

Sampling and Distortion Tradeoffs for Indirect Source Retrieval

Elaheh Mohammadi, Alireza Fallah, and Farokh Marvasti

Abstract—Consider a continuous signal that cannot be observed directly. Instead, one has access to multiple corrupted versions of the signal. The available corrupted signals are correlated because they carry information about the common remote signal. The goal is to reconstruct the original signal from the data collected from its corrupted versions. Known as the indirect or remote reconstruction problem, it has been mainly studied in the literature from an information theoretic perspective. A variant of this problem for a class of Gaussian signals, known as the “Gaussian CEO problem,” has received particular attention; for example, it has been shown that the problem of recovering the remote signal is equivalent with the problem of recovering the set of corrupted signals (separation principle). The information theoretic formulation of the remote reconstruction problem assumes that the corrupted signals are uniformly sampled and the focus is on optimal compression of the samples. On the other hand, in this paper, we revisit this problem from a sampling perspective. More specifically, assuming restrictions on the sampling rate from each corrupted signal, we look at the problem of finding the best sampling locations for each signal to minimize the total reconstruction distortion of the remote signal. In finding the sampling locations, one can take advantage of the correlation among the corrupted signals. The statistical model of the original signal and its corrupted versions adopted in this paper are similar to the one considered for the Gaussian CEO problem; i.e., we restrict to a class of Gaussian signals. Our main contribution is a fundamental lower bound on the reconstruction distortion for any arbitrary nonuniform sampling strategy. This lower bound is valid for any sampling rate. Furthermore, it is tight and matches the optimal reconstruction distortion in low and high sampling rates. Moreover, it is shown that in the low sampling rate region, it is optimal to use a certain nonuniform sampling scheme on all the signals. On the other hand, in the high sampling rate region, it is optimal to uniformly sample all the signals. We also consider the problem of finding the optimal sampling locations to recover the set of corrupted signals, rather than the remote signal. Unlike the information theoretic formulation of the problem in which these two problems were equivalent, we show that they are not equivalent in our setting. Finally, another contribution of this paper is a new reverse majorization inequality that might be of independent interest.

Index Terms—Remote source reconstruction, nonuniform sampling, distortion, majorization.

Manuscript received April 13, 2016; revised November 21, 2016; accepted June 6, 2017. Date of publication August 4, 2017; date of current version October 18, 2017. This work was supported by the Iran National Science Foundation. This paper has been presented in part at the 2016 GlobalSIP.

The authors are with the Department of Electrical Engineering, Advanced Communications Research Institute, Sharif University of Technology, Tehran 11365-11155, Iran (e-mail: marvasti@sharif.edu).

Communicated by A. Montanari, Associate Editor for Statistical Learning. Color versions of one or more of the figures in this paper are available online at <http://ieeexplore.ieee.org>.

Digital Object Identifier 10.1109/TIT.2017.2735960

I. INTRODUCTION

IN MANY applications, such as monitoring or sensing systems, one may be interested in reconstructing a stochastic source $S(t)$ that is not directly observable. Instead, access is provided to k correlated signals $S_1(t), S_2(t), \dots, S_k(t)$ that are corrupted versions of $S(t)$. The goal is to reconstruct $S(t)$ from limited information that one can obtain from $S_i(t), i = 1, 2, \dots, k$. While the source coding aspect of the problem is classical in information theory (see for instance [1, Sec. 3.5]), its signal processing and sampling aspect has not received much attention. In this paper, we study the sampling aspect of this problem for *stochastic* signal $S(t)$ and the k corrupted versions $S_i(t)$. We take into account the fact that the correlation among the signals $S_i(t), i = 1, 2, \dots, k$, can help decrease the sampling rate or improve the signal reconstruction accuracy.

A. System Model

It is known that any *deterministic* continuous function $s(t)$ defined on the interval $[0, T]$ can be expressed in terms of sinusoids as follows:

$$s(t) = \sum_{\ell=0}^{\infty} [a_{\ell} \cos(\ell\omega t) + b_{\ell} \sin(\ell\omega t)], \quad t \in [0, T],$$

where $\omega = 2\pi/T$. If the number of non-zero coefficients a_{ℓ} and b_{ℓ} are limited, the signal $s(t)$ is sparse in the frequency domain. Herein, we consider a *stochastic* signal $S(t)$, and show its Fourier coefficients by random variables A_{ℓ} and B_{ℓ} . We assume that the coefficients A_{ℓ} and B_{ℓ} are zero when $\ell > N_2$ or $\ell < N_1$ for some natural numbers $N_1 \leq N_2$, i.e., $S(t)$ is a bandpass stochastic signal:

$$S(t) = \sum_{\ell=N_1}^{N_2} [A_{\ell} \cos(\ell\omega t) + B_{\ell} \sin(\ell\omega t)], \quad t \in [0, T], \quad (1)$$

where the coefficients A_{ℓ} and B_{ℓ} for $N_1 \leq \ell \leq N_2$ are mutually independent identically distributed (i.i.d.) normal $\mathcal{N}(0, 1)$ variables, i.e., the original signal is white and Gaussian. We cannot observe $S(t)$ directly. Instead, we have $S_1(t), S_2(t), \dots, S_k(t)$, also defined on $t \in [0, T]$, that are corrupted versions of $S(t)$. The corrupted versions of the signal for $i = 1, 2, \dots, k$ can be expressed as

$$S_i(t) = \sum_{\ell=N_1}^{N_2} [A_{i\ell} \cos(\ell\omega t) + B_{i\ell} \sin(\ell\omega t)], \quad t \in [0, T], \quad (2)$$

TABLE I
DEFINITION OF THE MAIN PARAMETERS

Notation	Description
T, ω	T is signal period and $\omega = 2\pi/T$
k	Number of corrupted signals
$S(t)$ and $S_i(t)$	The original signal and the i th corrupted signal, respectively.
A_ℓ, B_ℓ	Fourier series coefficients of the original signal $S(t)$ $\ell \in [N_1 : N_2]$, $A_\ell, B_\ell \sim \mathcal{N}(0, 1)$.
$A_{i\ell}, B_{i\ell}$	Fourier series coefficients of the i th corrupted signal $S_i(t)$ $\ell \in [N_1 : N_2]$, $A_{i\ell}, B_{i\ell} \sim \mathcal{N}(0, 1 + \eta)$
η	Variance of the perturbation added to A_ℓ to produce $A_{i\ell}$ for $1 \leq i \leq k$.
N, N_1 and N_2	The support of the input signal in frequency domain is from $N_1\omega$ to $N_2\omega$. $N = N_2 - N_1 + 1$. $2N$ is the number of free variables of each signal
m_i	Number of noisy samples of the i th corrupted signal
σ_i^2	Variance of the sampling noise of the i th corrupted signal
$\{t_{i1}, t_{i2}, \dots, t_{im_i}\}$	Sampling time instances of the i th corrupted signal
$\phi_i = \frac{m_i}{2\sigma_i^2} + \frac{1}{\eta}$, $\Phi_p = \sum_{i=1}^k \phi_i^p$	Definitions of ϕ_i and Φ_p used in Theorems 1 and 2

where $A_{i\ell} = A_\ell + W_{i\ell}$ and $B_{i\ell} = B_\ell + V_{i\ell}$; here $W_{i\ell}$ and $V_{i\ell}$ are independent perturbations that are added to the original signal. It is assumed that the perturbations $W_{i\ell}$ and $V_{i\ell}$ for $i \in \{1, 2, \dots, k\}$ and $\ell \in \{N_1, N_1 + 1, \dots, N_2\}$ are i.i.d. variables according to $\mathcal{N}(0, \eta)$. The perturbations are also mutually independent of the signal coefficients A_ℓ and B_ℓ for $N_1 \leq \ell \leq N_2$. The statistical model assumed for the coefficients $A_\ell, B_\ell, V_{i\ell}, W_{i\ell}$ parallels the one for the ‘‘Gaussian CEO problem’’ [2], [3]. A summary of the model parameters is given in Table I.

We are allowed to take m_i samples from the i th corrupted signal $S_i(\cdot)$ at time instances $t_{i1}, t_{i2}, \dots, t_{im_i} \in [0, T]$ of our choice, for $i = 1, 2, \dots, k$. Therefore m_i/T can be viewed as the sampling rate of the $S_i(\cdot)$. We assume that the samples are noisy. The sampling noise can model quantization noise of an A/D converter, or the noise incurred by transmitting the samples to a fusion center over a communication channel. The sampling noise of each signal $S_i(t)$ is modeled by an independent zero-mean Gaussian random variable with variance σ_i^2 . We use the samples to reconstruct either the remote signal $S(t)$, or the collection of corrupted signals $S_1(t), S_2(t), \dots, S_k(t)$. The motivation for reconstructing $\{S_i(t), i = 1, \dots, k\}$ is twofold: firstly, this would parallel the literature on indirect source coding, where the reconstruction distortion of intermediate signals is shown to be equivalent with that of the original signal (the separation theorem [4]–[6]). Secondly, these individual signals $S_i(t)$ may contain some other information of interest besides $S(t)$, e.g., the differences $S(t) - S_i(t)$ might be correlated with some other signal of interest.

The reconstruction of the remote signal $S(t)$ and the corrupted signal $S_i(t)$ are denoted by $\hat{S}(t)$ and $\hat{S}_i(t)$, respectively. These reconstructions are calculated using the Minimum Mean Square Error (MMSE) criterion, i.e., $\hat{S}(t)$ is the conditional expectation of the $S(t)$ given all the samples. The goal is to optimize over the sampling times t_{ij} to minimize the distance between the signals and their reconstructions. More

specifically, we consider the minimization

$$D_{a \min} = \min_{\{t_{i1}, t_{i2}, \dots, t_{im_i}\}_{i=1}^k} \frac{1}{T} \int_{t=0}^T \mathbb{E}\{|\hat{S}(t) - S(t)|^2\} dt, \quad (3)$$

for the remote signal, or the minimization

$$D_{b \min} = \min_{\{t_{i1}, t_{i2}, \dots, t_{im_i}\}_{i=1}^k} \frac{1}{T} \int_{t=0}^T \sum_{i=1}^k \mathbb{E}\{|\hat{S}_i(t) - S_i(t)|^2\} dt, \quad (4)$$

for reconstruction of the k corrupted signals $S_i(t)$, $i = 1, 2, \dots, k$. Here t_{ij} is the j th sampling time of the i th signal.

B. Related Works

The problem that we defined above is novel. However, it relates to the literature on *remote signal reconstruction* and *distributed sampling*. The former has been only studied from an information theoretic and source coding perspective, while the latter has been mainly considered in the context of compressed sensing and wireless sensor networks. Finally, while we consider sampling of multiple signals, there are some previous works that study sampling rate and reconstruction distortion of a single source, e.g. see [7]–[10].

1) *Remote Signal Reconstruction*: Reconstruction distortion of correlated signals (lossy reconstruction) is a major theme in multi-user information theory for the class of discrete i.i.d. signals. However, the emphasis in multi-user source coding is generally on the quantization and compression rates of the sources, and not on the sampling rates. It is assumed that the signals are all uniformly sampled at the Nyquist rate. The indirect source coding problem was first introduced by Dobrushin and Tsybakov in information theory literature [6]. This work and subsequent information theoretic ones deal with discrete sources, by assuming that we have several bandlimited signals $S_i(t)$, that are sampled at the Nyquist rate with no distortion at the sampling phase. Then, assuming a

finite quantization rate for storing the samples, the task is to minimize the total reconstruction distortion (which is only due to quantization). On the other hand, our work in this paper is on indirect source *retrieval* and not indirect source *coding*, as we do not study the quantization aspect of the problem. Rather, we focus on the distortion incurred by the sampling rate (which can be below the Nyquist rate), and the additive noise on the samples. We also allow nonuniform sampling to decrease the distortion.

2) *Distributed Sampling*: From another perspective, our problem relates to the distributed sampling literature. In distributed compressed sensing, the structure of correlation among multiple signals is their joint-sparsity. This problem was studied in [11], where signal recovery algorithms using linear equations obtained by distributed sensors were given. Authors in [12] model the correlation of two signals by assuming that one is related to the other by an unknown *sparse* filtering operation. The problem of centralized reconstruction of two correlated signals based on their distributed samples is studied, and its similarities with the Slepian-Wolf theorem in information theoretic distributed compression are pointed out. Motivated by an application in array signal processing, the authors in [13] consider signal recovery for a specific type of correlated signals, assuming that the signals lie in an unknown but low dimensional linear subspace.

Spatio-temporal correlation of the distributed signals is a significant feature of wireless sensor networks and can be utilized for sampling and data collection [14]. In [15], the spatio-temporal sampling rate tradeoffs of a sensor network for minimum energy usage is studied. Authors in [16] provide a mathematical model for the spatio-temporal correlation of the signals observed by the sensor nodes. In [17], the spatio-temporal statistics of the distributed signals are used by the Principal Component Analysis (PCA) method to find transformations that sparsify the signal. Compressed sensing is then used for signal recovery. In [18], a compressive wireless sensing is given for signal retrieval at a fusion center from an ensemble of spatially distributed sensor nodes. See also [19] for a distributed algorithm based on sparse random projections for signal recovery in sensor networks.

C. Our Contributions

Having chosen a particular sampling strategy (such as uniform sampling), one obtains a value for its reconstruction distortion. This value serves as an *upper bound* on the optimal reconstruction distortion. But we are interested to know how close we come to the optimum distortion with this particular sampling strategy. To estimate this, it is desirable to find *fundamental lower bounds* on the reconstruction distortion which hold regardless of the sampling strategy. In this paper, we provide such a lower bound *for any arbitrary sampling rates* (i.e., any arbitrary values for m_i) for both of the problems of reconstructing the remote signal, or the collection of the corrupted signals. Furthermore, this lower bound is shown to coincide with the optimal distortion in the high and low sampling regions. In other words, while our fundamental lower bound applies to any arbitrary sampling rates

(and provides information about the general behavior of the optimal distortion), it is of particular interest in the high and low sampling regions for which it becomes tight.

1) *High Sampling Rate Region*: this refers to the case of $m_i > 2N_2$ for all i , i.e., we are sampling each of the signals above the Nyquist rate. Our result in the high sampling region, it is optimal to use uniform sampling for each signal, and the lower bound matches the distortion yielded by uniform sampling. From a practical perspective, the high sampling rate region is relevant to the case of *high sampling noise* (large σ_i). When each of the samples taken from a signal is very noisy, we desire to oversample. As an example, if the sampling noise is modeling the quantization noise of an A/D converter, we can consider a $\Sigma\Delta$ modulator that oversamples with high quantization noise. If the sampling noise models the channel noise incurred by transmitting the samples to a fusion center over a wireless medium, then oversampling provides a redundancy to combat the channel noise.

2) *Low Sampling Rate Region*: The low sampling rate region refers to the case of $\sum_{i=1}^k m_i \leq N$. Our result for the low sampling rate region states that a certain nonuniform sampling strategy is distortion optimal, and the lower bound matches the distortion yielded by this nonuniform sampling. In progressive or multi resolution applications, one wishes to recover a low-resolution version of a signal, and based on that decide whether to seek a higher resolution version. The low sampling rate region can be helpful in the low-resolution phase. We also make the following comments about the low-sampling rate region:

- Our result allows us to quantify the difference between the following two cases: (i) not taking any samples at all, and (ii) taking a total of N samples $\sum_{i=1}^k m_i \leq N$. By showing that the optimal distortion in case (ii) differs from the optimal distortion in case (i) by at most 3dB, we conclude the following negative result: there is no gain beyond 3dB by using any, however complicated, nonuniform sampling strategy.¹
- If a limitation on the number of samples that we can possibly take is enforced on us as a physical constraint, it is still of interest to know the best possible achievable distortion.

Finally, we comment on a difference between the low and high sampling rate regions. Suppose that we have a total budget on the number of samples $m = \sum_{i=1}^k m_i$ that we can take from the corrupted signals. Then, (i) for low sampling rates, $m \leq N$: if we want to reconstruct either the *remote signal* or the collection of *corrupted signals*, it is best to take the samples from the signal with the smallest sampling noise, i.e., if $\sigma_1 \leq \sigma_i$ for all i , it is optimal to choose $m_1 = m$, $m_i = 0$ for $i > 2$. (ii) for high sampling rates: for reconstructing the *remote signal*, taking more samples from the less noisy signals is advantageous, but to reconstruct the collection of *corrupted signals*, it is no longer true that we should take as many sample as possible from the less noisy signal. The optimum number of samples that we should take from each

¹However, one should also note in sensitive applications, such as radar, extra sampling to improve the resolution by 3dB can be valuable.

signal is an optimization problem, with a solution depending on the parameters of the problem.

D. Proof Techniques

The Gaussian assumption implies that the MMSE and linear MMSE (LMMSE) are identical. Since LMMSE estimator only depends on the second moments, the problem reduces to a linear algebra optimization problem. However, this optimization problem is not easy because the variables we are optimizing over, are sampling locations t_{ij} that show up as arguments of sine and cosine functions. Sine and cosine functions are nonlinear, albeit structured, functions. The goal would be to exploit their structure to solve the optimization problem.

To find fundamental lower bounds for the reconstruction distortion, we utilize various matrix inequalities: these include (i) an inequality in majorization theory that relates trace of a function of a matrix to the diagonal entries of the matrix (see [20, Ch. 2]), (ii) the matrix version of the arithmetic and harmonic means inequality (iii) the Löwner-Heinz theorem for operator convex functions, and (iv) more importantly a new reverse majorization inequality (Theorem 3). A contribution of this paper is this reverse majorization inequality that might be of independent interest. Majorization inequalities state that the diagonal entries of a Hermitian matrix F are majorized by the eigenvalues of F . Therefore, if F_{Diag} is a diagonal matrix wherein we have kept the diagonal entries of F and set the off-diagonals to zero, we will have

$$\text{Tr} [F^{-1}] \geq \text{Tr} [F_{\text{Diag}}^{-1}] \quad (5)$$

Our reverse majorization inequality goes in the reverse direction. For certain matrices F and G of our interest, we show that

$$\text{Tr} [F^{-1}G] \leq \text{Tr} [F_{\text{Diag}}^{-1}G_{\text{Diag}}]. \quad (6)$$

E. Notation and Organization

Uppercase letters are used for random variables and matrices, whereas lowercase letters show (non-random) values. Vectors are denoted by lowercase bold letters (such as \mathbf{x}), and random vectors are denoted by either uppercase bold letters (such as \mathbf{X}) or bold sans-serif letters (such as \mathbf{X}). The covariance of a random vector \mathbf{X} is denoted by $C_{\mathbf{X}}$. Given a matrix A , A_{Diag} denotes the matrix formed by keeping the diagonal entries of A and changing the off-diagonal entries to zero. Given a vector \mathbf{x} , $\text{Diag}(\mathbf{x})$ denotes the diagonal matrix where its diagonal entries are coordinates of \mathbf{x} . The symbol \oplus is used for the direct sum and \otimes is used for the Kronecker product of matrices. We write $A \leq B$ if $B - A$ is positive semi-definite. For a Hermitian matrix A with eigen decomposition PDP^{-1} and real function f , $f(A)$ is defined as $Pf(D)P^{-1}$ in which $f(D)$ is a diagonal matrix, where function f is applied to the diagonal entries of D .

The paper is organized as follows: In Section II, the main results of the paper are presented. In Section III, the problem formulation is derived. In Section IV, the proofs of the main results are given. Finally, in Appendices A and B, the mathematical tools and technical details which have been

used in the main proofs are given. In Appendix B, a new reverse majorization theorem is derived which might be of independent interest in linear algebra.

II. MAIN RESULTS

Let $N = N_2 - N_1 + 1$. We make the following definitions:

- We call $(N_2 - N_1 + 1)f_0 = Nf_0$ the *signal bandwidth* (of the bandlimited signal), where $f_0 = 1/T$.
- We call $2N_2f_0$ the *Nyquist rate* (twice the maximum frequency of the signal).
- We call $m_i/T = m_if_0$ the *sampling rate* of the i -th corrupted signal. It is the number of total samples from $S_i(t)$ in $[0, T]$, divided by period length T . If we periodically extend the m_i samples (periodic nonuniform sampling), m_i/T will be the number of samples taken per unit time, hence called the sampling rate.

To state the main results, we need a definition. For any real p , let $\phi_i = \frac{m_i}{2\sigma_i^2} + \frac{1}{\eta}$, and

$$\Phi_p = \sum_{i=1}^k \phi_i^p. \quad (7)$$

Theorem 1 (Reconstruction of the Original Signal): The following general lower bound on the optimal distortion (given in (3)) holds for any given sampling rates:

$$D_{\text{a min}} \geq \max \left(N - N \sum_{i=1}^k \frac{m_i}{2(1+\eta)N + 2\sigma_i^2}, \frac{N}{\left(1 + \frac{k}{\eta}\right) - \frac{k^2}{\eta^2}(\Phi_1)^{-1}} \right), \quad (8)$$

where Φ_1 is defined as in (7). Furthermore, the lower bound given in (8) is tight (the inequality is an equality) in the following cases:

- when $\sum_{i=1}^k m_i \leq N$: in this case, the optimal sampling points, t_{ij} , are all distinct for $1 \leq i \leq k, 1 \leq j \leq m_i$, and belong to the set $\{0, T/N, \dots, (N-1)T/N\}$.
- when $m_i > 2N_2$ for $1 \leq i \leq k$: in this case uniform sampling of each signal $S_i(t)$ is optimal.

Remark 1: From optimality of the lower bound for large values of m_i , we obtain that

$$\lim_{\forall i: m_i \rightarrow \infty} D_{\text{a min}} = \frac{N}{1 + \frac{k}{\eta}} > 0,$$

is strictly positive. The reason is that $m_i = \infty$ implies full access to the set of corrupted signals, $\{S_i(t)\}$, but even in this case we cannot perfectly reconstruct $S(t)$ if the corruption variance $\eta > 0$.

Proof: To prove the theorem, it suffices to show that for any arbitrary choice of sampling time instances, we have

$$D_{\text{a}} \geq N - N \sum_{i=1}^k \frac{m_i}{2(1+\eta)N + 2\sigma_i^2} \quad (9)$$

and moreover, equality in the above equation holds if $\sum_{i=1}^k m_i \leq N$. This claim is shown in Section IV-A; in this section, the optimality of choosing t_{ij} from

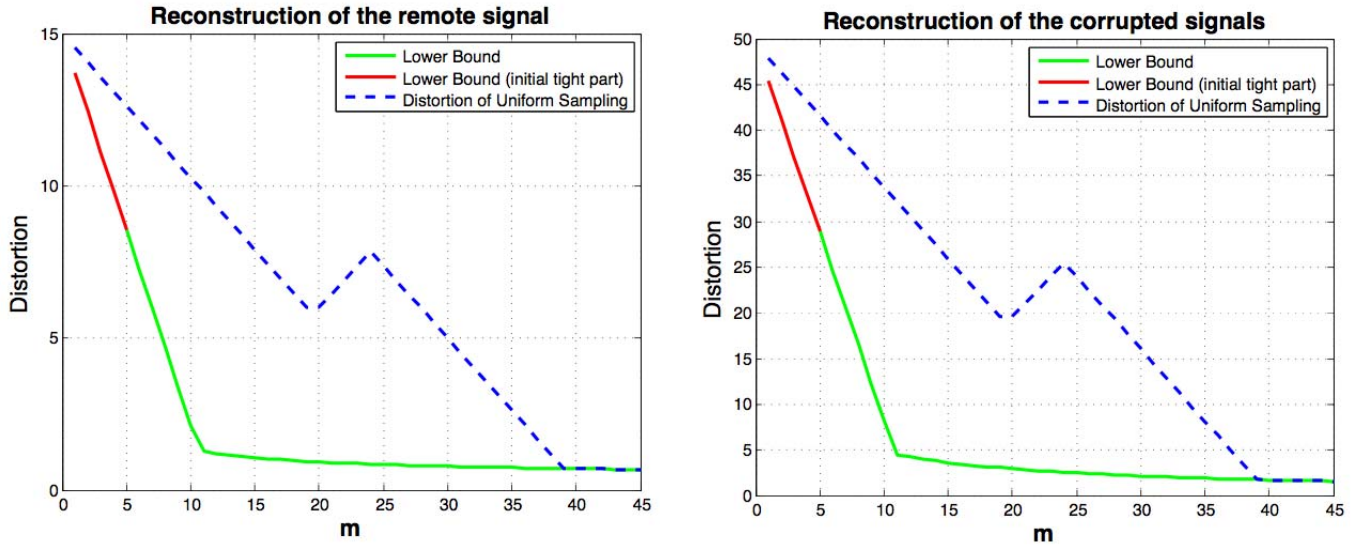


Fig. 1. The the lower bound versus the distortion obtained by the uniform sampling strategy. It is assumed that $m_i = m$ for all i . The parameters for this figure are $(k, N_1, N, N_2, \sigma_i^2, \eta) = (3, 5, 15, 19, 1, 0.1)$.

$\{0, T/N, \dots, (N-1)T/N\}$ is also established. Next, we also show that

$$D_{\text{amin}} \geq \frac{N}{(1+\frac{k}{\eta}) - \frac{k^2}{\eta^2} (\Phi_1)^{-1}}, \quad (10)$$

and furthermore equality in the above equation holds if $m_i > 2N_2$. The proof of this claim is given in Section IV-B; in this section, the optimality of uniform sampling is also established. This completes the proof. ■

Theorem 2 (Reconstruction of the Set of Corrupted Signals): The general lower bound on the optimal distortion (due to (4)) is given in (11) on top of the next page, in which Φ_{-1}, Φ_{-2} are defined as in (7). Furthermore, this lower bound is achieved in the following two cases:

- when $m = \sum_{i=1}^k m_i \leq N$: in this case, the optimal sampling points, t_{ij} , are all distinct for $1 \leq i \leq k, 1 \leq j \leq m_i$, and belong to the set $\{0, T/N, \dots, (N-1)T/N\}$.
- when $m_i > 2N_2$ for $1 \leq i \leq k$: in this case uniform sampling of each signal $S_i(t)$ is optimal.

Proof: We first show that

$$D_{\text{bmin}} \geq Nk(1+\eta) - N \left((1+\eta)^2 + (k-1) \right) \sum_{i=1}^k \frac{m_i}{2(1+\eta)N + 2\sigma_i^2}. \quad (12)$$

The proof of this claim is given in Section IV-C. There, we also prove that when $\sum_{i=1}^k m_i \leq N$, the equality in the above equation holds and the optimal time instances are distinct t_{ij} chosen from $\{0, T/N, \dots, (N-1)T/N\}$. Next, we show that

$$D_{\text{bmin}} \geq N\Phi_{-1} + \frac{N}{\eta(\eta+k) - \Phi_{-1}} \Phi_{-2}, \quad (13)$$

and furthermore, equality in the above equation holds if $m_i > 2N_2$. The proof is given in Section IV-D where we show that the optimal points are uniform samples of the corrupted signals. ■

Finally, as a technical result, we also derive a new reverse majorization inequality, given in Theorem 3 (B), which might be of interest in linear algebra.

In Fig. 1, the lower bounds are plotted assuming that $m_i = m$ for all i , for $k = 3, N_1 = 5, N = 15, N_2 = 19, \sigma_i^2 = 1, \eta = 0.1$. Also, the distortion of the uniform sampling strategy is also plotted, and serves as an upper bound for the optimal distortion curve; thus, optimal distortion curve lies in between the two curves. The two curves match when $m > 2N_2 = 38$. Also, the lower bound is known to be tight in the low sampling rates. This part of the lower bound is drawn in color red. While Fig. 1 is plotted for $\sigma_i^2 = 1, \eta = 0.1$, we observe from numerical simulation that the gap between the lower and upper bound decreases as we increase the sampling or corruption noises.

Discussion 1: We know the exact value of optimal distortion for $m_i > 2N_2$. As argued in Section I.C, this corresponds to the high sampling rate region and is practically relevant when we have high sampling noise. If the sampling noise models the quantization error of an A/D converter, this result allows us to answer the question of whether it is better to collect some accurate samples from the signals, or collect many more less accurate samples from them (a problem related to selecting an appropriate $\Sigma\Delta$ modulator).

Discussion 2: Assume that the sampling noise variances of the corrupted signals satisfy $\sigma_1^2 \leq \sigma_2^2 \leq \dots \leq \sigma_k^2$. Let us assume that we have a fixed total budget of m samples that we can distribute among the k signals, i.e., $\sum_{i=1}^k m_i = m$. Then if $m \leq N$, the lower bound will be tight. Regardless of whether we want to construct the original signal or the collection of corrupted signals, one can verify that the total distortion (subject to $\sum_{i=1}^k m_i = m$) is minimized when $m_1 = m$ and $m_i = 0$ for $i > 1$, i.e., all of the samples are taken from the first signal, $S_1(t)$, which has the minimum sampling noise variance.

$$D_{b\min} \geq \max \left(Nk(1+\eta) - N \left((1+\eta)^2 + (k-1) \right) \sum_{i=1}^k \frac{m_i}{2(1+\eta)N + 2\sigma_i^2}, N\Phi_{-1} + \frac{N}{\eta(\eta+k) - \Phi_{-1}} \Phi_{-2} \right), \quad (11)$$

However, the problems of reconstructing the remote signal and the set of corrupted signals are not equivalent. To see this assume that we keep the constraint $\sum_{i=1}^k m_i = m$ for some $m > 2N_2k$, and further assume that $m_i > 2N_2$. For the reconstruction of the *remote signal*, it can be verified that the total distortion (subject to $\sum_{i=1}^k m_i = m$, $m_i > 2N_2$) is minimized when m_i for $i \geq 2$ reaches its minimum of $2N_2 + 1$, which means that we take as many samples as possible from the signal with minimum sampling noise variance $S_1(t)$. On the other hand, if we wish to minimize the total distortion of the entire *corrupted signals*, the optimum value for m_i can vary depending on the value of the parameters; it is not necessarily true that it is better to sample from the less noisy signal. The intuitive reason for this is as follows: if sampling noise of a signal is very high, we need to take lots of samples from it to be able to have a reconstruction with low distortion. However, if the sampling noise of a signal is very low, we are able to have a good reconstruction with few samples; taking more samples yields a negligible improvement in distortion. Therefore, if we aim to reconstruct *all* the signals (i.e., to minimize the sum of the distortions of the signals), the most economic way could be taking less samples from the less noisy signal. For instance, let us consider the case of two signals, $k = 2$, and take the corruption noise variance to be $\eta = 0.5$. Assume that samples from the first signal are taken with variance $\sigma_1 = 1$, which is less than $\sigma_2 = 10$, the noise variance of samples from the second signal. Then for total sample budget $m = 100$, the optimum (m_1, m_2) subject to $m_i > 2N_2$ is given in the following for different values of N_2 . For $N_2 = 20$, the optimal choice is $(m_1, m_2) = (41, 59)$; observe that it is better to take 59 samples from the signal with more sampling noise. For $N_2 = 10$, the optimal choice is $(m_1, m_2) = (32, 68)$. Here $m_i > 2N_2 = 20$ and the optimal choice is not one of the boundary pairs $(m_1, m_2) = (21, 79)$ or $(79, 21)$.

III. PROBLEM FORMULATION

In this section, we state the matrix representation of our problem. A summary of the notation introduced in this section is given in Table II. Let \mathbf{X} be a column vector, consisting of the Fourier coefficients of $S(t)$

$$\mathbf{X} = [A_{N_1}, A_{N_1+1}, \dots, A_{N_2}, B_{N_1}, \dots, B_{N_2}]^T, \quad (14)$$

where T is used for the transpose operation. Similarly, \mathbf{X}_i is a column vector, consisting of the coefficients of $S_i(t)$

$$\mathbf{X}_i = [A_{iN_1}, A_{i(N_1+1)}, \dots, A_{iN_2}, B_{iN_1}, \dots, B_{iN_2}]^T \quad (15)$$

$$= \mathbf{X} + [V_{iN_1}, \dots, V_{iN_2}, W_{iN_1}, \dots, W_{iN_2}]^T \quad (16)$$

$$= \mathbf{X} + \Delta_i, \quad (17)$$

where $\Delta_i = [V_{iN_1}, \dots, V_{iN_2}, W_{iN_1}, \dots, W_{iN_2}]^T$. We assume that \mathbf{X} consists of mutually independent Gaussian random

variables $\mathcal{N}(0, 1)$, i.e., the signal $S(t)$ is white. Therefore, $C_{\mathbf{X}} = I_{2N \times 2N}$. Moreover, the random variables V_{ij} and W_{ij} are assumed to be independent with the probability distribution $\mathcal{N}(0, \eta)$ for some $\eta > 0$.

Vectors \mathbf{X}_i and \mathbf{X}_j are correlated, because they are both corrupted versions of \mathbf{X} . Their cross covariance can be computed as $C_{\mathbf{X}_i \mathbf{X}_i} = (1+\eta)I$ and $C_{\mathbf{X}_i \mathbf{X}_j} = I$ for $i \neq j \in \{1, 2, \dots, k\}$. One can verify that for any j , the covariance matrix for the k random variables A_{ij} for $i = 1, 2, \dots, k$ is equal to

$$\Lambda = \begin{pmatrix} 1+\eta & 1 & 1 & \dots & 1 \\ 1 & 1+\eta & 1 & \dots & 1 \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ 1 & 1 & 1 & \dots & 1+\eta \end{pmatrix}_{k \times k}. \quad (18)$$

Suppose that the i th signal, $S_i(t)$, is sampled at time instances t_{ij} for $i = 1, 2, \dots, k$ and $j = 1, 2, \dots, m_i$. Hence,

$$S_i(t_{ij}) = \sum_{\ell=N_1}^{N_2} [A_{i\ell} \cos(\ell\omega t_{ij}) + B_{i\ell} \sin(\ell\omega t_{ij})].$$

To represent the problem in a matrix form, we define \mathbf{S}_i to be the vector of samples as

$$\mathbf{S}_i = [S_i(t_{i1}), S_i(t_{i2}), \dots, S_i(t_{im_i})]^T.$$

Therefore, we have $\mathbf{S}_i = Q_i \mathbf{X}_i$, where \mathbf{X}_i is defined in (17) and Q_i is an $m_i \times 2(N_2 - N_1 + 1) = m_i \times 2N$ matrix of the form of (19) given on top of the next page.

Moreover, the observation vector for the i th signal is of the following form

$$\mathbf{Y}_i = \mathbf{S}_i + \mathbf{Z}_i = Q_i \mathbf{X}_i + \mathbf{Z}_i = Q_i \mathbf{X} + Q_i \Delta_i + \mathbf{Z}_i,$$

in which \mathbf{Z}_i is the i th noise vector with covariance of $C_{\mathbf{Z}_i} = \sigma_i^2 I_{m_i \times m_i}$.

Now, we define the vector of coefficients of all the k signals, \mathbf{X} , the vector of all the samples, \mathbf{S} , the vector of all the observations, \mathbf{Y}, \dots , as follows:

$$\mathbf{X} = [\mathbf{X}_1^T, \mathbf{X}_2^T, \dots, \mathbf{X}_k^T]^T, \quad (20)$$

$$\mathbf{S} = [\mathbf{S}_1^T, \mathbf{S}_2^T, \dots, \mathbf{S}_k^T]^T, \quad (21)$$

$$\mathbf{Y} = [\mathbf{Y}_1^T, \mathbf{Y}_2^T, \dots, \mathbf{Y}_k^T]^T, \quad (22)$$

$$\mathbf{Z} = [\mathbf{Z}_1^T, \mathbf{Z}_2^T, \dots, \mathbf{Z}_k^T]^T, \quad (23)$$

$$\Delta = [\Delta_1^T, \Delta_2^T, \dots, \Delta_k^T]^T, \quad (24)$$

$$Q_a = [Q_1^T, Q_2^T, \dots, Q_k^T]^T. \quad (25)$$

Furthermore, let

$$Q_b = \bigoplus_{i=1}^k Q_i \quad (26)$$

be the direct sum of the individual matrices Q_i . Then, we can write

$$\mathbf{Y} = \mathbf{S} + \mathbf{Z} = Q_b \mathbf{X} + \mathbf{Z} = Q_a \mathbf{X} + Q_b \Delta + \mathbf{Z}. \quad (27)$$

$$Q_i = \begin{pmatrix} \cos(N_1\omega t_{i1}) & \cos((N_1 + 1)\omega t_{i1}) & \cdots & \cos(N_2\omega t_{i1}) & \sin(N_1\omega t_{i1}) & \sin((N_1 + 1)\omega t_{i1}) & \cdots & \sin(N_2\omega t_{i1}) \\ \cos(N_1\omega t_{i2}) & \cos((N_1 + 1)\omega t_{i2}) & \cdots & \cos(N_2\omega t_{i2}) & \sin(N_1\omega t_{i2}) & \sin((N_1 + 1)\omega t_{i2}) & \cdots & \sin(N_2\omega t_{i2}) \\ \vdots & \vdots & \ddots & \vdots & \vdots & \vdots & \ddots & \vdots \\ \cos(N_1\omega t_{im_i}) & \cos((N_1 + 1)\omega t_{im_i}) & \cdots & \cos(N_2\omega t_{im_i}) & \sin(N_1\omega t_{im_i}) & \sin((N_1 + 1)\omega t_{im_i}) & \cdots & \sin(N_2\omega t_{im_i}) \end{pmatrix}. \quad (19)$$

One can verify that the covariance matrix of the noise is

$$C_Z = \bigoplus_{i=1}^k C_{Z_i} = \bigoplus_{i=1}^k \sigma_i^2 I_{m_i \times m_i}. \quad (28)$$

Moreover, $C_X = I_{2N \times 2N}$ and $C_{\mathbf{X}} = \Lambda \otimes I_{2N \times 2N}$, where Λ was given in (18), and $N = N_2 - N_1 + 1$.

A. Reconstruction of the Remote Signal $S(t)$ and Its Corrupted Version $S_i(t)$

Here, we first state a lemma, which is frequently used in the formulation and proofs of our problem. The Lemma provides the two alternative forms of the LMMSE estimator and the mean squared error. Next we use this lemma to formulate the reconstruction of the remote signal $S(t)$ and its corrupted versions $S_i(t)$ in the subsequent subsections.

Lemma 1 [21]: Suppose that $\mathbf{Y} = \mathbf{A}\mathbf{X} + \mathbf{Z}$ in which \mathbf{Y} is an observation vector, \mathbf{A} is a known matrix, \mathbf{X} is a vector to be estimated and \mathbf{Z} is an additive noise vector. In the case \mathbf{X} and \mathbf{Z} are mutually independent Gaussian vectors, LMMSE is optimal and the estimator and the mean square error, respectively, are given by

$$\begin{aligned} \hat{\mathbf{x}}_{\text{MMSE}}(\mathbf{y}) &= \mathbb{E}\{\mathbf{X}|\mathbf{y}\} = \mathbf{W}\mathbf{y}, \\ \mathbb{E}\|\mathbf{X} - \hat{\mathbf{X}}\|^2 &= \mathbb{E}_{\mathbf{Y}}\{\text{Var}[\mathbf{X}|\mathbf{Y}]\} = \text{Tr}(C_e), \end{aligned} \quad (29)$$

where the reconstruction matrix, \mathbf{W} , and the error covariance matrix, C_e , are of the following forms:

$$\begin{aligned} \mathbf{W} &= C_{\mathbf{X}\mathbf{Y}}C_{\mathbf{Y}}^{-1} = C_{\mathbf{X}}\mathbf{A}^{\text{T}}(AC_{\mathbf{X}}\mathbf{A}^{\text{T}} + C_{\mathbf{Z}})^{-1}, \\ C_e &= C_{\mathbf{X}} - C_{\mathbf{X}\mathbf{Y}}C_{\mathbf{Y}}^{-1}C_{\mathbf{Y}\mathbf{X}} \\ &= C_{\mathbf{X}} - C_{\mathbf{X}}\mathbf{A}^{\text{T}}(AC_{\mathbf{X}}\mathbf{A}^{\text{T}} + C_{\mathbf{Z}})^{-1}AC_{\mathbf{X}}. \end{aligned} \quad (30)$$

Or alternatively [22], using the matrix identity

$$C_{\mathbf{X}}\mathbf{A}^{\text{T}}(AC_{\mathbf{X}}\mathbf{A}^{\text{T}} + C_{\mathbf{Z}})^{-1} = (\mathbf{A}^{\text{T}}C_{\mathbf{Z}}^{-1}\mathbf{A} + C_{\mathbf{X}}^{-1})^{-1}\mathbf{A}^{\text{T}}C_{\mathbf{Z}}^{-1}, \quad (31)$$

the matrices \mathbf{W} and C_e are given by

$$\begin{aligned} \mathbf{W} &= (\mathbf{A}^{\text{T}}C_{\mathbf{Z}}^{-1}\mathbf{A} + C_{\mathbf{X}}^{-1})^{-1}\mathbf{A}^{\text{T}}C_{\mathbf{Z}}^{-1}, \\ C_e &= (\mathbf{A}^{\text{T}}C_{\mathbf{Z}}^{-1}\mathbf{A} + C_{\mathbf{X}}^{-1})^{-1}. \end{aligned} \quad (32)$$

1) *Reconstruction of $S(t)$:* Here, the goal is to reconstruct $S(t)$ with minimum distortion using the observation vector \mathbf{Y} . We use the MMSE criterion to minimize the average distortion subject to the samples. From the Parseval's theorem, we have

$$D_a = \frac{1}{T} \int_{t=0}^T \mathbb{E}\{|\hat{S}(t) - S(t)|^2\} dt = \frac{1}{2} \mathbb{E}\|\mathbf{X} - \hat{\mathbf{X}}\|^2, \quad (33)$$

where $\hat{S}(t)$ is the reconstructed signal and $\hat{\mathbf{X}}$ is the MMSE reconstruction of the coefficient vector \mathbf{X} from the observation vector \mathbf{Y} . Since the random variables are jointly Gaussian, the MMSE estimator is optimal. From the equation

$$\mathbf{Y} = \mathbf{Q}_a\mathbf{X} + \tilde{\mathbf{Z}},$$

where $\tilde{\mathbf{Z}} = \mathbf{Q}_b\mathbf{\Delta} + \mathbf{Z}$, the error of the linear MMSE estimator is equal to

$$\mathbb{E}\|\mathbf{X} - \hat{\mathbf{X}}\|^2 = \mathbb{E}_{\mathbf{Y}}\{\text{Var}[\mathbf{X}|\mathbf{Y}]\} = \text{Tr}(C_e^a), \quad (34)$$

where C_e^a has the following two alternative forms

$$C_e^a = C_{\mathbf{X}} - C_{\mathbf{X}}\mathbf{Q}_a^{\text{T}}(\mathbf{Q}_aC_{\mathbf{X}}\mathbf{Q}_a^{\text{T}} + C_{\tilde{\mathbf{Z}}})^{-1}\mathbf{Q}_aC_{\mathbf{X}}, \quad (35)$$

$$= (\mathbf{Q}_a^{\text{T}}C_{\tilde{\mathbf{Z}}}^{-1}\mathbf{Q}_a + C_{\mathbf{X}}^{-1})^{-1}. \quad (36)$$

In the above formula, the covariance matrix of $\tilde{\mathbf{Z}}$ is

$$C_{\tilde{\mathbf{Z}}} = \mathbf{Q}_bC_{\mathbf{\Delta}}\mathbf{Q}_b^{\text{T}} + C_{\mathbf{Z}} = \eta\mathbf{Q}_b\mathbf{Q}_b^{\text{T}} + C_{\mathbf{Z}}. \quad (37)$$

2) *Reconstruction of $S_i(t)$ for $i = 1, 2, \dots, k$:* Here, the goal is to reconstruct all the k signals with minimum distortion using the observation vector \mathbf{Y} . Again, from the Parseval's theorem, we have

$$D_b = \frac{1}{T} \sum_{i=1}^k \int_{t=0}^T \mathbb{E}\{|\hat{S}_i(t) - S_i(t)|^2\} dt = \frac{1}{2} \mathbb{E}\|\mathbf{X} - \hat{\mathbf{X}}\|^2, \quad (38)$$

in which $\hat{S}(t)$ and $\hat{\mathbf{X}}$ are the reconstructed signal and the estimated coefficients, respectively. From the equation

$$\mathbf{Y} = \mathbf{Q}_b\mathbf{X} + \mathbf{Z},$$

the LMMSE error is equal to

$$\mathbb{E}\|\mathbf{X} - \hat{\mathbf{X}}\|^2 = \mathbb{E}_{\mathbf{Y}}\{\text{Var}[\mathbf{X}|\mathbf{Y}]\} = \text{Tr}(C_e^b), \quad (39)$$

where C_e^b has the following two alternative forms

$$C_e^b = C_{\mathbf{X}} - C_{\mathbf{X}}\mathbf{Q}_b^{\text{T}}(\mathbf{Q}_bC_{\mathbf{X}}\mathbf{Q}_b^{\text{T}} + C_{\mathbf{Z}})^{-1}\mathbf{Q}_bC_{\mathbf{X}}, \quad (40)$$

$$= (\mathbf{Q}_b^{\text{T}}C_{\mathbf{Z}}^{-1}\mathbf{Q}_b + C_{\mathbf{X}}^{-1})^{-1}. \quad (41)$$

B. Some Helpful Facts

We provide a number of facts about the matrices that we have introduced before. These facts can be directly verified, and will be repeatedly used in the proofs. We have listed these facts here to improve the presentation of the proofs.

TABLE II
DEFINITION OF AUXILIARY PARAMETERS

Notation	Description	Helpful properties
\mathbf{X}	Column vector of size $2N$ of Fourier coefficients of $S(t)$	$C_{\mathbf{X}} = I_{2N \times 2N}$
\mathbf{X}_i	Column vector of size $2N$ of Fourier coefficients of $S_i(t)$ for $i = 1, 2, \dots, k$	$C_{\mathbf{X}_i} = (1 + \eta)I$ $C_{\mathbf{X}_i \mathbf{X}_j} = I$ for $i \neq j$
\mathbf{X}	Vectors \mathbf{X}_i Stacked on the bottom of each other; length $2Nk$	$C_{\mathbf{X}} = \Lambda \otimes I_{2N \times 2N}$
Λ	Symmetric $k \times k$ matrix given in (18)	
Γ	$\Gamma = \Lambda^{-1}$; Explicit formula given in (91)	
Δ_i	The i th corruption vector: $\Delta_i = \mathbf{X}_i - \mathbf{X}$	$C_{\Delta_i} = \eta I_{2N \times 2N}$
Δ	Column vectors Δ_i Stacked on the bottom of each other	$C_{\Delta} = \eta I_{2Nk \times 2Nk}$
\mathbf{S}_i	Column vector of size m_i ; Samples of $S_i(t)$ at t_{ij} , $j = 1, \dots, m_i$	$\mathbf{S}_i = Q_i \mathbf{X}_i$
Q_i	$m_i \times 2N$ Matrix specified by harmonics at t_{ij} , given in (19)	
Q_a	Matrices Q_i stacked on the bottom of each other; size: $(\sum m_i) \times 2N$	
Q_b	Direct sum of matrices Q_i ; size: $(\sum m_i) \times 2Nk$	
\mathbf{Y}_i	Noisy samples of the i th signal $S_i(t)$; length m_i	$\mathbf{Y}_i = \mathbf{S}_i + \mathbf{Z}_i$
\mathbf{Y}	Vectors \mathbf{Y}_i Stacked on the bottom of each other; length $m = \sum m_i$	
\mathbf{Z}_i	Sampling noise vector of the i th signal $S_i(t)$; length m_i	$C_{\mathbf{Z}_i} = \sigma_i^2 I_{m_i \times m_i}$
\mathbf{Z}	Vectors \mathbf{Z}_i Stacked on the bottom of each other; length $m = \sum m_i$	$C_{\mathbf{Z}} = \bigoplus_{i=1}^k \sigma_i^2 I_{m_i \times m_i}$
$\tilde{\mathbf{Z}}$	$\tilde{\mathbf{Z}}$ is defined as $Q_b \Delta + \mathbf{Z}$. It appears in $\mathbf{Y} = Q_a \mathbf{X} + \tilde{\mathbf{Z}}$.	$C_{\tilde{\mathbf{Z}}} = \eta Q_b Q_b^T + C_{\mathbf{Z}}$

- 1) The matrices $Q_b Q_b^T$ and $Q_a Q_a^T$ are $m \times m$ matrices of the following form

$$Q_b Q_b^T = \bigoplus_{i=1}^k Q_i Q_i^T = \begin{pmatrix} Q_1 Q_1^T & 0 & 0 & \dots & 0 \\ 0 & Q_2 Q_2^T & 0 & \dots & 0 \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & 0 & \dots & Q_k Q_k^T \end{pmatrix}$$

and

$$Q_a Q_a^T = \begin{pmatrix} Q_1 Q_1^T & Q_1 Q_2^T & Q_1 Q_3^T & \dots & Q_1 Q_k^T \\ Q_2 Q_1^T & Q_2 Q_2^T & Q_2 Q_3^T & \dots & Q_2 Q_k^T \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ Q_k Q_1^T & Q_k Q_2^T & Q_k Q_3^T & \dots & Q_k Q_k^T \end{pmatrix}, \quad (42)$$

where $m = \sum_{i=1}^k m_i$.

- 2) The rows of matrix Q_i are vectors of norm \sqrt{N} . Therefore, the matrix $Q_i Q_i^T$, for $i = 1, 2, \dots, k$, is of size $m_i \times m_i$ and with diagonal entries equal to N regardless of the value of t_{ij} . Therefore, $(Q_a Q_a^T)_{\text{Diag}} = (Q_b Q_b^T)_{\text{Diag}} = N I_{m \times m}$.
- 3) When $t_{ij} \in \{0, T/N, \dots, (N-1)T/N\}$ and they are distinct, the rows of matrix Q_i will be perpendicular to each other. Therefore, the matrix $Q_i Q_i^T$ will be equal to $N I_{m_i \times m_i}$. Similarly, if t_{ij} are distinct for all i, j , the rows of Q_i and Q_j for $i \neq j$ will be perpendicular to each other. Therefore, $Q_i Q_j^T = 0$ for $i \neq j$ in this case. Hence, using the definition of Q_a and Q_b given in (25) and (26), both $Q_a Q_a^T$ and $Q_b Q_b^T$ will become diagonal matrices $N I_{m \times m}$, where $m = \sum_{i=1}^k m_i$.
- 4) The diagonal elements of matrices $Q_i^T Q_i$ for $i = 1, 2, \dots, k$ give us the norm of the column vectors

of Q_i . They can be calculated as follows:

$$Q_i^T Q_i(l, l) = \begin{cases} \sum_{j=1}^{m_i} \cos^2((N_1 + l - 1)\omega t_{ij}); & \text{for } 1 \leq l \leq N \\ \sum_{j=1}^{m_i} \sin^2((N_1 + l - N - 1)\omega t_{ij}); & \text{for } N + 1 \leq l \leq .2N \end{cases} \quad (43)$$

When we use the uniform sampling strategy, i.e., $\{t_{ij}\} = \{0, T/m_i, 2T/m_i, \dots, (m_i - 1)T/m_i\}$, the diagonal entries of $Q_i^T Q_i$ will become equal to $m_i/2$. This is because, for instance,

$$\begin{aligned} & \sum_{j=1}^{m_i} \cos^2((N_1 + l - 1)\omega t_{ij}) \\ &= \sum_{j=0}^{m_i-1} \cos^2\left((N_1 + l - 1)\omega \frac{T}{m_i} j\right) \\ &= \sum_{j=0}^{m_i-1} \left(\frac{1}{2} + \frac{1}{2} \cos\left(2(N_1 + l - 1)\frac{2\pi}{m_i} j\right)\right) \\ &= \frac{m_i}{2}, \end{aligned} \quad (44)$$

where (44) follows from the fact that $m_i > 2N_2$. Moreover, the off-diagonal entries will be zero. For example, consider the entry

$$\begin{aligned} & Q_i^T Q_i(2, 3) \\ &= \sum_{j=1}^{m_i} \cos((N_1 + 1)\omega t_{ij}) \cos((N_1 + 2)\omega t_{ij}) \\ &= \sum_{j=0}^{m_i-1} \frac{1}{2} \left[\cos\left(\frac{\omega T}{m_i} j\right) + \cos\left((2N_1 + 3)\frac{\omega T}{m_i} j\right) \right] \\ &= \sum_{j=0}^{m_i-1} \frac{1}{2} \left[\cos\left(\frac{2\pi}{m_i} j\right) + \cos\left((2N_1 + 3)\frac{2\pi}{m_i} j\right) \right] \\ &= 0. \end{aligned} \quad (45)$$

In fact, with uniform sampling, different columns of the matrix Q_i will be perpendicular to each other and $Q_i^T Q_i$ will become $(m_i/2)I_{2N \times 2N}$.

IV. PROOFS

In this section, we state the proofs of our results. In the body of the proofs, we have used some lemmas, which are provided in the appendix.

A. Proof of Equation (9)

We start by computing the average distortion using equations (33), (34) and (35) as follows

$$\begin{aligned} 2D_a &= \text{Tr}(C_e^a) \\ &= \text{Tr}\left(C_X - C_X Q_a^T (Q_a C_X Q_a^T + C_{\bar{Z}})^{-1} Q_a C_X\right) \\ &= \text{Tr}(C_X) - \text{Tr}\left((Q_a C_X Q_a^T + C_{\bar{Z}})^{-1} Q_a C_X^2 Q_a^T\right) \quad (46) \\ &= 2N - \text{Tr}\left((Q_a Q_a^T + C_{\bar{Z}})^{-1} Q_a Q_a^T\right), \quad (47) \end{aligned}$$

where (46) results from the cyclic property of the trace and (47) follows from the fact that $C_X = I_{2N \times 2N}$.

We would like to show that for any arbitrary choice of sampling time instances, t_{ij} , the average distortion will be bounded from below as follows:

$$\begin{aligned} 2D_a &= 2N - \text{Tr}\left[(Q_a Q_a^T + C_{\bar{Z}})^{-1} Q_a Q_a^T\right] \\ &\geq 2N - N \sum_{i=1}^k \frac{m_i}{(1+\eta)N + \sigma_i^2}. \quad (48) \end{aligned}$$

In other words, from (47), we wish to prove that

$$\text{Tr}\left((Q_a Q_a^T + C_{\bar{Z}})^{-1} Q_a Q_a^T\right) \leq \sum_{i=1}^k \frac{Nm_i}{(1+\eta)N + \sigma_i^2}. \quad (49)$$

Equivalently, if we use (37) to replace $C_{\bar{Z}}$ with $(\eta Q_b Q_b^T + C_Z)$, we would like to show that

$$\text{Tr}\left((Q_a Q_a^T + \eta Q_b Q_b^T + C_Z)^{-1} Q_a Q_a^T\right) \leq \sum_{i=1}^k \frac{Nm_i}{(1+\eta)N + \sigma_i^2}. \quad (50)$$

This can be derived using Theorem 3 (given in Appendix B) with matrices $F = Q_a Q_a^T + \eta Q_b Q_b^T$, $G = Q_a Q_a^T$ and $C = C_Z$. From Fact 1 of Section III-B, observe that the matrices F and G are of the forms

$$F = \begin{pmatrix} (1+\eta)Q_1 Q_1^T & Q_1 Q_2^T & \cdots & Q_1 Q_k^T \\ Q_2 Q_1^T & (1+\eta)Q_2 Q_2^T & \cdots & Q_2 Q_k^T \\ \vdots & \vdots & \ddots & \vdots \\ Q_k Q_1^T & Q_k Q_2^T & \cdots & (1+\eta)Q_k Q_k^T \end{pmatrix}_{m \times m} \quad (51)$$

and

$$G = Q_a Q_a^T = \begin{pmatrix} Q_1 Q_1^T & Q_1 Q_2^T & Q_1 Q_3^T & \cdots & Q_1 Q_k^T \\ Q_2 Q_1^T & Q_2 Q_2^T & Q_2 Q_3^T & \cdots & Q_2 Q_k^T \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ Q_k Q_1^T & Q_k Q_2^T & Q_k Q_3^T & \cdots & Q_k Q_k^T \end{pmatrix}_{m \times m} \quad (52)$$

These matrices satisfy the required properties of Theorem 3, *i.e.*, the matrices F and G are positive semi-definite and $G = F \circ L$, where the matrix L has the form of (110) with parameters $a = 1 + \eta$ and $b = 1$. Therefore, we have the following inequality

$$\text{Tr}\left[(F + C)^{-1} G\right] \leq \text{Tr}\left[(F_{\text{Diag}} + C)^{-1} G_{\text{Diag}}\right]. \quad (53)$$

Hence, from Fact 2 of Section III-B which states that $F_{\text{Diag}} = (Q_a Q_a^T)_{\text{Diag}} + \eta (Q_b Q_b^T)_{\text{Diag}} = N(1 + \eta)I_{m \times m}$ and $G_{\text{Diag}} = (Q_a Q_a^T)_{\text{Diag}} = NI_{m \times m}$, the desired inequality in (50) concludes.

Furthermore, when $\sum_{i=1}^k m_i \leq N$, we would like to show that this lower bound is tight if we take distinct time instances, t_{ij} , from the set $\{0, T/N, \dots, (N-1)T/N\}$. Observe that this is possible since $\{0, T/N, \dots, (N-1)T/N\}$ has N elements. Fact 3 from Section III-B states that both the matrices $Q_a Q_a^T$ and $Q_b Q_b^T$ will become diagonal matrices $NI_{m \times m}$, and thus $C_{\bar{Z}}$ given in (37) will be

$$C_{\bar{Z}} = \eta Q_b Q_b^T + C_Z = \eta NI_{m \times m} + C_Z.$$

Hence, the minimum distortion will be

$$\begin{aligned} 2D_{a \min} &= 2N - N \cdot \text{Tr}\left[\left((1+\eta)NI + C_Z\right)^{-1}\right], \\ &= 2N - \sum_{i=1}^k \frac{Nm_i}{(1+\eta)N + \sigma_i^2}, \quad (54) \end{aligned}$$

where (54) is derived using the definition of the diagonal matrix C_Z given in (28). \square

B. Proof of Equation (10):

To compute the minimum average distortion, from $D_a = 1/2 \text{Tr}(C_e^a)$, we use the alternative form of LMMSE (given in (36)), in which C_e^a is of the form

$$C_e^a = (Q_a^T C_{\bar{Z}}^{-1} Q_a + C_X^{-1})^{-1}. \quad (56)$$

Hence, the average distortion will be

$$\begin{aligned} 2D_a &= \text{Tr}(C_e^a) \\ &= \text{Tr}\left[\left(\sum_{i=1}^k Q_i^T (\eta Q_i Q_i^T + \sigma_i^2 I)^{-1} Q_i + I\right)^{-1}\right], \quad (57) \end{aligned}$$

which results from the facts that $C_{\bar{Z}} = \eta Q_b Q_b^T + C_Z$ (Section III-A.1) and $C_X = I_{2N \times 2N}$.

Using the matrix identity given in (31) for matrix A to be $\sqrt{\eta} Q_i$, we have

$$Q_i^T (\eta Q_i Q_i^T + C_{Z_i})^{-1} Q_i = \left(\eta Q_i^T C_{Z_i}^{-1} Q_i + I\right)^{-1} Q_i^T C_{Z_i}^{-1} Q_i.$$

$$A(l, l) = \begin{cases} \sum_{i=1}^k \left(1 + \sum_{j=1}^{m_i} \frac{\eta}{\sigma_i^2} \cos^2((N_1 + l - 1)\omega t_{ij}) \right) \triangleq a_\ell; & \text{for } 1 \leq l \leq N \\ \sum_{i=1}^k \left(1 + \sum_{j=1}^{m_i} \frac{\eta}{\sigma_i^2} \sin^2((N_1 + l - N - 1)\omega t_{ij}) \right) \triangleq b_\ell; & \text{for } N + 1 \leq l \leq 2N, \end{cases} \quad (55)$$

Therefore, the matrix in the left-hand side of (57) will be

$$\begin{aligned} & \sum_{i=1}^k Q_i^T \left(\eta Q_i Q_i^T + C_{Z_i} \right)^{-1} Q_i \\ &= \sum_{i=1}^k \left(\eta Q_i^T C_{Z_i}^{-1} Q_i + I \right)^{-1} Q_i^T C_{Z_i}^{-1} Q_i \\ &= \frac{1}{\eta} \sum_{i=1}^k \left(\eta Q_i^T Q_i + \sigma_i^2 I \right)^{-1} \left(\eta Q_i^T Q_i + \sigma_i^2 I - \sigma_i^2 I \right) \quad (58) \\ &= \frac{1}{\eta} \sum_{i=1}^k \left(I - \left(\eta \frac{Q_i^T Q_i}{\sigma_i^2} + I \right)^{-1} \right) \\ &= \frac{1}{\eta} \sum_{i=1}^k (I - B_i) \quad (59) \\ &= \frac{1}{\eta} \left(kI - \sum_{i=1}^k B_i \right) \\ &\leq \frac{1}{\eta} \left(kI - k^2 \left(\sum_{i=1}^k B_i^{-1} \right)^{-1} \right), \quad (60) \end{aligned}$$

where (58) results from the fact that $C_{Z_i} = \sigma_i^2 I$, and in (59) matrix B_i stands for $(\eta Q_i^T Q_i / \sigma_i^2 + I)^{-1}$. Moreover, (60) is derived using Lemma 4 for positive definite matrices B_i ($2N \times 2N$ matrices) with the equality if and only if $B_1 = B_2 = \dots = B_k$. Notice that $0 < B_i \leq I$.

For any two symmetric positive definite matrices A and B , the relation $A \leq B$ implies that $B^{-1} \leq A^{-1}$. This is because the function $f(t) = -t^{-1}$ is operator monotone [23]. Hence, (60) implies that

$$\begin{aligned} & \left(I + \sum_{i=1}^k Q_i^T (\eta Q_i Q_i^T + \sigma_i^2 I)^{-1} Q_i \right)^{-1} \\ & \geq \left[I + \frac{1}{\eta} \left(kI - k^2 \left(\sum_{i=1}^k B_i^{-1} \right)^{-1} \right) \right]^{-1}. \quad (61) \end{aligned}$$

Let $A = \sum_{i=1}^k B_i^{-1} = \sum_{i=1}^k (\eta Q_i^T Q_i / \sigma_i^2 + I)$. Then, the relation between the traces of the above matrices is

$$\begin{aligned} & \text{Tr} \left(I + \sum_{i=1}^k Q_i^T (\eta Q_i Q_i^T + \sigma_i^2 I)^{-1} Q_i \right)^{-1} \\ & \geq \text{Tr} \left[I + \frac{1}{\eta} \left(kI - k^2 A^{-1} \right) \right]^{-1} \quad (62) \\ & \geq \text{Tr} \left[I + \frac{1}{\eta} \left(kI - k^2 A_{\text{Diag}}^{-1} \right) \right]^{-1}, \quad (63) \end{aligned}$$

where (63) results from Lemma 3 for the convex function $f(t) = (1 + k/\eta - k^2 t^{-1}/\eta)^{-1}$ when $t \geq k$ and the Hermitian matrix $A \geq kI$. To find A_{Diag} , we need to calculate the diagonal entries of matrix A . They are given in (55) on top of this page.

Since the diagonal elements of matrices $Q_i^T Q_i$ for $i = 1, 2, \dots, k$ are of the following forms (Fact 4 from Section III-B)

$$Q_i^T Q_i(l, l) = \begin{cases} \sum_{j=1}^{m_i} \cos^2((N_1 + l - 1)\omega t_{ij}); & \text{for } 1 \leq l \leq N \\ \sum_{j=1}^{m_i} \sin^2((N_1 + l - N - 1)\omega t_{ij}); & \text{for } N + 1 \leq l \leq 2N. \end{cases} \quad (64)$$

Substituting the diagonal entries of the matrix A in (63), we obtain

$$\begin{aligned} & \text{Tr} \left(I + \sum_{i=1}^k Q_i^T (\eta Q_i Q_i^T + \sigma_i^2 I)^{-1} Q_i \right)^{-1} \\ & \geq \text{Tr} \left[I + \frac{1}{\eta} \left(kI - k^2 A_{\text{Diag}}^{-1} \right) \right]^{-1} \\ & = \sum_{\ell=1}^N \frac{1}{\left(1 + \frac{k}{\eta} \right) - \frac{k^2}{\eta} a_\ell^{-1}} + \frac{1}{\left(1 + \frac{k}{\eta} \right) - \frac{k^2}{\eta} b_\ell^{-1}} \\ & \geq \sum_{\ell=1}^N \frac{2}{\left(1 + \frac{k}{\eta} \right) - \frac{k^2}{\eta} \left(k + \eta \sum_{i=1}^k \frac{m_i}{2\sigma_i^2} \right)^{-1}} \quad (65) \\ & = \frac{2N}{\left(1 + \frac{k}{\eta} \right) - \frac{k^2}{\eta} \left(k + \eta \sum_{i=1}^k \frac{m_i}{2\sigma_i^2} \right)^{-1}}, \quad (66) \end{aligned}$$

where (65) results from convexity of the function $f(t) = (1 - at^{-1})^{-1}$ for $t \geq a$.

Now suppose that each $m_i > 2N_2$ for $i = 1, 2, \dots, k$. If we uniformly sample the signals, *i.e.*, sample $S_i(t)$ at time instances $\{0, T/m_i, 2T/m_i, \dots, (m_i - 1)T/m_i\}$, from Fact 4 of Section III-B we conclude that the equality in the above equations holds, and thus the this lower bound is achieved. \square

C. Proof of Equation (12):

To compute the average distortion, here we use equations (38), (39) and (41). Hence,

$$\begin{aligned} 2D_b &= \text{Tr} \left(C_X - C_X Q_b^T (Q_b C_X Q_b^T + C_Z)^{-1} Q_b C_X \right) \\ &= \text{Tr}(C_X) - \text{Tr} \left((Q_b C_X Q_b^T + C_Z)^{-1} Q_b C_X^2 Q_b^T \right) \quad (67) \end{aligned}$$

$$= k(2N)(1 + \eta) - \text{Tr} \left((Q_b C_X Q_b^T + C_Z)^{-1} Q_b C_X^2 Q_b^T \right), \quad (68)$$

where (67) and (68) are achieved, respectively, by the trace cyclic property and the fact that $C_{\mathbf{X}} = \Lambda \otimes I_{2N \times 2N}$ (The matrix Λ has been defined in (18)).

Here we are interested in reconstructing the signals $S_i(t)$. Following the similar steps from Subsection IV-A, we use Theorem 3 with the choice of $F = \mathbf{Q}_b C_{\mathbf{X}} \mathbf{Q}_b^T$, $G = \mathbf{Q}_b C_{\mathbf{X}}^2 \mathbf{Q}_b^T$ and $C = C_{\mathbf{Z}}$. To demonstrate that these matrices have the required properties of Theorem 3, one can verify (by explicit evaluation) that F has the same expression as in (51), *i.e.*, $F = \mathbf{Q}_b C_{\mathbf{X}} \mathbf{Q}_b^T = \mathbf{Q}_b (\Lambda \otimes I_{2N \times 2N}) \mathbf{Q}_b^T$ is also equal to $\mathbf{Q}_a \mathbf{Q}_a^T + \eta \mathbf{Q}_b \mathbf{Q}_b^T$. Furthermore to compute G , observe that $C_{\mathbf{X}}^2 = \Lambda^2 \otimes I_{2N \times 2N}$ in which Λ^2 is a matrix of the following form

$$\Lambda^2(i, j) = \begin{cases} (1 + \eta)^2 + (k - 1) \triangleq \alpha; & i = j, \\ 2(1 + \eta) + (k - 2) \triangleq \beta; & i \neq j, \end{cases} \quad (69)$$

for $i, j = 1, 2, \dots, k$. Then, one can verify that

$$G = \begin{pmatrix} \alpha Q_1 Q_1^T & \beta Q_1 Q_2^T & \beta Q_1 Q_3^T & \cdots & \beta Q_1 Q_k^T \\ \beta Q_2 Q_1^T & \alpha Q_2 Q_2^T & \beta Q_2 Q_3^T & \cdots & \beta Q_2 Q_k^T \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ \beta Q_k Q_1^T & \beta Q_k Q_2^T & \beta Q_k Q_3^T & \cdots & \alpha Q_k Q_k^T \end{pmatrix}_{m \times m} \quad (70)$$

Therefore, applying Theorem 3 for the matrices F , G and $C = C_{\mathbf{Z}}$, we have

$$\text{Tr} \left((\mathbf{Q}_b C_{\mathbf{X}} \mathbf{Q}_b^T + C_{\mathbf{Z}})^{-1} \mathbf{Q}_b C_{\mathbf{X}}^2 \mathbf{Q}_b^T \right) \quad (71)$$

$$\begin{aligned} &= \text{Tr}[(F + C_{\mathbf{Z}})^{-1} G] \\ &\leq \text{Tr}[(F_{\text{Diag}} + C_{\mathbf{Z}})^{-1} G_{\text{Diag}}] \\ &= \text{Tr} \left[((1 + \eta)NI + C_{\mathbf{Z}})^{-1} \alpha NI \right] \end{aligned} \quad (72)$$

$$= \sum_{i=1}^k \frac{\alpha N m_i}{(1 + \eta)N + \sigma_i^2}, \quad (73)$$

where (72) results from the fact that the diagonal entries of the matrices F and G are $(1 + \eta)N$ and αN , respectively (Fact 2 from Section III-B and (70)). Consequently, the average distortion, for any arbitrary choice of sampling times, can be bounded as

$$\begin{aligned} 2D_b &= 2Nk(1 + \eta) - \text{Tr} \left((\mathbf{Q}_b C_{\mathbf{X}} \mathbf{Q}_b^T + C_{\mathbf{Z}})^{-1} \mathbf{Q}_b C_{\mathbf{X}}^2 \mathbf{Q}_b^T \right) \\ &\geq 2Nk(1 + \eta) - \sum_{i=1}^k \frac{\alpha N m_i}{(1 + \eta)N + \sigma_i^2}, \end{aligned}$$

in which α is the one defined in (69).

Furthermore, we would like to show that the equality holds when $\sum_{i=1}^k m_i \leq N$ and we take distinct time instances, t_{ij} , from the set $\{0, T/N, \dots, (N - 1)T/N\}$. To show that we compute the average distortion when t_{ij} are distinct and belong to $\{0, T/N, \dots, (N - 1)T/N\}$ for all i, j . Using Fact 3 from Section III-B, the two matrices F and G will become diagonal $m \times m$ matrices of the forms:

$$F = \mathbf{Q}_b C_{\mathbf{X}} \mathbf{Q}_b^T = (1 + \eta)NI, \quad G = \mathbf{Q}_b C_{\mathbf{X}}^2 \mathbf{Q}_b^T = \alpha NI \quad (74)$$

Hence, the average distortion will be

$$2D_{b \min} = 2Nk(1 + \eta) - \text{Tr} \left(((1 + \eta)NI_{m \times m} + C_{\mathbf{Z}})^{-1} \alpha NI_{m \times m} \right) \quad (75)$$

$$= 2Nk(1 + \eta) - \sum_{i=1}^k \frac{\alpha N m_i}{(1 + \eta)N + \sigma_i^2} \quad (76)$$

where (76) is derived using the diagonal matrix $C_{\mathbf{Z}}$ of (28). \square

D. Proof of Equation (13):

Here, we divide the proof into two parts. In the first part, we show that

$$D_{b \min} \geq N \cdot \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1},$$

and then we show that when $m_i > 2N_2$ (for each $i \in 1, 2, \dots, k$), the optimal sampling strategy is uniform sampling and the minimum distortion is equal to

$$D_{b \min} = N \cdot \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1}.$$

In the second part, we simplify the above equation to obtain the expression given in the statement of the theorem.

Part (i): To compute the minimum average distortion, we use $D_b = 1/2 \text{Tr}(C_e^b)$ and the alternative form of LMMSE, where C_e^b is of the form

$$C_e^b = \left(\mathbf{Q}_b^T C_{\mathbf{Z}}^{-1} \mathbf{Q}_b + C_{\mathbf{X}}^{-1} \right)^{-1}. \quad (77)$$

In the above formula,

$$C_{\mathbf{Z}}^{-1} = \bigoplus_{i=1}^k \frac{1}{\sigma_i^2} I_{2N \times 2N}, \quad C_{\mathbf{X}}^{-1} = \Gamma_{k \times k} \otimes I_{2N \times 2N}, \quad (78)$$

where the matrix Γ is the inverse of the matrix Λ , given in (18). Therefore, the average distortion will be

$$2D_b = \text{Tr}(C_e^b) = \text{Tr} \left[\left(\bigoplus_{i=1}^k \frac{Q_i^T Q_i}{\sigma_i^2} \right) + \Gamma \otimes I \right]^{-1}. \quad (79)$$

First notice that due to Lemma 5, matrices $\Gamma \otimes I$ and $I \otimes \Gamma$ are permutation similar, *i.e.*, there exists a unique permutation matrix $P(k, 2N)$ of size $2Nk \times 2Nk$ such that

$$\Gamma \otimes I = P(k, 2N)^T (I \otimes \Gamma) P(k, 2N),$$

and moreover, $P(k, 2N)$ has the property

$$P(k, 2N) = P(k, 2N)^T = P(k, 2N)^{-1}.$$

$$H_{ll} = \begin{pmatrix} \frac{1}{\sigma_1^2} \sum_{j=1}^{m_1} \cos^2((N_1+l-1)\omega t_{1j}) & 0 & 0 \cdots & 0 \\ 0 & \frac{1}{\sigma_2^2} \sum_{j=1}^{m_2} \cos^2((N_1+l-1)\omega t_{2j}) & 0 \cdots & 0 \\ \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & 0 \cdots & \frac{1}{\sigma_k^2} \sum_{j=1}^{m_k} \cos^2((N_1+l-1)\omega t_{kj}) \end{pmatrix}_{k \times k} \quad (80)$$

for $l = 1, 2, \dots, N$.

$$H_{ll} = \begin{pmatrix} \frac{1}{\sigma_1^2} \sum_{j=1}^{m_1} \sin^2((N_1+l-N-1)\omega t_{1j}) & 0 & \cdots & 0 \\ 0 & \frac{1}{\sigma_2^2} \sum_{j=1}^{m_2} \sin^2((N_1+l-N-1)\omega t_{2j}) & \cdots & 0 \\ \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & \cdots & \frac{1}{\sigma_k^2} \sum_{j=1}^{m_k} \sin^2((N_1+l-N-1)\omega t_{kj}) \end{pmatrix}_{k \times k} \quad (81)$$

for $l = N + 1, \dots, 2N$.

Using the permutation matrix $P(k, 2N)$ and the cyclic property of the trace, we have

$$\begin{aligned} 2D_b &= \text{Tr}(C_e^b) \\ &= \text{Tr}\left(P(k, 2N)^T C_e^b P(k, 2N)\right) \\ &= \text{Tr}\left[P(k, 2N)^T \left(\bigoplus_{i=1}^k \frac{Q_i^T Q_i}{\sigma_i^2}\right) P(k, 2N)\right. \\ &\quad \left.+ P(k, 2N)^T (\Gamma \otimes I) P(k, 2N)\right]^{-1} \\ &= \text{Tr}(H + I \otimes \Gamma)^{-1}, \end{aligned} \quad (82)$$

where matrix H denotes the permuted matrix $P(k, 2N)^T \left(\bigoplus_{i=1}^k Q_i^T Q_i / \sigma_i^2\right) P(k, 2N)$. Notice that the matrix

$$\bigoplus_{i=1}^k Q_i^T Q_i = \sum_{i=1}^k G_i \otimes Q_i^T Q_i,$$

where G_i is a $k \times k$ matrix defined as follows: $G_i(i', j') = 0$ if $(i', j') \neq (i, i)$, and $G_i(i', j') = 1$ if $(i', j') = (i, i)$. Therefore, the matrix H can be written as

$$\begin{aligned} H &= P(k, 2N)^T \left(\bigoplus_{i=1}^k Q_i^T Q_i / \sigma_i^2\right) P(k, 2N) \\ &= \sum_{i=1}^k Q_i^T Q_i \otimes G_i. \end{aligned} \quad (83)$$

If we partition H into $k \times k$ submatrices H_{ij} as follows

$$H = \begin{pmatrix} H_{11} & H_{12} & \cdots & H_{1(2N)} \\ H_{21} & H_{22} & \cdots & H_{2(2N)} \\ \vdots & \vdots & \vdots & \vdots \\ H_{(2N)1} & H_{(2N)2} & \cdots & H_{(2N)(2N)} \end{pmatrix}, \quad (84)$$

all the H_{ij} submatrices will be diagonal matrices because they are weighted sums of diagonal matrices G_i . More precisely, the submatrices H_{ll} for $l \leq N$ can be computed as follows, using Fact 4 from Section III-B that gives us the diagonal entries of $Q_i^T Q_i$ (the entries of matrix H_{ll} for $1 \leq l \leq N$ and $N \leq l \leq 2N$ are the l th and the $(l + N)$ th diagonal entries of matrices $Q_i^T Q_i / \sigma_i^2$, given in (64)), respectively). They are given in equations (80) and (81) on top of this page.

Applying Lemma 3 for the Hermitian matrix $(H + I \otimes \Gamma)$ and the convex function $f(x) = x^{-1}$ for $x > 0$, we attain a lower bound on the average distortion as:

$$\begin{aligned} 2D_b &= \text{Tr}(C_e^b) = \text{Tr}(H + I \otimes \Gamma)^{-1}, \\ &\geq \text{Tr}(H_{\text{BDiag}} + I \otimes \Gamma)^{-1}, \end{aligned} \quad (85)$$

in which the matrix H_{BDiag} is the block diagonal form of matrix H , where all the submatrices other than the $2N$ block diagonal submatrices, H_{ll} , are zero. Consequently, the matrix $(H_{\text{BDiag}} + I \otimes \Gamma)$ is of the form

$$\begin{aligned} &(H_{\text{BDiag}} + I \otimes \Gamma) \\ &= \begin{pmatrix} H_{11} + \Gamma & 0 & \cdots & 0 \\ 0 & H_{22} + \Gamma & \cdots & 0 \\ \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & \cdots & H_{(2N)(2N)} + \Gamma \end{pmatrix}_{2Nk \times 2Nk}. \end{aligned}$$

Therefore,

$$\text{Tr}(H_{\text{BDiag}} + I \otimes \Gamma)^{-1} = \sum_{l=1}^{2N} \text{Tr}(H_{ll} + \Gamma)^{-1}.$$

Using Lemma 6 (given in Appendix B), for any $1 \leq l \leq N$, we obtain

$$\begin{aligned} & \frac{1}{2} (\text{Tr}(H_{ll} + \Gamma)^{-1} + \text{Tr}(H_{(l+N)(l+N)} + \Gamma)^{-1}) \\ & \geq \text{Tr} \left(\frac{1}{2} (H_{ll} + H_{(l+N)(l+N)}) + \Gamma \right)^{-1} \end{aligned} \quad (86)$$

$$= \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1}. \quad (87)$$

Consequently,

$$\begin{aligned} \text{Tr}(H_{\text{BDiag}} + I \otimes \Gamma)^{-1} &= \sum_{l=1}^{2N} \text{Tr}(H_{ll} + \Gamma)^{-1} \\ &\geq 2N \cdot \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1}. \end{aligned}$$

Therefore, from (85) for any choice of sampling time instances t_{ij} , we obtain

$$D_{\text{b}} \geq N \cdot \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1}. \quad (88)$$

Moreover, suppose that each $m_i > 2N_2$ for $i = 1, 2, \dots, k$. If we employ uniform strategy, the matrices $Q_i^T Q_i$ ($i = 1, 2, \dots, k$) become diagonal matrices with diagonal entries equal to $m_i/2$ (Fact 4 from Section III-B). Then, the matrix in the right hand side of (79) will be

$$\begin{aligned} & \left[\left(\bigoplus_{i=1}^k \frac{m_i}{2\sigma_i^2} I_{2N \times 2N} \right) + \Gamma \otimes I \right]^{-1} \\ &= \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) \otimes I + \Gamma \otimes I \right)^{-1} \end{aligned} \quad (89)$$

$$= \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1} \otimes I_{2N \times 2N}. \quad (90)$$

Therefore, from (79) and the fact that $\text{Tr}(A \otimes B) = \text{Tr}(A)\text{Tr}(B)$, we get

$$D_{\text{bmin}} \leq N \cdot \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1}.$$

Therefore, from (88) and the above inequality, we conclude that

$$D_{\text{bmin}} = N \cdot \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1}.$$

Part (ii): So far we have shown that when $m_i > 2N_2$ for $i = 1, 2, \dots, k$, the minimal distortion is

$$D_{\text{bmin}} = N \cdot \text{Tr} \left(\text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \right)^{-1}.$$

Here, we wish to simplify the above equation to obtain the final distortion bound given in (13). Observe that $\Gamma = \Lambda^{-1}$

can be computed from (18) as follows:

$$\Gamma = \begin{pmatrix} \mathbf{a} & \mathbf{b} & \mathbf{b} & \dots & \mathbf{b} \\ \mathbf{b} & \mathbf{a} & \mathbf{b} & \dots & \mathbf{b} \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ \mathbf{b} & \mathbf{b} & \mathbf{b} & \dots & \mathbf{a} \end{pmatrix}_{k \times k}, \quad (91)$$

in which

$$\mathbf{a} = \frac{\eta + k - 1}{\eta(\eta + k)} \quad \text{and} \quad \mathbf{b} = \frac{-1}{\eta(\eta + k)}. \quad (92)$$

For simplicity define:

$$A := \text{Diag} \left(\left[\frac{m_1}{2\sigma_1^2}, \frac{m_2}{2\sigma_2^2}, \dots, \frac{m_k}{2\sigma_k^2} \right] \right) + \Gamma \quad (93)$$

Therefore, we need to find $\text{Tr}(A^{-1})$ which is equal to $\sum_{i=1}^k (A^{-1})_{i,i}$. To calculate the diagonal entries of A^{-1} , we use the Cramer's rule as follows:

$$(A^{-1})_{i,i} = \frac{\det(A_i)}{\det(A)} \quad (94)$$

where A_i is the remaining matrix after removing the i th row and column of A . According to Lemma 7 (given in Appendix B), we deduce:

$$(A^{-1})_{i,i} = \frac{\prod_{\substack{j=1 \\ j \neq i}}^k \phi_j + \mathbf{b} \sum_{j=1}^k \prod_{\substack{l=1 \\ l \neq i, j}}^k \phi_l}{\prod_{j=1}^k \phi_j + \mathbf{b} \sum_{j=1}^k \prod_{\substack{l=1 \\ l \neq j}}^k \phi_l} \quad (95)$$

in which $\phi_i = \frac{m_i}{2\sigma_i^2} + \mathbf{a} - \mathbf{b} = \frac{m_i}{2\sigma_i^2} + \frac{1}{\eta}$. Hence,

$$\text{Tr}(A^{-1}) = \sum_{i=1}^k (A^{-1})_{i,i} \quad (96)$$

$$\begin{aligned} &= \frac{\sum_{i=1}^k \frac{1}{\phi_i} + \mathbf{b} \sum_{\substack{i, j=1 \\ j \neq i}}^k \frac{1}{\phi_i \phi_j}}{1 + \mathbf{b} \left(\sum_{i=1}^k \frac{1}{\phi_i} \right)}, \end{aligned} \quad (97)$$

in which (97) is derived by replacing (95) in (96) and dividing both numerator and denominator by $\prod_{i=1}^k \phi_i$. Observe that by

our definition $\Phi_{-1} := \sum_{i=1}^k \frac{1}{\phi_i}$, we get

$$\begin{aligned} \text{Tr}(A^{-1}) &= \frac{\Phi_{-1} + \mathbf{b}(\Phi_{-1}^2 - \sum_{i=1}^k \frac{1}{\phi_i^2})}{1 + \mathbf{b}\Phi_{-1}} \\ &= \Phi_{-1} - \frac{\mathbf{b}}{1 + \mathbf{b}\Phi_{-1}} \Phi_{-2}. \end{aligned} \quad (98)$$

We get the desired result by replacing $\mathbf{b} = -1/(\eta(\eta + k))$. \square

APPENDIX A
LEMMAS

In this section, we state some lemmas that have been used in the proof section.

A. Majorization Inequalities

A vector $\mathbf{x} \in \mathbb{R}^n$ is majorized by $\mathbf{y} \in \mathbb{R}^n$ if after sorting the two vectors in decreasing order, the following inequalities hold:

$$\sum_{i=1}^k x_i \leq \sum_{i=1}^k y_i \quad (1 \leq k \leq n), \quad \sum_{i=1}^n x_i = \sum_{i=1}^n y_i. \quad (99)$$

A fundamental result in majorization theory states that for any Hermitian matrix A of size $n \times n$, the diagonal entries of A are majorized by its eigenvalues [20]. The extension of the above result to the block Hermitian matrices is also true (e.g. see [24, Sec. 1]):

Lemma 2 (Block Majorization Inequality): If a Hermitian matrix A is partitioned into block matrices

$$A = \begin{pmatrix} M_{11} & M_{12} & \dots & M_{1k} \\ M_{21} & M_{22} & \dots & M_{2k} \\ \vdots & & \ddots & \\ M_{k1} & M_{k2} & \dots & M_{kk} \end{pmatrix} \quad (100)$$

for matrices M_{ij} , $i, j = 1, 2, \dots, k$, then the eigenvalues of $\bigoplus_{i=1}^k M_{ii}$ are majorized by the eigenvalues of A .

If a vector \mathbf{x} is majorized by \mathbf{y} , then for any convex functions $f: \mathbb{R} \mapsto \mathbb{R}$, we have $\sum_i f(x_i) \leq \sum_i f(y_i)$ [20]. This implies that

Lemma 3 Let Ω be a closed interval in \mathbb{R} . For any Hermitian matrix A with eigenvalues in Ω , and any convex function f on Ω ,

$$\text{Tr}(f(A)) \geq \text{Tr}(f(A_{\text{Diag}})). \quad (101)$$

More generally by Lemma 2, for a Hermitian matrix A partitioned into block matrices M_{ij} , $i, j = 1, 2, \dots, k$, as in (100),

$$\text{Tr}(f(A)) \geq \text{Tr}\left(f\left(\bigoplus_{i=1}^k M_{ii}\right)\right). \quad (102)$$

B. Other Useful Definitions and Inequalities

Lemma 4 [25]: Let w_1, w_2, \dots, w_k be non-negative weights adding up to one, and let B_1, B_2, \dots, B_k be $n \times n$ positive definite matrices. Consider the weighted arithmetic and harmonic means of the matrices B_i

$$A \triangleq w_1 B_1 + w_2 B_2 + \dots + w_k B_k, \quad (103)$$

$$H \triangleq (w_1 B_1^{-1} + w_2 B_2^{-1} + \dots + w_k B_k^{-1})^{-1}. \quad (104)$$

Then, the following inequality holds,

$$H \leq A,$$

with equality if and only if $B_1 = B_2 = \dots = B_k$.

Definition 1 [26]: A real-valued continuous function $f(t)$ on a real interval I is called operator monotone if

$$A \leq B \Rightarrow f(A) \leq f(B), \quad (105)$$

for Hermitian matrices A and B with eigenvalues in I . Furthermore, f is called operator convex if

$$f(\lambda A + (1 - \lambda)B) \leq \lambda f(A) + (1 - \lambda)f(B),$$

for any $0 \leq \lambda \leq 1$ and Hermitian matrices A and B with eigenvalues that are contained in I , and f is said to be operator concave if $-f$ is operator convex.

Lemma 5 [27, p. 260]: Let m and n be given positive integers and matrices A and B be any square matrices of sizes $m \times m$ and $n \times n$, respectively. Then, matrix $B \otimes A$ is permutation similar to matrix $A \otimes B$, i.e., there is a unique matrix P such that

$$B \otimes A = P(m, n)^T (A \otimes B) P(m, n) \quad (106)$$

where $P(m, n)$ is the following $mn \times mn$ permutation matrix:

$$P(m, n) = \sum_{i=1}^n \sum_{j=1}^m E_{ij} \otimes E_{ij}^T \quad (107)$$

in which E_{ij} is an $m \times n$ matrix such that only the (i, j) th entry is unity and the other entries are zero. Furthermore, the useful following property holds

$$P(m, n) = P(n, m)^T = P(n, m)^{-1}. \quad (108)$$

Lemma 6: Assume that B_1, B_2 are positive definite matrices. Let $B = (B_1 + B_2)/2$, then

$$2\text{Tr}(B^{-1}) \leq \text{Tr}(B_1^{-1}) + \text{Tr}(B_2^{-1}).$$

Proof: The Löwner-Heinz theorem implies that the function $f(t) = t^{-1}$ for $t > 0$ is operator convex [23]. From the fact that $B = (B_1 + B_2)/2$, we conclude the desired inequality. ■

Lemma 7: Given real non-negative a_1, a_2, \dots, a_k and positive b , let

$$M(a_1, \dots, a_k, b) := \begin{pmatrix} a_1 & -b & -b & \dots & -b \\ -b & a_2 & -b & \dots & -b \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ -b & -b & -b & \dots & a_k \end{pmatrix}_{k \times k}. \quad (109)$$

Then,

$$\det(M(a_1, \dots, a_k, b)) = \prod_{i=1}^k (a_i + b) - b \sum_{i=1}^k \prod_{\substack{j=1 \\ j \neq i}}^k (a_j + b).$$

Proof: The elementary row operations do not change the determinant. If we first subtract the first row from all the other rows, and then multiply the i th row of the matrix by $b/(a_i + b)$ for $i = 2, 3, \dots, k$, and add it to the first row, we end up with an lower triangular matrix with diagonal entries $\{a', a_2 + b, a_3 + b, \dots, a_k + b\}$, where $a' = a_1 - \sum_{i=2}^k \frac{(a_1 + b)b}{(a_i + b)}$.

$$A = \begin{pmatrix} F_{1:m_1 \times 1:m_1} & 0 & 0 \cdots & 0 \\ 0 & F_{(m_1+1:m_1+m_2) \times (m_1+1:m_1+m_2)} & 0 \cdots & 0 \\ \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & 0 \cdots & F_{(m_1+\cdots+m_{k-1}+1) \times (m_1+\cdots+m_{k-1}+m_k)} \end{pmatrix}_{m \times m}. \quad (112)$$

Since the determinant of an upper triangular matrix is equal to product of the diagonal elements, we have

$$\begin{aligned} \det(M(a_1, \dots, a_k, b)) &= a'(a_2 + b) \dots (a_k + b) \\ &= a_1(a_2 + b) \dots (a_k + b) - b \sum_{i=2}^k \prod_{\substack{j=1 \\ j \neq i}}^k (a_j + b) \\ &= \prod_{i=1}^k (a_i + b) - b \sum_{i=1}^k \prod_{\substack{j=1 \\ j \neq i}}^k (a_j + b). \end{aligned}$$

■

APPENDIX B

A NEW REVERSE MAJORIZATION INEQUALITY

Theorem 3: Take two positive semidefinite matrices F and G of sizes $m \times m$ satisfying $G = F \circ L$, where \circ is the Hadamard product and L is a matrix of the following form:

$$L = \begin{pmatrix} a\mathbf{1}_{m_1 \times m_1} & b\mathbf{1}_{m_1 \times m_2} & \cdots & b\mathbf{1}_{m_1 \times m_k} \\ b\mathbf{1}_{m_2 \times m_1} & a\mathbf{1}_{m_2 \times m_2} & \cdots & b\mathbf{1}_{m_2 \times m_k} \\ \vdots & \vdots & \vdots & \vdots \\ b\mathbf{1}_{m_k \times m_1} & b\mathbf{1}_{m_k \times m_2} & \cdots & a\mathbf{1}_{m_k \times m_k} \end{pmatrix}, \quad (110)$$

where $\mathbf{1}$ is a matrix with all one coordinates and a and b are two positive real numbers, where $0 \leq a \leq b$. Then, for any positive definite diagonal matrix C , we have

$$\text{Tr}[(F + C)^{-1}G] \leq \text{Tr}[(F_{\text{Diag}} + C)^{-1}G_{\text{Diag}}], \quad (111)$$

where F_{Diag} is a diagonal matrix formed by taking the diagonal entries of F , and the matrix G_{Diag} is defined similarly.

Proof: If the statement of theorem holds for the matrix F , it will also hold for the matrix kF for any positive constant k . Therefore, without loss of generality, we assume that $a = 1$ and hence, $b \geq 1$. From the Hadamard product relation, this implies that F and G are equal on block matrices on the diagonal. Let A denote this common part as given in equation (112) on the top of the this page.

One can find matrix B such that $F = A + B$ and $G = A + bB$. Observe that wherever A is non-zero, B is zero and vice versa. Substituting F and G in the left hand side of (111), one attains

$$\begin{aligned} &\text{Tr}[(F + C)^{-1}G] \\ &= \text{Tr}[(A + B + C)^{-1}(A + bB)] \\ &= \text{Tr}[(A + B + C)^{-1}(A + B)] \\ &\quad + \text{Tr}[(A + B + C)^{-1}(b - 1)B]. \end{aligned} \quad (113)$$

We will show that the first term in the above formula is less than or equal to the right hand side of (111) and the second term is non-positive.

Start with the first term of (113)

$$\begin{aligned} &\text{Tr}[(A + B + C)^{-1}(A + B)] \\ &= \text{Tr}(I_{m \times m}) - \text{Tr}((A + B + C)^{-1}C), \end{aligned} \quad (114)$$

where the second term in (114) can be bounded as follows:

$$\begin{aligned} &\text{Tr}((A + B + C)^{-1}C) \\ &= \text{Tr}[(C^{-\frac{1}{2}}AC^{-\frac{1}{2}} + C^{-\frac{1}{2}}BC^{-\frac{1}{2}} + I)^{-1}] \\ &\geq \text{Tr}[(C^{-\frac{1}{2}}A_{\text{Diag}}C^{-\frac{1}{2}} + I)^{-1}]. \end{aligned} \quad (115)$$

The last inequality comes from Lemma 3.

Hence, (114) will be bounded as

$$\begin{aligned} &\text{Tr}[(A + B + C)^{-1}(A + B)] \\ &\leq \text{Tr}(I_{m \times m}) - \text{Tr}[(C^{-\frac{1}{2}}A_{\text{Diag}}C^{-\frac{1}{2}} + I)^{-1}] \\ &= \text{Tr}[(A_{\text{Diag}} + C)^{-1}(A_{\text{Diag}} + C)] \\ &\quad - \text{Tr}[C^{\frac{1}{2}}(A_{\text{Diag}} + C)^{-1}C^{\frac{1}{2}}] \\ &= \text{Tr}[(A_{\text{Diag}} + C)^{-1}A_{\text{Diag}}] \\ &= \text{Tr}[(F_{\text{Diag}} + C)^{-1}G_{\text{Diag}}], \end{aligned} \quad (116) \quad (117) \quad (118) \quad (119)$$

in which (118) comes from the trace interchange property and (119) is derived since A is defined to be the common part of the two matrices F and G .

To complete the proof, it remains to show that the right hand side of (113) is non-positive, i.e.,

$$(b - 1) \cdot \text{Tr}[(A + B + C)^{-1}B] \leq 0. \quad (120)$$

Since we have assumed that $b \geq 1$, we need to show that the trace function is non-positive. For the positive definite matrix $(A + C)$, we have

$$\begin{aligned} &\text{Tr}((A + B + C)^{-1}B) \\ &= \text{Tr}(I) - \text{Tr}((A + C + B)^{-1}(A + C)) \\ &= \text{Tr}(I) \\ &\quad - \text{Tr}((A + C)^{\frac{1}{2}}(A + C + B)^{-1}(A + C)^{\frac{1}{2}}) \end{aligned} \quad (121)$$

$$\begin{aligned} &= \text{Tr}(I) - \text{Tr}((I + (A + C)^{-\frac{1}{2}}B(A + C)^{-\frac{1}{2}})^{-1}) \\ &\leq \text{Tr}(I) - \text{Tr}(I) \\ &= 0. \end{aligned} \quad (122)$$

in which the inequality (121) follows from the trace interchange property and (122) is derived using Lemma 3 for the Hermitian matrix $(I + (A + C)^{\frac{1}{2}}B(A + C)^{\frac{1}{2}})^{-1}$. Note that the matrix $(A + C)^{\frac{1}{2}}B(A + C)^{\frac{1}{2}}$ is a block off-diagonal matrix, since matrices A and B has been defined to be, respectively, block diagonal and off-diagonal matrices such that wherever A_{ij} is zero, B_{ij} is non-zero and vice versa. This completes the proof of Theorem 3. ■

ACKNOWLEDGEMENTS

The authors would like to thank Amin Gohari and Arash Amini for many helpful discussions.

REFERENCES

- [1] T. Berger, *Rate-Distortion Theory*. Hoboken, NJ, USA: Wiley, 1971.
- [2] T. Berger, Z. Zhang, and H. Viswanathan, "The CEO problem [multiterminal source coding]," *IEEE Trans. Inf. Theory*, vol. 42, no. 3, pp. 887–902, May 1996.
- [3] H. Viswanathan and T. Berger, "The quadratic Gaussian CEO problem," *IEEE Trans. Inf. Theory*, vol. 43, no. 5, pp. 1549–1559, Sep. 1997.
- [4] H. Witsenhausen, "Indirect rate distortion problems," *IEEE Trans. Inf. Theory*, vol. 26, no. 5, pp. 518–521, Sep. 1980.
- [5] J. Wolf and J. Ziv, "Transmission of noisy information to a noisy receiver with minimum distortion," *IEEE Trans. Inf. Theory*, vol. 16, no. 4, pp. 406–411, Apr. 1970.
- [6] R. Dobrushin and B. Tsybakov, "Information transmission with additional noise," *IRE Trans. Inf. Theory*, vol. 8, no. 5, pp. 293–304, Jun. 1962.
- [7] S. Feizi, V. K. Goyal, and M. Medard, "Time-stampless adaptive nonuniform sampling for stochastic signals," *IEEE Trans. Signal Process.*, vol. 60, no. 10, pp. 5440–5450, Oct. 2012.
- [8] V. P. Boda and P. Narayan, "Sampling rate distortion," in *Proc. IEEE Int. Symp. Inf. Theory (ISIT)*, Apr. 2014, pp. 3057–3061.
- [9] A. Kipnis, A. Goldsmith, Y. C. Eldar, and T. Weissman, "Distortion rate function of sub-Nyquist sampled Gaussian sources," *IEEE Trans. Inf. Theory*, vol. 62, no. 1, pp. 401–429, Jan. 2016.
- [10] E. Mohammadi and F. Marvasti, "Sampling and distortion tradeoffs for bandlimited periodic signals," in *Proc. Int. Conf. Sampling Theory Appl. (SampTA)*, 2015, pp. 468–472.
- [11] D. Baron, M. B. Wakin, M. F. Duarte, S. Sarvotham, and R. G. Baraniuk, "Distributed compressed sensing," Dept. Elect. Comput. Eng., Rice Univ., Houston, TX, USA, Tech. Rep. ECE-0612, Dec. 2006.
- [12] A. Hormati, O. Roy, Y. M. Lu, and M. Vetterli, "Distributed sampling of signals linked by sparse filtering: Theory and applications," *IEEE Trans. Signal Process.*, vol. 58, no. 3, pp. 1095–1109, Apr. 2010.
- [13] A. Ahmed and J. Romberg. (Jan. 2015). "Compressive sampling of ensembles of correlated signals." [Online]. Available: <https://arxiv.org/abs/1501.06654>
- [14] D. Ganesan, D. Estrin, and J. Heidemann, "DIMENSIONS: Why do we need a new data handling architecture for sensor networks?" *ACM SIGCOMM Comput. Commun. Rev.*, vol. 33, no. 1, pp. 143–148, 2003.
- [15] S. Bandyopadhyay, Q. Tian, and E. J. Coyle, "Spatio-temporal sampling rates and energy efficiency in wireless sensor networks," *IEEE/ACM Trans. Netw.*, vol. 13, no. 6, pp. 1339–1352, Dec. 2005.
- [16] D. Zordan, G. Quer, M. Zorzi, and M. Rossi, "Modeling and generation of space-time correlated signals for sensor network fields," in *Proc. IEEE Global Telecommun. Conf. (GLOBECOM)*, Dec. 2011, pp. 1–6.
- [17] R. Masiero, G. Quer, D. Munaretto, M. Rossi, J. Widmer, and M. Zorzi, "Data acquisition through joint compressive sensing and principal component analysis," in *Proc. IEEE Globecom*, Honolulu, Hawaii, Nov./Dec. 2009, pp. 1–6.
- [18] W. Bajwa, J. Haupt, A. Sayeed, and R. Nowak, "Compressive wireless sensing," in *Proc. Int. Conf. Inf. Process. Sens. Netw.*, Apr. 2006, pp. 134–142.
- [19] W. Wang, M. Garofalakis, and K. Ramchandran, "Distributed sparse random projections for refinable approximation," in *Proc. Int. Conf. Inf. Process. Sens. Netw.*, Apr. 2007, pp. 331–339.
- [20] R. Bhatia, *Matrix Analysis*. New York, NY, USA: Springer-Verlag, 1997.
- [21] S. M. Kay, *Fundamentals of Statistical Signal Processing: Estimation Theory*. Upper Saddle River, NJ, USA: Prentice-Hall, 1993.
- [22] D. G. Luenberger, *Optimization by Vector Space Methods*. Hoboken, NJ, USA: Wiley, 1990.
- [23] E. Carlen, "Trace inequalities and quantum entropy: An introductory course," *Entropy Quantum*, vol. 529, pp. 73–140, Apr. 2010.
- [24] M. Lin and H. Wolkowicz, "An eigenvalue majorization inequality for positive semidefinite block matrices," *Linear Multilinear Algebra*, vol. 60, nos. 11–12, pp. 1365–1368, 2012.
- [25] M. Sagae and K. Tanabe, "Upper and lower bounds for the arithmetic-geometric-harmonic means of positive definite matrices," *Linear Multilinear Algebra*, vol. 37, no. 4, pp. 279–282, 1994.
- [26] X. Zhan, *Matrix Inequalities*. Berlin, Germany: Springer-Verlag, 2002.
- [27] R. A. Horn and C. R. Johnson, *Topics in Matrix Analysis*, vol. 2. Cambridge, U.K.: Cambridge Univ. Press, 1991.

Elaheh Mohammadi is currently a PhD student at Sharif University of Technology, Tehran, Iran. She obtained her Msc and Bsc degrees from Amirkabir and Tehran Universities in Electrical Engineering, respectively. Her research interests are sampling theory and signal processing.

Alireza Fallah is a final year Bachelor student studying both Electrical Engineering and Mathematics at Sharif University of Technology, Tehran, Iran. As a high school student, he participated in the International Mathematical Olympiad (IMO) for two years and won two silver medals in those competitions. Also, as an undergraduate student, he succeeded to achieve the first prize in International Mathematics Competition (IMC) in 2016. His current research interests are Optimization, Statistics, and Signal Processing.

Farokh Marvasti received his BS, Ms and PhD degrees all from Rensselaer Polytechnic Institute in 1970, 1971 and 1973, respectively. He has worked, consulted and taught in various industries and academic institutions since 1972. Among which are Bell Labs, University of California Davis, Illinois Institute of Technology, University of London, King's College. He was one of the editors and associate editors of IEEE TRANSACTIONS ON COMMUNICATIONS AND SIGNAL PROCESSING from 1990-1997. He has about 100 Journal publications and has written several reference books; he has also several international patents. His last book is on *Non-uniform Sampling: Theory and Practice* (Springer, 2001). He was also a guest editor on Special Issue on Non-uniform Sampling for the *Sampling Theory and Signal and Image Processing* Journal, May 2008. Besides being the co-founders of two international conferences (ICT's and SampTA's), he has been the organizer and special session chairs of many IEEE conferences including ICASSP conferences. Recently, he was the Lead editor on "Sparse Signal Processing" for the Special Issue of *Eurasip Journal on Advanced in Signal Processing*. Dr. Marvasti is currently a professor at Sharif University of Technology and the director of Advanced Communications Research Institute (ACRI) and a former head of Center for Multi-Access Communications Systems. He spent his sabbatical leave at the Communications and Information Systems Group of University College London (UCL) in 2013. Prof Marvasti received a distinguished award from the Iranian Academy of Sciences in 2014 and a 5 year term chair position from Iranian National Science foundation in 2015