in Theorem 2.

To evaluate the performance of the designed estimator, we define

$$\text{GAP} = E [(\hat{\mathbf{x}} - \hat{\mathbf{x}}_{\text{LMMSE}})^2] \quad (46)$$

where $\hat{\mathbf{x}}$ is the state estimation obtained by the designed estimator. By (4), the MSE of the designed estimator is GAP plus a constant that is independent of the estimator design.

In Fig. 2 and Fig. 3, it is shown that if the GAP is bounded by 5×10^{-3} for all the cases where $c_1 \leq 10, c_2 \leq 10, c_1 \cdot c_2 \neq 0$, which is small compared with the LMMSE that is 0.0417. It is shown that the performance of the proposed algorithm is very good such that it is able to compress the raw measurements into quite small size with little performance loss. Moreover, in Fig. 3, it is shown that the theoretical results match the simulation results obtained by Monte-Carlo simulations.

The algorithm proposed in this paper is compared with Algorithm 1 in [18]. As shown in Fig. 4, the algorithm proposed in this paper has better performance than that in [18], especially in the cases where c_2 is low. Moreover, in this example, the algorithm in [18] converges within less than 10 iterations. However, the performance is not improved with more iterations. In contrast, the algorithm proposed in this paper has good enough performance with 10 iterations while at the same time the performance continues to improve, even after 1000 iterations.

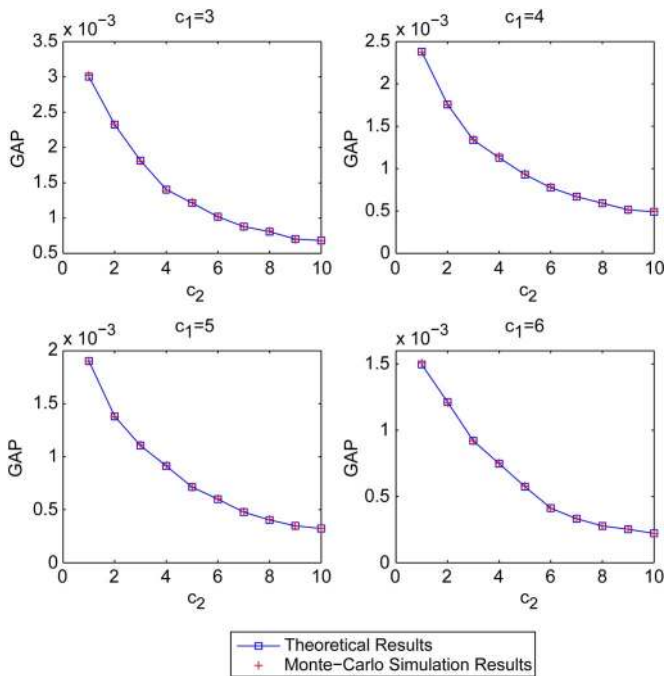


Fig. 3. Performance of lossy preprocessing and estimation with fixed c_1 .

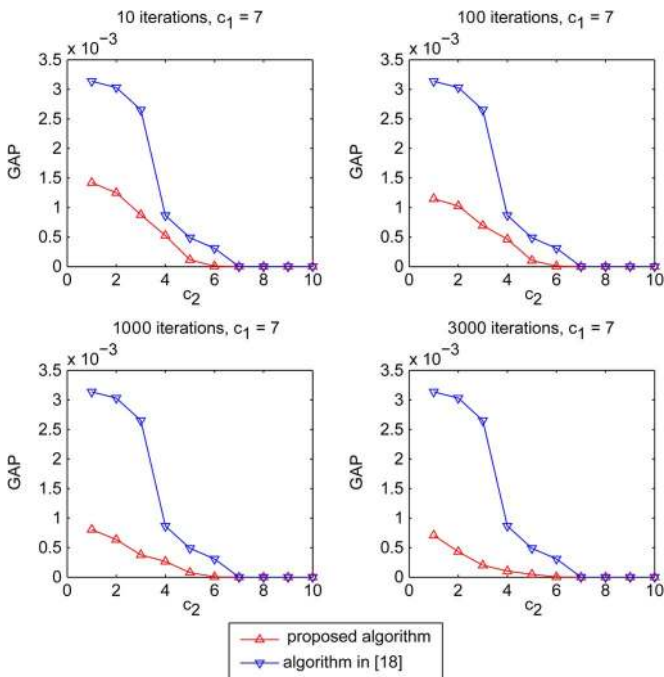


Fig. 4. Comparison with Algorithm 1 in [18] with different number of iterations.

B. Power Grid

In the power grid, there are generally two levels for the power to be delivered from the generation plant to household: the transmission level and the distribution level. The former delivers the power from generation plants to substations and between substations; the latter delivers the power from substations to households locally. They have quite different structures and physical properties.

TABLE II
THE GAP FOR THE CASES $c_1 \leq \text{rank}(\mathbf{H}_1)$, $c_2 \leq \text{rank}(\mathbf{H}_2)$

c_1	c_2	GAP	c_1	c_2	GAP
8	10	4.97×10^{-4}	10	10	2.48×10^{-5}
8	11	5.20×10^{-4}	10	11	1.48×10^{-5}
8	12	5.05×10^{-4}	10	12	1.42×10^{-5}
9	10	8.29×10^{-5}	11	10	6.57×10^{-6}
9	11	7.30×10^{-5}	11	11	6.40×10^{-7}
9	12	7.23×10^{-5}	11	12	0

We consider the state estimation problem in the transmission level, specifically in the IEEE 14 bus system model where there are 14 interconnected substations. Engineers are interested in the complex voltage phase differences in the substations, which is costly to directly measure. One way to get to know these states is to estimate them by power flow measurements which include the power injections at substations and the power flows in each transmission line. We assume that this system is fully measured, i.e., each substation would provide one power injection measurement while each transmission line would provide two power flow measurements, all of which are contaminated by random noise. In this system, there are 14 power injection measurements and 40 power flow measurements, while 13 phase differences are to be estimated. It can be modeled by (1) where \mathbf{H} is the measurement matrix related with the grid topology, \mathbf{z} is composed of the 54 measurements, \mathbf{x} is composed of the 13 system states and \mathbf{v} is the noise vector.

In practice, the substations are usually distantly distributed. Therefore, transmitting a lot of measurements to the estimation center is expensive and sometimes even impossible. We partition the system into several groups and the measurements inside each group is preprocessed to be compatible with the transmission capability of the system. In this example, we partition the substations indexed 1,2,3,5,7,9,10 as one group while the rest as the other group. The power injection measurements are associated with the corresponding substation while the power flow measurements are associated with the originating substation. From Theorem 1, the sufficient and necessary conditions to enable LMMSE estimation is $c_1 \geq \text{rank}(\mathbf{H}_1) = 11$, $c_2 \geq \text{rank}(\mathbf{H}_2) = 12$, where c_1 and c_2 stand for the communication capability from the preprocessor of each group to the estimation center. If these conditions cannot be satisfied, the low rank estimators are designed using the proposed algorithm.

By doing simulation using Matpower [37], Table II illustrates that the GAP decreases asymptotically to 0 as c_1 and c_2 increase. Moreover, it is 0 if and only if both $c_1 \geq 11$ and $c_2 \geq 12$.

Looking at the cases where both $c_1 < \text{rank}(\mathbf{H}_1)$ and $c_2 < \text{rank}(\mathbf{H}_2)$, as shown in Table II and Fig. 5, the GAP is still close to 0 even if c_1 and c_2 are slightly below the $\text{rank}(\mathbf{H}_1)$ and $\text{rank}(\mathbf{H}_2)$. It means that the estimation obtained by the low-rank estimator is quite close to the LMMSE estimation in the mean square sense.

An interesting observation found in Fig. 5 is that there exists a sudden change of the GAP as c_1 and c_2 varies. Moreover, it seems like a boarder line such that if $c_1 + c_2$ is greater than a specific value, then the GAP is insignificant compared with that when $c_1 + c_2$ is below that value. It becomes more clear if we see the examples in Fig. 6. If $c_1 = 5$, the sudden change is at $c_2 = 8$;

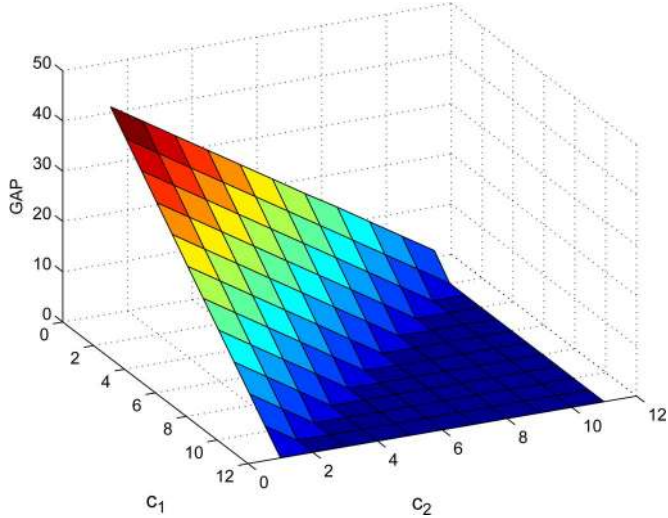


Fig. 5. Performance of lossy preprocessing and estimation with combinations of c_1 and c_2 .

if $c_1 = 6$, the sudden change is at $c_2 = 7$, and so on. The boarder line is $c_1 + c_2 = 13$. The reason why it is 13 is that in this system, the number of system states is 13. In other words, the estimator is using the preprocessed data with size $c_1 + c_2$ to estimate the 13 states of the system. If $c_1 + c_2 < 13$, according to (3), the matrix \mathbf{L} has more rows than columns, i.e., the size of data is smaller than that of the states of the system. In this case, there is no way for any linear preprocessing to keep the linear independence of all the elements in the system state vector, i.e., to preserve enough information of all the system states. On the other hand, if $c_1 + c_2 \geq 13$, i.e., it is possible to linearly combine the measurements to preserve information of all the system states, the algorithm would always find the way to do so by designing the appropriate low-rank estimator with performance close to the LMMSE estimation. In other words, in this application, the proposed algorithm has pushed the linear preprocessing to its limit, which is to compress measurements to totally as few values as the dimension of the system, which demonstrates the effectiveness of the proposed algorithm. In Fig. 6, it is also shown that the theoretical results match the simulation results obtained by 5000 Monte-Carlo trials.

In this example, the n -restricted isometry constant δ_n is hard to obtain [33] and a lower bound can be provided. It is somehow surprising that $\delta_n > 10^4$. Although the condition $\delta_n < \sqrt{5} - 2$ in Theorem 2 no longer holds, the algorithm still converges. Moreover, the performance of the designed estimator is good enough.

Comparing the algorithm proposed in this paper with that in [18] as shown in Fig. 7, at 10 iterations, the two methods perform really close in some cases while in other cases the proposed algorithm has better performance than that in [18]. The performance of the proposed algorithm is further improved by increasing the number of iterations from 10 to 100 while the algorithm in [18] does not have much improvement, where the proposed algorithm outperforms that in [18] in all the cases. Moreover, since it is free of matrix inversion in each iteration, the computation cost of each iteration in the proposed algorithm

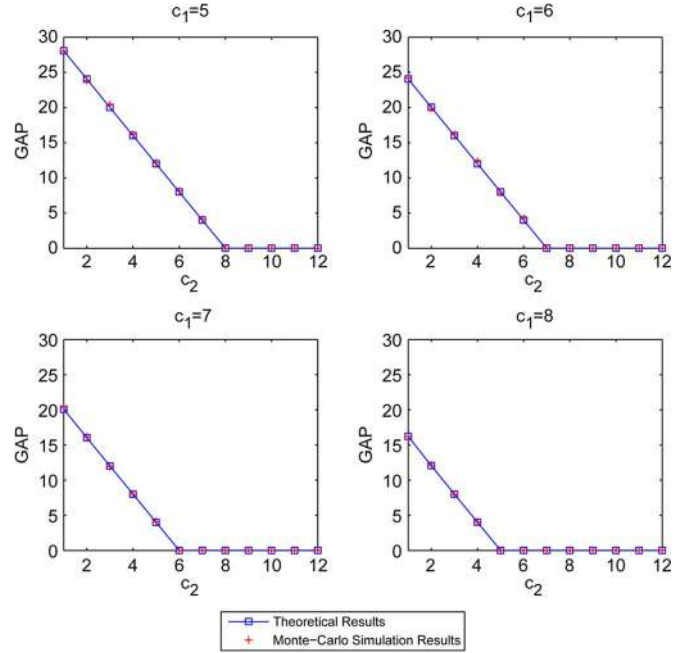


Fig. 6. Performance of lossy preprocessing and estimation with fixed c_1 .

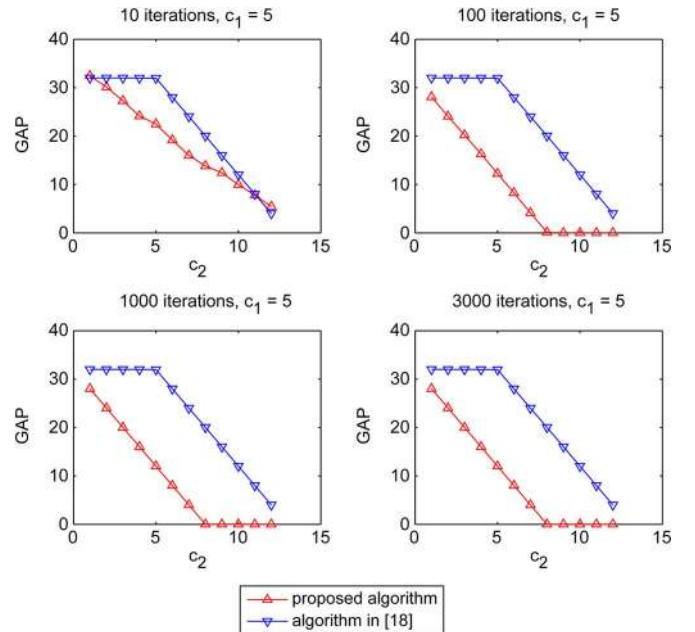


Fig. 7. Comparison with algorithm 1 in [18] with different number of iterations.

is much lower than that in [18]. Therefore, the proposed algorithm can afford more iterations to improve the performance.

C. Target Tracking

Target tracking is one of the primary applications of wireless sensor network (WSN). Each sensor would observed the target and report the measurement to a fusion center for the purpose of tracking, i.e., estimating the location of the target. By combining the measurements from multiple sensors, it can be modeled using (1) where \mathbf{x} is the state containing the location information of the target, \mathbf{H} is the measurement matrix composed of the measurement matrices of all sensors, \mathbf{z} is the vector

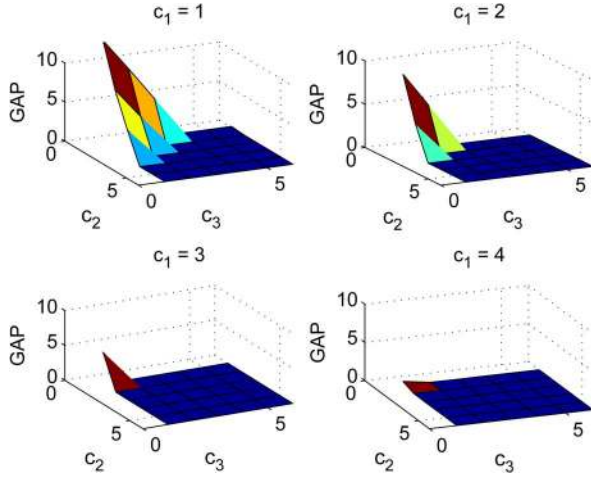


Fig. 8. Performance of lossy preprocessing and estimation with combinations of c_1 , c_2 and c_3 .

containing the measurements from all users and \mathbf{v} is the noise vector.

In reality, there are many inherent constraints in this system. For example, due to the stringent power and spectrum budget, the communication capability of the system is limited. Sometimes it is impossible for the sensor to directly report the measurements to the fusion center.

One of the solutions is using the block-wise structure that the sensors are partitioned into several clusters. Each sensor first reports the measurements to a local substation within the cluster with higher communication capability that is responsible for providing information for the fusion center. However, in some applications, the communication capability of the substation is still not enough for reporting all the raw measurements. In this example, we use the proposed preprocessing method to compress the measurements reported by the sensors at the substation to meet its communication capability.

Suppose that 2 targets that are randomly distributed in a particular area monitored by a sensor network composed by multiple sensors with block-wise structure, and thus the dimension of the system is 6. The sensor network is composed of $p = 30$ sensors that partitioned into 3 clusters, each containing 10 sensors and equipped with a substation. Assume the distribution of the target and the noise are both zero-mean, with variance 5 and 0.1. We use the algorithms proposed in this paper to design the preprocessors and the estimator to satisfy the communication capability from the substation to the fusion center.

In Fig. 8, it is shown that with any fixed c_1 , when both c_2 and c_3 are slightly below the lower bounds $\text{rank}(\mathbf{H}_2) = 6$ and $\text{rank}(\mathbf{H}_3) = 6$, the GAP is close to 0. Recall in (4) that the MSE of the estimation is the sum of GAP and the LMMSE, it is still possible to estimate the target location with satisfactory MSE. For example, if $c_1 = 2$, $c_2 = 3$, $c_3 = 4$, the GAP of the estimation is 0.0004. Since the LMMSE of this problem is 0.02, the MSE of the estimation is 0.0204 which is slightly increased. However, instead of transmitting the 30 measurements, only 9 preprocessed values are transmitted from the substations to the fusion center, saving 70% of the communication load.

In Fig. 9, similar sudden changes in the GAP are observed. It is shown that if $c_1 + c_2 + c_3 \geq 6$, which is the dimension of

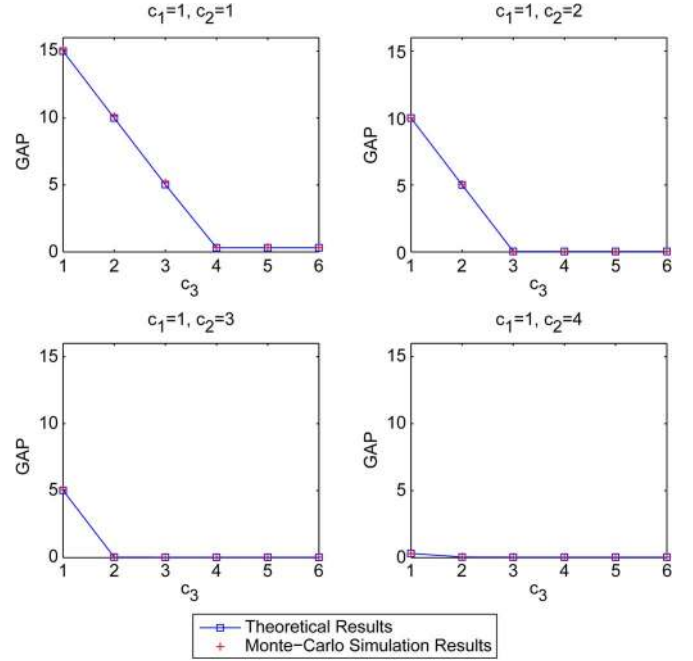


Fig. 9. Performance of lossy preprocessing and estimation with fixed c_1 and c_2 .

this system, the performance is satisfactory. However, if $c_1 + c_2 + c_3 < 6$, then GAP will grow dramatically. Moreover, it also illustrates that the theoretical results match the simulation results obtained by 5000 Monte-Carlo trials.

In the above, it is assumed that the target evolution model is unknown, i.e., we are doing instantaneous estimations at each time slot. However, it is worth noting that the proposed methods could be generalized to the case that the system evolution models are known. In this case, since the measurements in consecutive time slots are correlated, the proposed preprocessing method can be used to combine the measurements obtained by neighboring sensors at consecutive time slots. For example, suppose that a target is moving in a particular area that monitored by a sensor network composed by homogeneous sensors with hierarchical structure modeled by

$$\begin{aligned} \mathbf{x}(k+1) &= \mathbf{P}\mathbf{x}(k) + \mathbf{u}(k) \\ \mathbf{z} &= \mathbf{H}\mathbf{x} + \mathbf{v} \end{aligned} \quad (47)$$

where \mathbf{x} is the state vector, \mathbf{P} is the state evolution matrix and \mathbf{u} is the system noise. \mathbf{z} is the measurement vector, $\mathbf{H} = (\mathbf{H}_1^T \mathbf{H}_2^T \dots \mathbf{H}_p^T)^T$ is the measurement matrix composed of the measurement matrix of p clusters where \mathbf{H}_i is the measurement matrix associated with the cluster indexed i , \mathbf{v} is the noise vector.

In this example, consider the preprocessing at cluster i where the measurements of f time instants are compressed into one single vector. By combining the states of f time instants, define

$$\tilde{\mathbf{x}} \triangleq \begin{pmatrix} \mathbf{x}_1 \\ \mathbf{x}_2 \\ \vdots \\ \mathbf{x}_f \end{pmatrix} \quad (48)$$

where \mathbf{x}_j is the states at the j -th instant. The measurement model for the cluster indexed i is

$$\tilde{\mathbf{z}}(i) = \begin{pmatrix} \mathbf{z}(i, 1) \\ \mathbf{z}(i, 2) \\ \vdots \\ \mathbf{z}(i, f) \end{pmatrix} = \begin{pmatrix} \mathbf{H}_i \\ \mathbf{H}_i \mathbf{P} \\ \vdots \\ \mathbf{H}_i \mathbf{P}^{(f-1)} \end{pmatrix} \tilde{\mathbf{x}} + \mathbf{H}_i \begin{pmatrix} \mathbf{0} \\ \mathbf{u}(1) \\ \vdots \\ \sum_{j=0}^{(f-2)} \mathbf{P}^j \mathbf{u}(f-1-j) \end{pmatrix} + \begin{pmatrix} \mathbf{v}(i, 1) \\ \mathbf{v}(i, 2) \\ \vdots \\ \mathbf{v}(i, f) \end{pmatrix} \quad (49)$$

where $\mathbf{v}(i, j)$ is the measurement noise in cluster i at time instant j , $\mathbf{z}(i, j)$ is the measurements in cluster i at time instant j .

Define

$$\tilde{\mathbf{H}}_i \triangleq \begin{pmatrix} \mathbf{H}_i \\ \mathbf{H}_i \mathbf{P} \\ \vdots \\ \mathbf{H}_i \mathbf{P}^{(f-1)} \end{pmatrix} \quad \tilde{\mathbf{v}}_i \triangleq \mathbf{H}_i \begin{pmatrix} \mathbf{0} \\ \mathbf{u}(1) \\ \vdots \\ \sum_{j=0}^{(f-2)} \mathbf{P}^j \mathbf{u}(f-1-j) \end{pmatrix} + \begin{pmatrix} \mathbf{v}(i, 1) \\ \mathbf{v}(i, 2) \\ \vdots \\ \mathbf{v}(i, f) \end{pmatrix}, \quad (50)$$

then it can be re-written as

$$\tilde{\mathbf{z}}(i) = \tilde{\mathbf{H}}_i \tilde{\mathbf{x}} + \tilde{\mathbf{v}}_i \quad (51)$$

that corresponding with the model in (1) thus the proposed preprocessing method could be applied to compress the measurements. Besides using the correlation of measurements in the spatial domain, the proposed method also utilizes the correlation of the measurements in the time domain to compress them. More specifically, the measurement matrix \mathbf{H}_i enables us to leverage the spatial correlation between the measurements obtained by neighboring sensors. If we further have the knowledge of the temporal correlation from (47), they can be combined in (51) to compress the measurements jointly using spatial domain and time domain correlations of the measurements.

VII. CONCLUSION

In this paper, we addressed the problem of preprocessing the measurements to balance between the estimation accuracy and the data conciseness. It is shown that there is a bound on the compressed data size to enable lossless compression. Moreover, an algorithm to design the preprocessor and the estimator when the required size of the compressed data is below the bound is proposed. The estimator designed by the algorithm has the performance guarantee that the MSE of the estimation is at most a constant factor times the MSE of the global optimal solution under some additional conditions. Moreover, the proposed algorithm has the advantage in both robustness and computation complexity compared with existing works. Three applications are shown to demonstrate the effectiveness of the proposed method.

In the future, it might be interesting to investigate distributed algorithms for designing the estimators to meet time-varying

conditions where the estimators need to be re-designed periodically. Also, in this paper, it was assumed that the grouping of the measurements is fixed. Since different grouping may lead to quite different performance, it might also be interesting to investigate the problem of optimal grouping of the measurements.

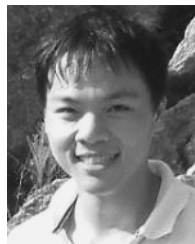
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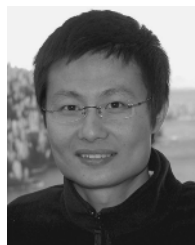
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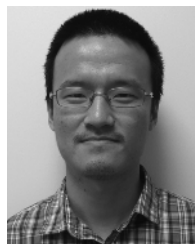
Dr. Chen is the recipient of multiple honors and awards including best paper award from IEEE GLOBECOM in 2013, Future Faculty Fellowship and Distinguished Dissertation Fellowship Honorable Mention from Department of Electrical and Computer Engineering in 2010 and 2011, respectively, Finalist of Deans Doctoral Research Award from A. James Clark School of Engineering at the University of Maryland in 2011, and Chinese Government Award for outstanding students abroad in 2011.



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