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An Efficient Interference-Aware Constrained Massive MIMO Beamforming for mm-Wave JSDM

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ABSTRACT Low-complexity beamformer design with practical constraints is an attractive research area for hybrid analog/digital systems in mm-wave massive multiple-input multiple-output (MIMO). This paper investigates interference-aware pre-beamformer (analog beamformer) design for joint spatial division and multiplexing (JSDM) which is a user-grouping based two-stage beamforming method. Single-carrier frequency domain equalization (SC-FDE) is employed in uplink frequency-selective channels. First, unconstrained slowly changing statistical analog beamformer of each group, namely, generalized eigenbeamformer (GEB) which has strong interference suppression capability is designed by maximizing the mutual information in reduced dimension. Then, constant-modulus constrained approximations of unconstrained beamformer are obtained by utilizing alternating minimization algorithms for fully connected arrays and fixed subarrays. In addition, a dynamic subarray algorithm is proposed where the connections between radio frequency (RF) chains and antennas are changed with changing channel statistics. Convergence of the proposed alternating minimization-based algorithms is provided along with their complexity analysis. It is observed that the additional complexity of proposed algorithms is insignificant for the overall system design. Although most of the interference is suppressed with the help of proposed constrained beamformers, there may be some residual interference after analog beamforming stage. Thus, minimum mean square error (MMSE) criterion based iterative block decision feedback equalization (IB-DFE) method, which takes the residual interference in reduced dimension into account, is promoted for digital beamforming stage. Simulation results verify the superiority of the proposed interference-aware constrained design over existing approaches in terms of beam pattern, spectral efficiency, outage capacity, bit-error rate (BER), and channel estimation accuracy.

INDEX TERMS Hybrid beamforming, mm-wave systems, massive MIMO, constrained beamformer, dynamic subarray.

I. INTRODUCTION

Massive multiple-input multiple-output (MIMO) at mm-wave bands is the frontier technology for next generation wireless communication systems due to the availability of vast spectrum [1], [2]. Hybrid analog/digital structure is an efficient approach for these systems as it has lower hardware complexity and energy consumption compared to fully digital beamforming, and mm-wave bands exhibit joint angle-delay sparsity [3]. Hence, researchers have been working on low-complexity beamformer design for hybrid

systems [4]–[8]. Joint spatial division and multiplexing (JSDM) is a user-grouping based two-stage beamforming method where analog beamformers are designed by using slowly varying channel statistics, i.e., channel covariance matrices (CCMs) [9], [10]. User-grouping reduces signal processing complexity since digital processing of each group becomes independent from each other. In addition, only instantaneous intra-group channels in reduced dimension need to be learned which reduces channel estimation overhead. In this paper, we adopt JSDM framework in frequency-selective channels where we employ single-carrier frequency domain equalization (SC-FDE) for uplink transmission.

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A. RELATED WORK

Most of the existing researches for massive MIMO systems adopt flat-fading channel model by considering the use of orthogonal frequency division multiplexing (OFDM) [2]. The reason to prefer single-carrier (SC) modulation over OFDM in this paper is its superiority for systems having nonlinearities, such as quantized MIMO [11], [12], owing to its lower peak-to-average power ratio (PAPR) [13] and robustness to carrier-frequency-offset (CFO) errors [14]. Having lower PAPR is a critical property for systems with non-linear elements [15], where the advantages of SC transmission over the multicarrier are demonstrated. Moreover, even for unquantized linear systems, many recent studies motivate the use of SC especially for MIMO systems in mm-wave bands [16]–[19]. In these studies, the mitigation of inter-symbol interference (ISI) is fulfilled via reduced complexity signal processing.

Analog beamformer design for JSDM framework is a critical task since inter-group interference should be mitigated at this stage. There are several papers that find unconstrained analog beamformers that take interference into account [9], [10], [20]. However, they do not suppress inter-group interference perfectly. In [20], a weighted minimum mean square error (WMMSE) type digital beamformer is proposed to suppress residual interference. A near-optimal generalized eigenbeamformer (GEB), which has strong interference suppression capability, is proposed for SC transmission at mm-wave bands in [18], [19]. However, none of these analog beamformers obey constant-modulus constraint. For two-stage beamforming concept, DFT beamformer and phase of unconstrained beamformers are considered as state-of-the-art constrained analog beamformer alternatives [9], [21]–[24].

On the other hand, constrained analog beamformer design for hybrid analog/digital systems is a well-studied research area [5]–[7]. These papers use the instantaneous full dimensional channel state information (CSI) for beamformer design with fully connected arrays. In [23], two-stage beamforming approach is adopted where columns of the unconstrained analog beamformer are selected as eigenvectors of sum of CCMs of users. Furthermore, constrained analog beamformer is obtained via alternating minimization algorithm, and a compensation matrix is applied in digital baseband to orthogonalize the overall analog beamformer. On the other hand, compared to fully connected arrays, partially connected ones require much fewer phase shifters at the expense of spectral efficiency. Compromise between spectral efficiency and energy consumption for fully connected arrays and fixed subarrays is extensively studied in [6], [25].

Recently, a challenging problem called dynamic subarray design is studied where the aim is to obtain the optimal connection between radio frequency (RF) chains and antennas for partially connected arrays [24], [26]–[31]. In these papers, it is shown that dynamic subarrays outperform fixed ones in terms of several metrics such as spectral efficiency. [24],

[26]–[28] consider a point-to-point hybrid analog/digital system, whereas [29]–[31] consider multiple users. In [29], frequency-selective channels are considered in uplink transmission where ISI is mitigated with frequency domain iterative block decision feedback equalization (IB-DFE), which is similar to our work. However, dynamic connections in that paper depend on instantaneous CSI, which is not practical, whereas other papers on dynamic subarray design and our work make use of slowly changing parameters (e.g., CCMs) to find optimal connections. For dynamic subarrays, connection change can be realized with switches which are shown to be energy-efficient components for mm-wave massive MIMO systems [32], [33].

Table 1 summarizes the contributions of the existing work on constrained beamformer design. None of them are interested in constrained pre-beamformer (analog beamformer) design even for fully connected arrays where mitigation of interference is realized at this stage. That is, interference nulling capability of statistical pre-beamformers is ignored while considering constant-modulus constraint in prior work. Furthermore, dynamic subarray design is not considered for interference-aware pre-beamformers in the literature.

B. CONTRIBUTIONS

In this paper, we design interference-aware analog beamformers for JSDM framework utilizing SC-FDE in uplink transmission. Our contributions in this paper are as follows:

- We firstly design a near-optimal unconstrained analog beamformer that considers interference. Unlike most prior work, we concentrate on analog beamformers which are updated with slowly varying CCMs. We try to maximize mutual information between the frequency domain received signal and intended group's signal in reduced dimension, i.e., after the analog beamformer. Although we employ a frequency-selective channel model, cost function turns out to be independent of frequency bin for spatial-narrowband arrays. Near-optimal unconstrained analog beamformer that maximizes the mutual information in reduced dimension coincides with GEB concept. This type of beamformers have strong interference capability, and their optimality can be shown with respect to several criteria [18], [19].
- We propose algorithms to obtain constant-modulus constrained approximations of GEB for both fully and partially connected arrays. We decompose the analog beamformer into two stages, namely, constrained analog beamformer and compensation matrix. Compensation matrix is applied in the digital baseband just before digital beamformers. Compensation matrix increases the degrees of freedom in the optimization problem which is formulated as minimization of the Euclidean distance between GEB and combination of constrained analog beamformer and compensation matrix. We obtain a constrained solution for fully connected arrays by utilizing an alternating minimization-based algorithm. Then,

TABLE 1. Summary of existing work on constrained beamformer design with fully connected and partially connected arrays (fixed and dynamic subarrays).

Papers	[5]	[6]	[7]	[9], [10], [20]	[18], [19]	[23]	[24]	[25]	[26]	[27], [28]	[29]	[30], [31]	This Work
Frequency-Selectivity	✗	✓	✗	✗	✓	✗	✓	✗	✗	✓	✓	✗	✓
Multuser	✗	✗	✓	✓	✓	✓	✗	✗	✗	✗	✓	✓	✓
Two-Stage Beamforming (Spatial Statistical Pre-Beamformer)	✗	✗	✓	✓	✓	✓	✓	✓	✗	✗	✗	✓	✓
Interference-Aware Pre-Beamformer	✗	✗	✗	✓	✓	✗	✗	✓	✗	✗	✗	✗	✓
Constrained Analog Beamformer	✓	✓	✓	✗	✗	✓	✗	✓	✓	✓	✓	✓	✓
Fully Connected Array	✓	✓	✓	✓	✓	✓	✓	✗	✗	✗	✗	✗	✓
Fixed Subarray	✗	✓	✗	✗	✗	✗	✓	✓	✓	✓	✓	✓	✓
Dynamic Subarray	✗	✗	✗	✗	✗	✗	✓	✗	✓	✓	✓	✓	✓

we present the algorithm to find the constrained solution for partially connected arrays with any fixed connection structure. Furthermore, we propose a low-complexity algorithm to find the optimal partially connected array structure that yields the maximum expected signal-to-interference-plus-noise ratio (SINR) in reduced dimension by exploiting the properties of GEB. The last problem is known as dynamic subarray design. To the authors' best knowledge, this is the first dynamic subarray algorithm for slowly varying statistical analog beamformers in JSDM framework.

- We promote the usage of minimum mean square error (MMSE) criterion based iterative equalizers for digital beamforming stage where the residual interference after constrained analog beamformers, which cannot mitigate the inter-group interference completely, is considered. We adopt IB-DFE method that is shown to be an effective approach for SC transmission [34]. In this way, interference is taken into account in both analog and digital beamforming stages.
- In JSDM framework, users usually have single antenna elements [10], and multiantenna user extension of statistical analog beamforming with spatially correlated channels is not studied in the literature. We provide an analysis to address this issue by showing that proposed unconstrained GEB and constrained analog beamformer design algorithms can be readily extended to multiantenna user scenario. Moreover, our analysis implies that digital beamforming and channel estimation schemes would be unchanged.
- We provide a comprehensive analysis by using several performance measures; beam pattern, spectral efficiency, outage capacity, bit-error rate (BER), and accuracy of channel estimation in reduced dimension. Beam pattern analysis of constrained analog beamformers, especially partially connected arrays, is lacking in the literature. Our analysis provides novel insights about

the interference suppression capabilities of constrained analog beamformers.

In summary, we design interference-aware slowly varying analog beamformers in JSDM framework, unlike most prior work as it can be seen from Table 1. However, our design can be used in any framework with the presence of interference whose covariance matrix is known. In addition, proposed algorithms are evaluated with simulations by using the provided analysis tools. We consider a scenario where a group is assumed to be mobile so that its multipath components (MPCs) can be overlapped with angular regions of other groups, which leads to an increase in inter-group interference in certain situations. In this way, interference suppression capabilities of constrained analog beamformers can be observed. The results show that proposed fully connected array algorithm is superior to DFT beamformer and phase of GEB which are considered as state-of-the-art. Proposed algorithm attains the performance of GEB with moderate interference strength. Furthermore, proposed dynamic subarray algorithm outperforms commonly used fixed subarray structures.

Notations: Scalars, column vectors and matrices are denoted by lowercase (e.g., x), lowercase boldface (e.g., \mathbf{x}) and uppercase boldface (e.g., \mathbf{X}) letters, respectively. $|x|$ and x^* are the magnitude and complex conjugate of scalar x , respectively. \mathbf{X}^T , \mathbf{X}^H , \mathbf{X}^{-1} and $|\mathbf{X}|$ represent the transpose, Hermitian, inverse and determinant of matrix \mathbf{X} , respectively. $[\mathbf{X}]_{(i,j)}$ is the entry of matrix \mathbf{X} at i^{th} row and j^{th} column. $[\mathbf{X}]_{(i,:)}$ and $[\mathbf{X}]_{(:,j)}$ are used to extract i^{th} row and j^{th} column of matrix \mathbf{X} , respectively. $\mathbb{E}\{\cdot\}$ and $\text{Tr}\{\cdot\}$ are the expectation and trace operators, respectively. $\|\mathbf{x}\|$ and $\|\mathbf{X}\|_F$ denote the Euclidean norm of vector \mathbf{x} and Frobenius norm of matrix \mathbf{X} , respectively. $\text{diag}\{\cdot\}$ and $\text{blkdiag}\{\cdot\}$ are used to construct diagonal and block diagonal matrices, respectively. \mathbf{I}_N is the identity matrix with size $N \times N$, $\angle(\cdot)$ extracts the phase of given input, $\text{vec}(\mathbf{X})$ represents the vectorization of matrix \mathbf{X} , $\mathcal{CN}(\mathbf{x}, \mathbf{R})$ is a complex Gaussian random vector with mean

\mathbf{x} and covariance \mathbf{R} , $P(\cdot)$ represents the probability value, \otimes denote the Kronecker product between two matrices, and δ_{ij} is the Kronecker delta function.

II. SYSTEM MODEL

We consider an uplink SC-FDE system where the base station (BS) with M antennas serves K single-antenna users.¹ Total number of RF chains is denoted by D where $D \leq M$. Users are partitioned into G groups according to their channel covariance eigenspaces.² Number of users in group g is denoted by K_g , and number of RF chains allocated to this group is D_g , where $K_g \leq D_g$ relation should be satisfied to successfully separate intra-group user signals. Considering $K = \sum_{g=1}^G K_g$ and $D = \sum_{g=1}^G D_g$, the constraint on total number of RF chains can be stated as $K \leq D \leq M$.

A. WIDEBAND SC-FDE UPLINK TRANSMISSION

Received signal vector $\mathbf{y}_n \in \mathbb{C}^{M \times 1}$ at the BS at n^{th} time instance is expressed as

$$\mathbf{y}_n = \sum_{g=1}^G \sum_{l=0}^{L-1} \mathbf{H}_l^{(g)} \mathbf{x}_{(n-l)_N}^{(g)} + \mathbf{n}_n, \quad (1)$$

for $n = 0, 1, \dots, N - 1$, where block length is denoted by N . In (1), $\mathbf{H}_l^{(g)} \triangleq [\mathbf{h}_l^{(g_1)}, \dots, \mathbf{h}_l^{(g_{K_g})}] \in \mathbb{C}^{M \times K_g}$ represent the channel matrix of group g at l^{th} MPC where the channel vector of user m in group g is denoted by $\mathbf{h}_l^{(g_m)}$ and the total number of MPCs is denoted by L . A cyclic prefix with length larger than L is used to obtain circulant channel matrices while preventing the inter-block interference. Transmitted symbol vector of group g is defined as $\mathbf{x}_n^{(g)} \triangleq [x_n^{(g_1)}, \dots, x_n^{(g_{K_g})}]^T \in \mathbb{C}^{K_g \times 1}$ with $\mathbb{E}\{x_n^{(g_m)}(x_{n'}^{(g_{m'})})^*\} = \frac{E_s^{(g)}}{K_g} \delta_{gg'} \delta_{mm'} \delta_{nn'}$ where $x_n^{(g_m)}$ is the transmitted symbol of user m in group g . Total transmit energy of group g is denoted by $E_s^{(g)}$. Noise vector \mathbf{n}_n is comprised of zero-mean circularly symmetric complex Gaussian random variables with variance N_0 .

Channel vectors obey the spatially correlated Rayleigh channel model with $\mathbf{h}_l^{(g_m)} \sim \mathcal{CN}(\mathbf{0}, \mathbf{R}_l^{(g_m)})$ where $\mathbf{R}_l^{(g_m)}$ is the CCM of user m in group g at l^{th} MPC [9], [10], [18]–[20], [23]. User channel vectors are assumed to be mutually uncorrelated among MPCs and users which is expressed as

$$\mathbb{E}\left\{\mathbf{h}_l^{(g_m)} \left[\mathbf{h}_{l'}^{(g_{m'})}\right]^H\right\} = \mathbf{R}_l^{(g_m)} \delta_{gg'} \delta_{mm'} \delta_{ll'}. \quad (2)$$

¹Users are assumed to have single-antenna elements for simplicity. However, this framework can be extended to multi-antenna users and proposed methods in this paper can be used as shown in Appendix B.

²The design of user grouping algorithms is out of the scope of this paper. Efficient user-grouping procedures can be found in [35], [36].

Considering the one-ring scattering model [9], [10], [23], CCM of a user at l^{th} MPC is defined as

$$\mathbf{R}_l^{(g_m)} \triangleq \gamma^{(g_m)} \int_{\mu_l^{(g_m)} - \frac{\Delta_l^{(g_m)}}{2}}^{\mu_l^{(g_m)} + \frac{\Delta_l^{(g_m)}}{2}} \rho_l^{(g_m)}(\theta) \mathbf{u}(\theta) \mathbf{u}(\theta)^H d\theta, \quad (3)$$

where $\sqrt{\gamma^{(g_m)}}$ is the channel gain of user m in group g satisfying $\sum_{l=0}^{L-1} \text{Tr}\{\mathbf{R}_l^{(g_m)}\} = \gamma^{(g_m)}$ relation. For l^{th} MPC of user m in group g , angular power profile function, mean angle of arrival (AoA) and angular spread (AS) are denoted by $\rho_l^{(g_m)}(\theta)$, $\mu_l^{(g_m)}$ and $\Delta_l^{(g_m)}$, respectively. Unit norm steering vector $\mathbf{u}(\theta) \in \mathbb{C}^{M \times 1}$ of uniform linear array (ULA) with half the wavelength spacing corresponding to azimuth angle θ is defined as

$$\mathbf{u}(\theta) \triangleq \frac{1}{\sqrt{M}} [1 e^{j\pi \sin(\theta)} \dots e^{j(M-1)\pi \sin(\theta)}]^T. \quad (4)$$

Although we considered ULA, formulations in this paper can be used for other array structures. The BS observes a sparse angle-delay profile at mm-wave frequencies [3]. That is, users have a few active MPCs with narrow angular spread which makes user-grouping feasible. In this paper, it is assumed that users in the same group have the same active MPCs with similar mean AoAs constituting MPC clusters.

B. TWO-STAGE BEAMFORMING CONCEPT

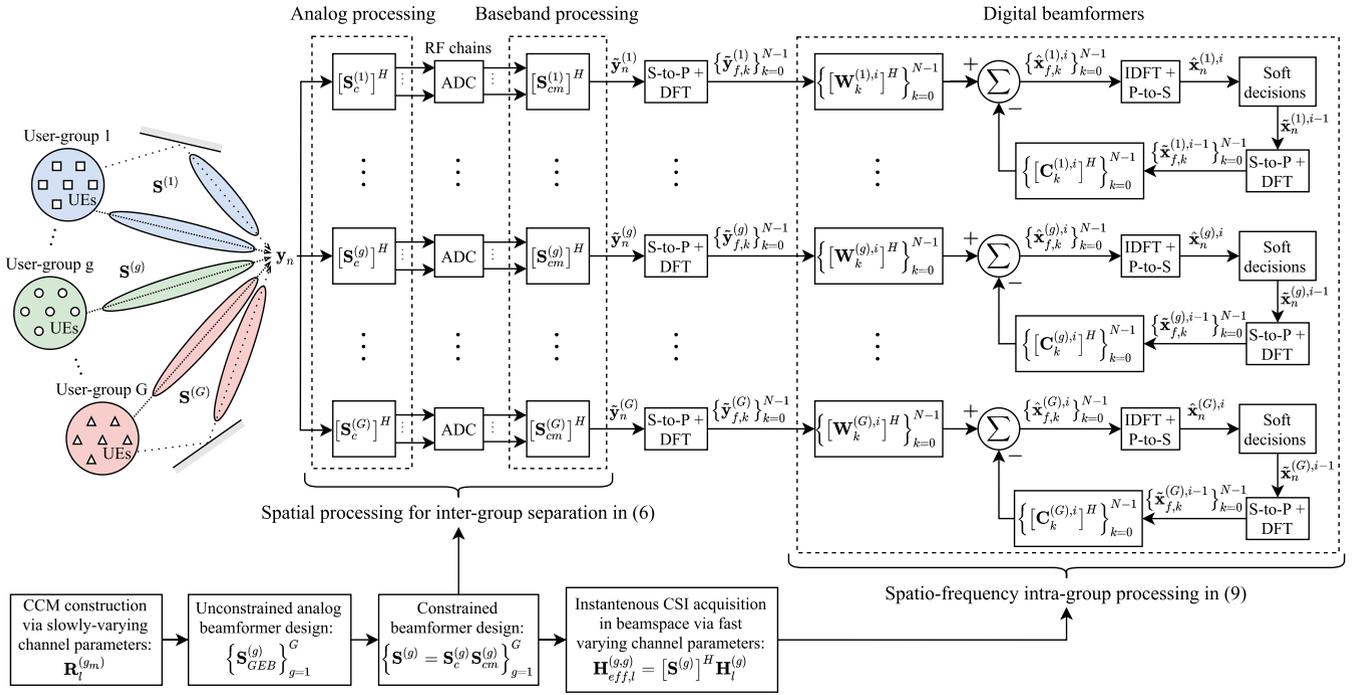
This paper considers a two-stage beamforming architecture based on JSDM where there is an analog beamformer $\mathbf{S}^{(g)} \in \mathbb{C}^{M \times D_g}$ for each group suppressing channels of users in other groups. Let the received signal in (1) is rewritten as

$$\mathbf{y}_n = \underbrace{\sum_{l=0}^{L-1} \mathbf{H}_l^{(g)} \mathbf{x}_{(n-l)_N}^{(g)}}_{\mathbf{s}_n^{(g)}: \text{Intra-Group Signals}} + \underbrace{\sum_{g' \neq g} \sum_{l=0}^{L-1} \mathbf{H}_l^{(g')} \mathbf{x}_{(n-l)_N}^{(g')}}_{\mathbf{\eta}_n^{(g)}: \text{Inter-Group Signals + Noise Terms}} + \mathbf{n}_n, \quad (5)$$

where intra-group and inter-group signals are separated for group g . Then, the signal vector of group g after its analog beamformer can be expressed as

$$\tilde{\mathbf{y}}_n^{(g)} = [\mathbf{S}^{(g)}]^H \mathbf{y}_n = \underbrace{\sum_{l=0}^{L-1} \mathbf{H}_{\text{eff},l}^{(g,g)} \mathbf{x}_{(n-l)_N}^{(g)}}_{\tilde{\mathbf{s}}_n^{(g)}: \text{Intra-Group Signals in RD}} + \underbrace{\sum_{g' \neq g} \sum_{l=0}^{L-1} \mathbf{H}_{\text{eff},l}^{(g,g')} \mathbf{x}_{(n-l)_N}^{(g')}}_{\tilde{\mathbf{\eta}}_n^{(g)}: \text{Inter-Group Signals + Noise Terms in RD}} + [\mathbf{S}^{(g)}]^H \mathbf{n}_n, \quad (6)$$

where effective channel matrices at l^{th} MPC are denoted by $\mathbf{H}_{\text{eff},l}^{(g,g')} \triangleq [\mathbf{S}^{(g)}]^H \mathbf{H}_l^{(g')} \in \mathbb{C}^{D_g \times K_{g'}}$. Intra-group signals and sum of inter-group signals and noise in reduced dimension (RD) are denoted by $\tilde{\mathbf{s}}_n^{(g)} = [\mathbf{S}^{(g)}]^H \mathbf{s}_n^{(g)} \in \mathbb{C}^{D_g \times 1}$ and $\tilde{\mathbf{\eta}}_n^{(g)} = [\mathbf{S}^{(g)}]^H \mathbf{\eta}_n^{(g)} \in \mathbb{C}^{D_g \times 1}$, respectively.


FIGURE 1. Block diagram of the overall system design with constrained analog beamformers.

In this paper, frequency domain equalization (FDE) is employed to mitigate ISI caused by SC transmission. Since FDE is employed, DFT operation is applied to the sequence $\{\tilde{\mathbf{y}}_n^{(g)}\}_{n=0}^{N-1}$. Using the circularity in (6), one can obtain the signal at the k^{th} frequency bin after the DFT operation as

$$\tilde{\mathbf{y}}_{f,k}^{(g)} = \underbrace{\Lambda_{\text{eff},k}^{(g,g)} \mathbf{x}_{f,k}^{(g)}}_{\tilde{\mathbf{s}}_{f,k}^{(g)}} + \underbrace{\sum_{g' \neq g} \Lambda_{\text{eff},k}^{(g,g')} \mathbf{x}_{f,k}^{(g')}}_{\tilde{\mathbf{q}}_{f,k}^{(g)}} + \tilde{\mathbf{n}}_{f,k}^{(g)}, \quad (7)$$

for $k = 0, 1, \dots, N-1$, where the DFT of effective channel matrices are denoted by $\Lambda_{\text{eff},k}^{(g,g')} = \sum_{l=0}^{L-1} \mathbf{H}_{\text{eff},l}^{(g,g')} e^{-j\frac{2\pi}{N}kl}$. In (7), $\tilde{\mathbf{y}}_{f,k}^{(g)}$, $\tilde{\mathbf{s}}_{f,k}^{(g)}$, $\mathbf{x}_{f,k}^{(g)}$, $\tilde{\mathbf{q}}_{f,k}^{(g)}$ and $\tilde{\mathbf{n}}_{f,k}^{(g)}$ are normalized DFTs of vector sequences $\tilde{\mathbf{y}}_n^{(g)}$, $\tilde{\mathbf{s}}_n^{(g)}$, $\mathbf{x}_n^{(g)}$, $\tilde{\mathbf{q}}_n^{(g)}$ and $\tilde{\mathbf{n}}_n^{(g)} = [\mathbf{S}^{(g)}]^H \mathbf{n}_n$, respectively. Note that normalized DFT of any vector sequence $\{\mathbf{v}_n\}_{n=0}^{N-1}$ can be calculated as

$$\mathbf{v}_{f,k} \triangleq \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} \mathbf{v}_n e^{-j\frac{2\pi}{N}kn}, \quad k = 0, 1, \dots, N-1. \quad (8)$$

Digital beamformers are used to separate intra-group signals of users in JSDM framework. Since we employed FDE structure, digital beamformers are applied in frequency domain where IB-DFE method is utilized via feedforward and feedback filters which are updated at each iteration. Feedforward and feedback filters of group g for k^{th} frequency bin at the i^{th} equalizer iteration are denoted by $\mathbf{W}_k^{(g),i} \in \mathbb{C}^{D_g \times K_g}$ and $\mathbf{C}_k^{(g),i} \in \mathbb{C}^{K_g \times K_g}$, respectively. The output of DFE at the i^{th} iteration, which is found by using (7) and soft decisions of

previous iteration, is expressed as follows:

$$\hat{\mathbf{x}}_{f,k}^{(g),i} = [\mathbf{W}_k^{(g),i}]^H \tilde{\mathbf{y}}_{f,k}^{(g)} - [\mathbf{C}_k^{(g),i}]^H \hat{\mathbf{x}}_{f,k}^{(g),i-1}, \quad (9)$$

$$\hat{\mathbf{x}}_n^{(g),i} = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} \hat{\mathbf{x}}_{f,k}^{(g),i} e^{j\frac{2\pi}{N}kn} \quad (10)$$

for $k, n = 0, 1, \dots, N-1$. Finally, time-domain estimates $\hat{\mathbf{x}}_n^{(g),i} \triangleq [\hat{x}_n^{(g_1),i}, \dots, \hat{x}_n^{(g_{K_g}),i}]^T$ are used to demodulate the transmitted symbols, where $\hat{x}_n^{(g_m),i}$ is the estimate of $x_n^{(g_m)}$. In (9), $\hat{\mathbf{x}}_{f,k}^{(g),i-1}$ is the soft decision vector of group g obtained from the previous iteration. Block diagram summarizing the overall two-stage beamforming architecture described in this section is given in Fig. 1. Note that analog beamforming stage is given for the constrained beamformer design that will be introduced in Section IV.

III. UNCONSTRAINED HYBRID MASSIVE MIMO BEAMFORMER DESIGN

In JSDM framework, analog beamformers are designed by using CCMs that are assumed to be known perfectly in this paper.³ Assuming inter-group interference is completely mitigated at analog beamforming stage, digital processing of each group is implemented independently. In this section, analog and digital beamformer designs are considered.

³CCM estimation is out of the scope of this paper. However, there are several papers proposing efficient methods for CCM estimation [37]–[40]. It can be assumed that CCMs can be learned by using the methods mentioned in these papers. It is important to note that CCMs depend on slowly varying parameters such as AoAs, which is the reason why they do not need to be updated frequently. Hence, CCM estimation overhead does not cause a significant increase in overall system complexity.

A. OPTIMAL ANALOG BEAMFORMER DESIGN: GENERALIZED EIGENBEAMFORMER (GEB)

In this section, we find a near-optimal unconstrained analog beamformer structure for JSDM framework where frequency-selectivity of channels is considered. We approach the analog beamformer design as a dimension reduction problem. That is, we aim to find a near-optimal unconstrained analog beamformer, which results in a linear transformation, that preserves the mutual information in reduced dimension. The cost function that needs to be maximized in reduced dimension is the mutual information between $\tilde{\mathbf{y}}_{f,k}^{(g)}$ and $\tilde{\mathbf{s}}_{f,k}^{(g)}$ in (7) which can be expressed as

$$I(\tilde{\mathbf{s}}_{f,k}^{(g)}; \tilde{\mathbf{y}}_{f,k}^{(g)}) = \log_2 \left(\mathbf{I}_{D_g} + \left[\mathbf{R}_{\tilde{\eta}_f}^{(g)} \right]^{-1} \mathbf{R}_{\tilde{\mathbf{s}}_f}^{(g)} \right), \quad (11)$$

if $\tilde{\mathbf{y}}_{f,k}^{(g)}$ and $\tilde{\mathbf{s}}_{f,k}^{(g)}$ are assumed to be Gaussian random variables. Covariance matrices of $\tilde{\mathbf{s}}_{f,k}^{(g)}$ and $\tilde{\eta}_{f,k}^{(g)}$ are denoted by $\mathbf{R}_{\tilde{\mathbf{s}}_f}^{(g)} = \mathbb{E} \left\{ \tilde{\mathbf{s}}_{f,k}^{(g)} [\tilde{\mathbf{s}}_{f,k}^{(g)}]^H \right\}$ and $\mathbf{R}_{\tilde{\eta}_f}^{(g)} = \mathbb{E} \left\{ \tilde{\eta}_{f,k}^{(g)} [\tilde{\eta}_{f,k}^{(g)}]^H \right\}$ for $k = 0, 1, \dots, N - 1$, respectively. Note that these covariance matrices are independent of frequency bin index k according to Appendix A.⁴ Hence, there will be a single cost function to maximize. Furthermore, it is easy to see that maximization of mutual information is equivalent to maximization of the determinant inside the logarithm in (11). If we replace $\mathbf{R}_{\tilde{\mathbf{s}}_f}^{(g)}$ and $\mathbf{R}_{\tilde{\eta}_f}^{(g)}$ with their expressions given in Appendix A, the cost function that needs to be maximized under the constraint that $\mathbf{S}^{(g)}$ is a full column matrix can be written as

$$\begin{aligned} & \text{maximize}_{\mathbf{S}^{(g)}} \left| \mathbf{I}_{D_g} + \left([\mathbf{S}^{(g)}]^H \mathbf{R}_{\tilde{\eta}_f}^{(g)} \mathbf{S}^{(g)} \right)^{-1} \left([\mathbf{S}^{(g)}]^H \mathbf{R}_{\tilde{\mathbf{s}}_f}^{(g)} \mathbf{S}^{(g)} \right) \right| \\ & \text{subject to } \text{rank}(\mathbf{S}^{(g)}) = D_g \end{aligned} \quad (12)$$

where covariance matrix of intra-group signals and covariance matrix of sum of inter-group signals and noise terms in (5) are expressed as $\mathbb{E} \left\{ \mathbf{s}_n^{(g)} [\mathbf{s}_n^{(g)}]^H \right\} = \mathbf{R}_s^{(g)} \delta_{nn'}$ and $\mathbb{E} \left\{ \eta_n^{(g)} [\eta_n^{(g)}]^H \right\} = \mathbf{R}_{\eta}^{(g)} \delta_{nn'}$, respectively. Considering the uncorrelated nature of transmitted symbols among users and time in addition to the uncorrelated channel vectors in (2), these two covariance matrices are computed as

$$\mathbf{R}_s^{(g)} = \frac{E_s^{(g)}}{K_g} \sum_{m=1}^{K_g} \sum_{l=0}^{L-1} \mathbf{R}_l^{(g_m)}, \quad (13)$$

$$\mathbf{R}_{\eta}^{(g)} = \sum_{g' \neq g} \frac{E_s^{(g')}}{K_{g'}} \sum_{m=1}^{K_{g'}} \sum_{l=0}^{L-1} \mathbf{R}_l^{(g'_m)} + N_0 \mathbf{I}_M. \quad (14)$$

⁴In this paper, although frequency-wideband effect is taken into account, spatial-wideband (beam-squint) effect is ignored as bandwidth is assumed to be not that wide [41]. If a spatial-wideband system was considered, spatial covariance matrices of signals would depend on the frequency bin index. However, proposed analog beamforming could still be used either by using the mean spatial covariance matrices, which are averaged over frequency bins, to find the unconstrained analog beamformer as in [27], [42] (which do not consider the interference rejection capability of pre-beamformer) or by finding the unconstrained analog beamformer with spatial-narrowband assumption and applying a beam-squint compensation in digital domain [43].

Lemma 1: If multiantenna users were considered, design of statistical analog beamformer $\mathbf{S}^{(g)}$, which is based on the optimization problem in (12), would remain the same with additional knowledge of scaling constants that appear in (13) and (14).

Proof: See Appendix B. ■

Analog beamformer $\tilde{\mathbf{S}}^{(g)}$ that maximizes the cost function is found as [44]

$$\tilde{\mathbf{S}}^{(g)} = [\mathbf{v}_1, \mathbf{v}_2, \dots, \mathbf{v}_{D_g}], \quad (15)$$

where \mathbf{v}_i is the eigenvector corresponding to the i^{th} most dominant eigenvalue (i.e., λ_i) of the generalized eigenvalue problem which is defined as

$$\mathbf{R}_s^{(g)} \mathbf{v}_i = \lambda_i \mathbf{R}_{\eta}^{(g)} \mathbf{v}_i, \quad i = 1, 2, \dots, M. \quad (16)$$

This solution is known as generalized eigenbeamformer which creates deep nulls for interfering signals. It is meaningful to use $\mathbf{R}_s^{(g)}$, which covers CCMs of all MPCs of intended group, in the generalized eigenvalue problem in (16) since SC-FDE is employed. In order to obtain orthonormalized analog beamformers, QR decomposition of GEB in (15) is found as $\tilde{\mathbf{S}}^{(g)} = \mathbf{S}^{(g)} \tilde{\mathbf{R}}^{(g)}$, which is permissible according to Lemma 2. Equivalent analog beamformer $\mathbf{S}^{(g)}$ found by QR decomposition is used as GEB for the rest of the paper.

Lemma 2: Using $\mathbf{S}^{(g)} \mathbf{A}$ instead of $\mathbf{S}^{(g)}$ does not change the cost function where $\mathbf{A} \in \mathbb{C}^{D_g \times D_g}$ is any invertible matrix.

Proof: See Appendix C. ■

B. DIGITAL BEAMFORMER DESIGN: IB-DFE

As digital processing of each user group is independent from each other, only intra-group effective channels of users are required for digital beamformer design. Effective channel vectors are learned in time division duplexing (TDD) mode as in Section V-B. Intra-group channel estimation in reduced dimension reduces the channel estimation overhead significantly. In addition, dimensions of feedforward and feedback filters are reduced with the help of user-grouping since IB-DFE is used to separate intra-group user signals.

Frequency domain equalizer output at i^{th} iteration was defined in (9) where feedforward and feedback filters were utilized. Optimal filters at each iteration can be found by using MMSE criterion where the total mean square error (MSE) of a block for group g at i^{th} iteration is defined as

$$MSE_i^{(g)} = \mathbb{E} \left\{ \sum_{k=0}^{N-1} \left\| \hat{\mathbf{x}}_{f,k}^{(g),i} - \mathbf{x}_{f,k}^{(g),i} \right\|^2 \right\}. \quad (17)$$

In addition, following constraint is imposed on feedback filters to avoid self-cancellation: $\sum_{k=0}^{N-1} [\mathbf{C}_k^{(g),i}]_{(m,m)} = 0$ for $m = 1, 2, \dots, K_g$. In [34], it is shown that minimization of (17) with the given constraint reduces to following N parallel equations

$$\mathbf{R}_{\tilde{\mathbf{y}}_{f,k}^{(g)}} \mathbf{W}_k^{(g),i} = \Lambda_{\text{eff},k}^{(g,g)} \left(\frac{E_s^{(g)}}{K_g} \mathbf{I}_{K_g} + \mathbf{P}_i^{(g)} \mathbf{C}_k^{(g),i} \right), \quad (18)$$

$$\mathbf{P}_i^{(g)} \mathbf{C}_k^{(g),i} = [\mathbf{P}_i^{(g)}]^H \left([\mathbf{\Lambda}_{eff,k}^{(g,g)}]^H \mathbf{W}_k^{(g),i} - \mathbf{I}_{K_g} \right) + \mathbf{\Delta}_i^{(g)} \quad (19)$$

for $k = 0, 1, \dots, N-1$, where $\mathbf{P}_i^{(g)} \in \mathbb{C}^{K_g \times K_g}$ is the reliability matrix and $\mathbf{\Delta}_i^{(g)} = \text{diag}\{\Delta_i^{(g,1)}, \dots, \Delta_i^{(g,K_g)}\}$ is a consequence of the constraint on feedback filters. Reliability matrices have non-zero diagonal entries that are updated at each iteration and are the same for all frequency bins. Diagonal entries of reliability matrices can be calculated by using soft decisions as done in [34]. Considering the fact that transmitted symbol vectors of users are uncorrelated and independent from the noise vector, covariance matrix of $\tilde{\mathbf{y}}_{f,k}^{(g)}$ in (7) can be expressed as

$$\mathbf{R}_{\tilde{\mathbf{y}}_{f,k}^{(g)}} = \mathbb{E} \left\{ \tilde{\mathbf{y}}_{f,k}^{(g)} [\tilde{\mathbf{y}}_{f,k}^{(g)}]^H \right\} = \frac{E_s^{(g)}}{K_g} \mathbf{\Lambda}_{eff,k}^{(g,g)} [\mathbf{\Lambda}_{eff,k}^{(g,g)}]^H + \mathbf{R}_{\tilde{\eta}_f}^{(g)}, \quad (20)$$

for given frequency-domain intra-group channel matrix $\mathbf{\Lambda}_{eff,k}^{(g,g)}$. Recall that intra-group channels are available at the BS whereas inter-group channel matrices $\mathbf{\Lambda}_{eff,k}^{(g,g')}$ are not known. That is why inter-group channel matrices are treated as random matrices in (20) and covariance matrix of sum of inter-group interference and noise terms in reduced dimension $\mathbf{R}_{\tilde{\eta}_f}^{(g)}$ whose expression is given in Appendix A is used.⁵

If $\mathbf{W}_k^{(g),i}$ is substituted in (19) for each frequency bin while considering the constraint on feedback filters, $\mathbf{C}_k^{(g),i}$ and $\Delta_i^{(g,m)}$ can be found as

$$\mathbf{C}_k^{(g),i} = \mathbf{A}_k^{(g),i} \left(\mathbf{D}_k^{(g),i} + \mathbf{\Delta}_i^{(g)} \right), \quad (21)$$

$$\Delta_i^{(g,m)} = \frac{-\sum_{k=0}^{N-1} [\mathbf{A}_k^{(g),i}]_{(m,:)} [\mathbf{D}_k^{(g),i}]_{(:,m)}}{\sum_{k=0}^{N-1} [\mathbf{A}_k^{(g),i}]_{(m,m)}}, \quad (22)$$

where

$$\mathbf{A}_k^{(g),i} = \left(\mathbf{P}_i^{(g)} - [\mathbf{P}_i^{(g)}]^H [\mathbf{\Lambda}_{eff,k}^{(g,g)}]^H \mathbf{R}_{\tilde{\mathbf{y}}_{f,k}^{(g)}}^{-1} \mathbf{\Lambda}_{eff,k}^{(g,g)} \mathbf{P}_i^{(g)} \right)^{-1}, \quad (23)$$

$$\mathbf{D}_k^{(g),i} = [\mathbf{P}_i^{(g)}]^H [\mathbf{\Lambda}_{eff,k}^{(g,g)}]^H \mathbf{R}_{\tilde{\mathbf{y}}_{f,k}^{(g)}}^{-1} \mathbf{\Lambda}_{eff,k}^{(g,g)} \frac{E_s^{(g)}}{K_g} - [\mathbf{P}_i^{(g)}]^H. \quad (24)$$

Lastly, feedforward filters can be found by substituting $\mathbf{C}_k^{(g),i}$ in (18) for each frequency bin. At initialization stage, reliability matrices are set to $\mathbf{P}_0^{(g)} = \text{diag}\{\mathbf{0}_{K_g}\}$ where $\mathbf{0}_{K_g}$ is a zero vector with length K_g . Consequently, there is no feedback at initialization while feedforward filters are reduced to linear minimum mean square error (LMMSE) type digital beamformers that can be expressed as

$$\mathbf{W}_k^{(g),0} = \frac{E_s^{(g)}}{K_g} \mathbf{R}_{\tilde{\mathbf{y}}_{f,k}^{(g)}}^{-1} \mathbf{\Lambda}_{eff,k}^{(g,g)}. \quad (25)$$

If analog beamformers are properly designed, inter-group interference terms in (7) will be close to zero. In this case,

⁵Estimation of only intra-group effective channels reduces the channel estimation overhead. Furthermore, there may be other sources of interference with known spatial covariance matrices such as inter-cell interference. In that case, instantaneous channel estimation for the interfering source would not be possible, yet our framework would still be applicable.

zero-forcing (ZF) approach can be adopted at initialization. ZF type digital beamformer of group g is expressed as

$$\mathbf{W}_k^{(g),0} = \mathbf{\Lambda}_{eff,k}^{(g,g)} \left([\mathbf{\Lambda}_{eff,k}^{(g,g)}]^H \mathbf{\Lambda}_{eff,k}^{(g,g)} \right)^{-1}. \quad (26)$$

The main advantage of ZF is its computation complexity since it involves inversion of matrices with size $K_g \times K_g$ per frequency bin resulting in a complexity order of $O(NK_g^3)$, whereas LMMSE involves inversion of covariance matrix of $\tilde{\mathbf{y}}_{f,k}^{(g)}$ per frequency bin that results in a complexity order of $O(ND_g^3)$. On the other hand, IB-DFE structure requires the calculation of feedforward and feedback filters at each iteration. Computation complexity of feedforward filters is equivalent to LMMSE filter while feedback filters require inversion of matrices with size $K_g \times K_g$ and $D_g \times D_g$ per frequency bin which results in a complexity order of $O(ND_g^3)$ as $K_g \leq D_g$. Thus, overall complexity of IB-DFE becomes $O(i_{max}ND_g^3)$ where i_{max} is the number of equalizer iterations. It is important to note that complexity order of described digital beamforming design scales with the third power of either K_g or D_g which are limited in JSDM framework. Hence, signal processing complexity of digital beamforming is highly reduced via user-grouping and two-stage beamforming approach.

IV. CONSTRAINED ANALOG BEAMFORMER DESIGN

In this paper, optimality is imposed at the analog beamforming stage. Near-optimal GEB introduced in Section III-A is an unconstrained beamformer, i.e., it does not obey constant-modulus constraint. However, phase-shifter networks are preferred for the analog beamforming stage in practice [3]. Hence, our goal is to find a constant-modulus approximation to near-optimal GEB of each group. In addition, we provide a DFT beamformer structure which is obtained by using AoAs belonging to MPC clusters of user groups. In this section, near-optimal GEB of group g found in (15) in Section III-A and its constant-modulus constrained equivalent are denoted by $\mathbf{S}_{GEB}^{(g)} \in \mathbb{C}^{M \times D_g}$ and $\mathbf{S}_c^{(g)} \in \mathbb{C}^{M \times D_g}$, respectively.

A. FULLY CONNECTED ARRAY

1) DFT BEAMFORMER

DFT beamformer, which will be used for comparison purposes, is widely used in the literature [9], [21], [22]. It is not an interference aware beamformer. In other words, it only considers the AoAs of MPC clusters in the intended group and does not consider the channels or signals of other groups. Let $\mathbf{Q} \in \mathbb{C}^{M \times M}$ be the normalized DFT matrix with entries $[\mathbf{Q}]_{(m,n)} \triangleq \frac{1}{\sqrt{M}} e^{-j\frac{2\pi}{M}mn}$ for $m, n = 0, 1, \dots, M-1$. Then, DFT beamformer of group g is defined as

$$\mathbf{S}_c^{(g)} \triangleq \mathbf{Q}\mathbf{\Gamma}^{(g)}, \quad (27)$$

where $\mathbf{\Gamma}^{(g)} = [\mathbf{e}_{g,1}, \mathbf{e}_{g,2}, \dots, \mathbf{e}_{g,D_g}] \in \mathbb{Z}^{M \times D_g}$ is used as the column selection matrix for group g . Columns of $\mathbf{\Gamma}^{(g)}$ are denoted by $\mathbf{e}_i \in \mathbb{Z}^{M \times 1}$ which is a binary vector whose i^{th} entry is one whereas others are zero. This vector selects

the columns of the DFT matrix for the analog beamformer. For any group, firstly, closest DFT vectors to mean AoAs of active MPCs are selected. If the number of available RF chains is greater than the number of active MPCs, which is the desired case, remaining columns are selected from the adjacent DFT vectors to the ones closest to mean AoAs. Analog beamformer used in practice is expressed as $\mathbf{S}^{(g)} = \mathbf{S}_c^{(g)}$.

2) PHASE EXTRACTION (PE)

Constrained analog beamformer $\mathbf{S}_c^{(g)}$ has only constant-modulus constraint for the fully connected structure, i.e., RF chains are connected to all the antennas. Since our goal is to find the best approximation to GEB, an optimization problem can be constructed as follows:

$$\begin{aligned} & \underset{\mathbf{S}_c^{(g)}}{\text{minimize}} \quad \left\| \mathbf{S}_{GEB}^{(g)} - \mathbf{S}_c^{(g)} \right\|_F^2 \\ & \text{subject to} \quad |[\mathbf{S}_c^{(g)}]_{(i,j)}| = \frac{1}{\sqrt{M}}, \quad \forall i, j, \end{aligned} \quad (28)$$

which has the simple phase extraction solution where optimum constrained analog beamformer can be expressed as $\mathbf{S}_c^{(g)*} = \frac{1}{\sqrt{M}} e^{j\angle(\mathbf{S}_{GEB}^{(g)})}$. This solution is used in the literature due to its simplicity. If multiuser MIMO without user-grouping is considered, phase extraction offers acceptable performance [23], [24]. However, deep nulls introduced by GEB may disappear which results in increase in inter-group interference for JSDM. Similar to DFT beamformer, practical analog beamformer is expressed as $\mathbf{S}^{(g)} = \mathbf{S}_c^{(g)}$.

3) PHASE EXTRACTION WITH ALTERNATING MINIMIZATION (PE-AM)

In order to improve the accuracy of approximation, we introduce a compensation matrix $\mathbf{S}_{cm}^{(g)} \in \mathbb{C}^{D_g \times D_g}$ which will be applied in the digital baseband after the constrained analog beamformer $\mathbf{S}_c^{(g)}$. This can be considered as dividing the analog beamformer into two stages. A new optimization problem with the compensation matrix can be constructed as

$$\begin{aligned} & \underset{\mathbf{S}_c^{(g)}, \mathbf{S}_{cm}^{(g)}}{\text{minimize}} \quad \left\| \mathbf{S}_{GEB}^{(g)} - \mathbf{S}_c^{(g)} \mathbf{S}_{cm}^{(g)} \right\|_F^2 \\ & \text{subject to} \quad |[\mathbf{S}_c^{(g)}]_{(i,j)}| = \frac{1}{\sqrt{M}}, \quad \forall i, j. \end{aligned} \quad (29)$$

Optimization problem given above can be solved with manifold optimization-based algorithms which have high complexity [6]. Columns of near-optimal GEB are mutually orthonormal whereas there is no constraint on the compensation matrix. Although optimum structure of the compensation matrix is not known, orthogonality constraint on compensation matrix is imposed to obtain a simplified optimization problem. It can be shown that the cost function in (29) is upper-bounded by $\left\| \mathbf{S}_{GEB}^{(g)} [\mathbf{S}_{cm}^{(g)}]^H - \mathbf{S}_c^{(g)} \right\|_F^2$ due to the orthogonality constraint of the compensation matrix. Modified

optimization problem with this upper bound is expressed as

$$\begin{aligned} & \underset{\mathbf{S}_c^{(g)}, \mathbf{S}_{cm}^{(g)}}{\text{minimize}} \quad \left\| \mathbf{S}_{GEB}^{(g)} [\mathbf{S}_{cm}^{(g)}]^H - \mathbf{S}_c^{(g)} \right\|_F^2 \\ & \text{subject to} \quad |[\mathbf{S}_c^{(g)}]_{(i,j)}| = \frac{1}{\sqrt{M}}, \quad \forall i, j; \quad [\mathbf{S}_{cm}^{(g)}]^H \mathbf{S}_{cm}^{(g)} = \mathbf{I}_{D_g}. \end{aligned} \quad (30)$$

Note that $\mathbf{S}_{GEB}^{(g)} [\mathbf{S}_{cm}^{(g)}]^H$ is an equivalent generalized eigen-beamformer according to Lemma 2 since $[\mathbf{S}_{cm}^{(g)}]^H$ is invertible. Given $\mathbf{S}_{cm}^{(g)}$, optimum solution for the constrained analog beamformer $\mathbf{S}_c^{(g)*}$ is a phase extraction of the equivalent beamformer that is expressed as

$$\mathbf{S}_c^{(g)*} = \frac{1}{\sqrt{M}} e^{j\angle(\mathbf{S}_{GEB}^{(g)} [\mathbf{S}_{cm}^{(g)}]^H)}. \quad (31)$$

For given constrained analog beamformer $\mathbf{S}_c^{(g)}$, optimization problem given in (30) reduces to

$$\begin{aligned} & \underset{\mathbf{S}_{cm}^{(g)}}{\text{minimize}} \quad \left\| \mathbf{S}_{GEB}^{(g)} [\mathbf{S}_{cm}^{(g)}]^H - \mathbf{S}_c^{(g)} \right\|_F^2 \\ & \text{subject to} \quad [\mathbf{S}_{cm}^{(g)}]^H \mathbf{S}_{cm}^{(g)} = \mathbf{I}_{D_g}, \end{aligned} \quad (32)$$

which is similar to orthogonal Procrustes problem [45]. Optimal compensation matrix can be found as

$$\mathbf{S}_{cm}^{(g)*} = \mathbf{V}^{(g)} [\mathbf{U}^{(g)}]^H, \quad (33)$$

where following singular value decomposition (SVD) is used, $[\mathbf{S}_{GEB}^{(g)}]^H \mathbf{S}_c^{(g)} = \mathbf{U}^{(g)} \mathbf{\Sigma}^{(g)} [\mathbf{V}^{(g)}]^H$ [6]. Alternating minimization algorithm is adopted in order to obtain optimum solutions where $\mathbf{S}_c^{(g)*}$ is calculated for given $\mathbf{S}_{cm}^{(g)}$ and $\mathbf{S}_{cm}^{(g)*}$ is calculated by using given $\mathbf{S}_c^{(g)}$ repeatedly until convergence is achieved. Summary of the described solution is given in Algorithm 1.

Algorithm 1 PE-AM

Input: $\mathbf{S}_{GEB}^{(g)}$

Initialize: $\mathbf{S}_{c,0}^{(g)} = \frac{1}{\sqrt{M}} e^{j\angle(\mathbf{S}_{GEB}^{(g)})}$, iteration index $n = 0$

1: **repeat**

2: Fix $\mathbf{S}_{c,n}^{(g)}$, find $[\mathbf{S}_{GEB}^{(g)}]^H \mathbf{S}_{c,n}^{(g)} = \mathbf{U}_n^{(g)} \mathbf{\Sigma}_n^{(g)} [\mathbf{V}_n^{(g)}]^H$

3: $\mathbf{S}_{cm,n}^{(g)} = \mathbf{V}_n^{(g)} [\mathbf{U}_n^{(g)}]^H$

4: Fix $\mathbf{S}_{cm,n}^{(g)}$, find $\mathbf{S}_{c,n+1}^{(g)} = \frac{1}{\sqrt{M}} e^{j\angle(\mathbf{S}_{GEB}^{(g)} [\mathbf{S}_{cm,n}^{(g)}]^H)}$

5: $n \leftarrow n + 1$

6: **until** $\left\| \mathbf{S}_{GEB}^{(g)} [\mathbf{S}_{cm,n}^{(g)}]^H - \mathbf{S}_{c,n+1}^{(g)} \right\|_F$ converges

Output: $\mathbf{S}_c^{(g)} = \mathbf{S}_{c,n+1}^{(g)}$ and $\mathbf{S}_{cm}^{(g)} = \mathbf{S}_{cm,n}^{(g)}$

For compensation matrix-based design, constrained beamformer $\mathbf{S}_c^{(g)}$ is applied in analog domain whereas compensation matrix $\mathbf{S}_{cm}^{(g)}$ is applied in digital baseband just before the digital beamformers. Thus, overall analog beamformer of group g becomes $\mathbf{S}^{(g)} = \mathbf{S}_c^{(g)} \mathbf{S}_{cm}^{(g)}$ as it is seen from Fig. 1.

B. PARTIALLY CONNECTED ARRAY

Unlike fully connected arrays, each antenna element is connected to only one RF chain for partially connected designs. In other words, analog beamformer has a sparse structure resulting in fewer phase shifters. Let the binary connection matrix of group g is denoted by $\mathbf{\Pi}^{(g)} \in \mathbb{Z}^{M \times D_g}$ where each row contains only one non-zero entry. Column where i^{th} row has the non-zero entry is denoted by $j(i)$. Constrained beamforming matrix $\mathbf{S}_c^{(g)}$ has non-zero entries where the connection matrix $\mathbf{\Pi}^{(g)}$ has non-zero entries. Similar to PE-AM algorithm, we consider joint optimization of constrained beamformer and compensation matrix. There are two types of partially connected arrays. In the first one, optimization problem is solved by using a fixed connection matrix. The second one considers finding the optimum connection matrix for a given unconstrained beamformer.

1) FIXED SUBARRAY DESIGN

Joint optimization problem of constrained analog beamformer with partially connected array and compensation matrix can be written as

$$\begin{aligned} & \underset{\mathbf{S}_c^{(g)}, \mathbf{S}_{cm}^{(g)}}{\text{minimize}} \left\| \mathbf{S}_{GEB}^{(g)} - \mathbf{S}_c^{(g)} \mathbf{S}_{cm}^{(g)} \right\|_F^2 \\ & \text{subject to } |[\mathbf{S}_c^{(g)}]_{(i,j)}| = \frac{1}{\sqrt{M}} [\mathbf{\Pi}^{(g)}]_{(i,j)}, \quad \forall i, j, \end{aligned} \quad (34)$$

for a given connection matrix $\mathbf{\Pi}^{(g)}$. Note that since there is only one non-zero element in each row of $\mathbf{S}_c^{(g)}$, non-zero element in i^{th} row is multiplied with $j(i)^{\text{th}}$ row of the compensation matrix $\mathbf{S}_{cm}^{(g)}$. By using this property, optimization problem in (34) is simplified as

$$\underset{\{\beta_i^{(g)}\}_{i=1}^M}{\text{minimize}} \left\| [\mathbf{S}_{GEB}^{(g)}]_{(i,:)} - \frac{1}{\sqrt{M}} e^{j\beta_i^{(g)}} [\mathbf{S}_{cm}^{(g)}]_{(j(i),:)} \right\|_F^2 \quad (35)$$

for given compensation matrix $\mathbf{S}_{cm}^{(g)}$, where $\beta_i^{(g)}$ is the phase of non-zero entry in the i^{th} row of $\mathbf{S}_c^{(g)}$. It can be seen that optimization of phase values is decoupled. Hence, optimum phase values $\{\beta_i^{(g)\star}\}_{i=1}^M$ are found as

$$\beta_i^{(g)\star} = \angle \left([\mathbf{S}_{GEB}^{(g)}]_{(i,:)} [\mathbf{S}_{cm}^{(g)}]_{(j(i),:)}^H \right). \quad (36)$$

Given constrained beamformer $\mathbf{S}_c^{(g)}$, optimization problem to find the compensation matrix becomes

$$\underset{\mathbf{S}_{cm}^{(g)}}{\text{minimize}} \left\| \mathbf{S}_{GEB}^{(g)} - \mathbf{S}_c^{(g)} \mathbf{S}_{cm}^{(g)} \right\|_F^2. \quad (37)$$

As there is no constraint on the compensation matrix, optimum solution is the least squares (LS) solution given as

$$\mathbf{S}_{cm}^{(g)\star} = \left([\mathbf{S}_c^{(g)}]^H \mathbf{S}_c^{(g)} \right)^{-1} [\mathbf{S}_c^{(g)}]^H \mathbf{S}_{GEB}^{(g)}. \quad (38)$$

Alternating minimization approach is adopted as in PE-AM algorithm to find constrained analog beamformer and compensation matrix. Summary of the fixed partially connected array design is given in Algorithm 2.

Algorithm 2 Fixed Subarray Design

Input: $\mathbf{S}_{GEB}^{(g)}, \mathbf{\Pi}^{(g)}$

Initialize: Set random phases to $\mathbf{S}_{c,0}^{(g)}$, iteration index $n = 0$

1: **repeat**

2: Fix $\mathbf{S}_{c,n}^{(g)}$

3: Find $\mathbf{S}_{cm,n}^{(g)} = \left([\mathbf{S}_{c,n}^{(g)}]^H \mathbf{S}_{c,n}^{(g)} \right)^{-1} [\mathbf{S}_{c,n}^{(g)}]^H \mathbf{S}_{GEB}^{(g)}$

4: Fix $\mathbf{S}_{cm,n}^{(g)}$, find $\mathbf{S}_{c,n+1}^{(g)}$ by (36)

5: $n \leftarrow n + 1$

6: **until** $\left\| \mathbf{S}_{GEB}^{(g)} - \mathbf{S}_{c,n+1}^{(g)} \mathbf{S}_{cm,n}^{(g)} \right\|_F$ converges

Output: $\mathbf{S}_c^{(g)} = \mathbf{S}_{c,n+1}^{(g)}$ and $\mathbf{S}_{cm}^{(g)} = \mathbf{S}_{cm,n}^{(g)}$

2) DYNAMIC SUBARRAY DESIGN

This section aims to find an algorithm that finds the optimum connection matrix $\mathbf{\Pi}^{(g)}$ for a given GEB. Let $\tilde{\mathbf{S}}_c^{(g)} \in \mathbb{C}^{M \times D_g}$ be a partially connected constrained beamformer whose non-zero entry locations are unknown. This matrix needs to satisfy following constraints

$$|[\tilde{\mathbf{S}}_c^{(g)}]_{(i,j)}| \in \{0, 1\}, \quad \forall i, j, \quad (39a)$$

$$\sum_{j=1}^{D_g} |[\tilde{\mathbf{S}}_c^{(g)}]_{(i,j)}| = 1, \quad \forall i, \quad (39b)$$

$$\sum_{i=1}^M |[\tilde{\mathbf{S}}_c^{(g)}]_{(i,j)}| \geq 1, \quad \forall j, \quad (39c)$$

where the first one is constant-modulus constraint while second one needs to be satisfied to make sure that each antenna is connected to only one RF chain. The third constraint should be satisfied so that each RF chain is connected to at least one antenna. Let the equivalent near-optimal generalized eigenbeamformer be $\mathbf{S}_{GEB}^{(g)} \mathbf{A}^{(g)}$ where $\mathbf{A}^{(g)} \in \mathbb{C}^{D_g \times D_g}$ is an arbitrary unitary matrix. Note that $\mathbf{S}_{GEB}^{(g)} \mathbf{A}^{(g)}$ can replace $\mathbf{S}_{GEB}^{(g)}$ as long as $\mathbf{A}^{(g)}$ is invertible according to Lemma 2. Let us ignore the constraint in (39c) for now and construct an optimization problem given as

$$\underset{\mathbf{A}^{(g)} \in \mathcal{A}, \tilde{\mathbf{S}}_c^{(g)}}{\text{minimize}} \left\| \mathbf{S}_{GEB}^{(g)} \mathbf{A}^{(g)} - \tilde{\mathbf{S}}_c^{(g)} \right\|_F^2$$

$$\text{subject to } |[\tilde{\mathbf{S}}_c^{(g)}]_{(i,j)}| \in \{0, 1\}, \quad \forall i, j;$$

$$\sum_{j=1}^{D_g} |[\tilde{\mathbf{S}}_c^{(g)}]_{(i,j)}| = 1, \quad \forall i. \quad (40)$$

where \mathcal{A} is the set of unitary matrices which are invertible. Given $\mathbf{A}^{(g)}$, entries of optimal constrained analog beamformer can be found as

$$[\tilde{\mathbf{S}}_c^{(g)\star}]_{(i,j)} = \begin{cases} e^{j\angle([\mathbf{S}_{GEB}^{(g)} \mathbf{A}^{(g)}]_{(i,j)})}, & \text{if } j = j^\star(i) \\ 0, & \text{otherwise,} \end{cases} \quad (41)$$

where $j^\star(i) = \arg \max_{1 \leq j \leq D_g} |[\mathbf{S}_{GEB}^{(g)} \mathbf{A}^{(g)}]_{(i,j)}|$ for $i = 1, 2, \dots, M$ [26]. For given $\tilde{\mathbf{S}}_c^{(g)}$, optimization problem in (40)

reduces to orthogonal Procrustes problem expressed as

$$\underset{\mathbf{A}^{(g)} \in \mathcal{A}}{\text{minimize}} \left\| \mathbf{S}_{GEB}^{(g)} \mathbf{A}^{(g)} - \tilde{\mathbf{S}}_c^{(g)} \right\|_F^2, \quad (42)$$

which has the same solution as in the PE-AM algorithm. Optimum unitary matrix is found as

$$\mathbf{A}^{(g)*} = \mathbf{V}^{(g)} [\mathbf{U}^{(g)}]^H, \quad (43)$$

where the following SVD is used: $[\mathbf{S}_{GEB}^{(g)}]^H \tilde{\mathbf{S}}_c^{(g)} = \mathbf{U}^{(g)} \boldsymbol{\Sigma}^{(g)} [\mathbf{V}^{(g)}]^H$. Similar to previous algorithms, alternating minimization is applied to find partially connected constrained beamformer which is summarized in Algorithm 3.

Algorithm 3 Dynamic Connection Design

Input: $\mathbf{S}_{GEB}^{(g)}$

Initialize: Set random phases to $\tilde{\mathbf{S}}_{c,0}^{(g)}$, iteration index $n = 0$

- 1: **repeat**
- 2: Fix $\tilde{\mathbf{S}}_{c,n}^{(g)}$, find $[\mathbf{S}_{GEB}^{(g)}]^H \tilde{\mathbf{S}}_{c,n}^{(g)} = \mathbf{U}_n^{(g)} \boldsymbol{\Sigma}_n^{(g)} [\mathbf{V}_n^{(g)}]^H$
- 3: $\mathbf{A}_n^{(g)} = \mathbf{V}_n^{(g)} [\mathbf{U}_n^{(g)}]^H$
- 4: Fix $\mathbf{A}_n^{(g)}$, find $\tilde{\mathbf{S}}_{c,n+1}^{(g)}$ by (41)
- 5: $n \leftarrow n + 1$
- 6: **until** $\left\| \mathbf{S}_{GEB}^{(g)} \mathbf{A}_n^{(g)} - \tilde{\mathbf{S}}_{c,n+1}^{(g)} \right\|_F$ converges

Output: $\tilde{\mathbf{S}}_c^{(g)} = \tilde{\mathbf{S}}_{c,n+1}^{(g)}$

Remark 1: The result of the Algorithm 3, which solves a non-convex problem, depends on the initial random phases which leads to different antenna connections for each result. Hence, this algorithm should be repeated several times with different initial phases and the result yielding maximum expected SINR for group g in reduced dimension, which is the SINR after statistical pre-beamformer $\mathbf{S}^{(g)}$ (i.e., just before digital beamforming), should be used. Expression for the expected SINR is given in (68) in Appendix D, where we denote the expected SINR of group g by $\overline{\text{SINR}}^{(g)}$.

Remark 2: Constraint (39c) is not considered in Algorithm 3 which is repeated several times as stated in Remark 1. Optimum connection matrix should be selected from the results where there is a connection between any RF chain and at least one antenna to satisfy this constraint.

It is important to note that only constrained analog beamformers can be found by using Algorithm 3. However, we need to find a compensation matrix to improve the result. We propose that result of Algorithm 3 should be used in order to find the connection matrix as $\boldsymbol{\Pi}^{(g)} = |\tilde{\mathbf{S}}_c^{(g)}|$. Then, this connection matrix should be used as an input to Algorithm 2 which jointly optimizes constrained analog beamformer and compensation matrix. Furthermore, initial phases of constrained analog beamformer for Algorithm 2 is selected as the phases of $\tilde{\mathbf{S}}_c^{(g)}$. Therefore, it can be seen that Algorithm 3 is used to find the connection matrix. Summary of the overall dynamic subarray design is given in Algorithm 4 where N_{iter} is the number of times Algorithm 3 is repeated.

Algorithm 4 Dynamic Subarray Design

Input: $\mathbf{S}_{GEB}^{(g)}, N_{iter}$

Initialize: Iteration index $t = 0$

- 1: **repeat**
- 2: $t \leftarrow t + 1$
- 3: Run Algorithm 3 to find $\tilde{\mathbf{S}}_c^{(g)}$ and set $\tilde{\mathbf{S}}_{c,t}^{(g)} \leftarrow \tilde{\mathbf{S}}_c^{(g)}$
- 4: **if** $\sum_{i=1}^M |[\tilde{\mathbf{S}}_{c,t}^{(g)}]_{(i,j)}| \geq 1, \forall j$ **then**
- 5: Calculate $\overline{\text{SINR}}_t^{(g)}$ by setting $\mathbf{S}^{(g)} \leftarrow \tilde{\mathbf{S}}_{c,t}^{(g)}$ in (68)
- 6: **else**
- 7: Set $\overline{\text{SINR}}_t^{(g)}$ to 0
- 8: **end if**
- 9: **until** $t = N_{iter}$
- 10: $t^* = \arg \max_{1 \leq t \leq N_{iter}} \overline{\text{SINR}}_t^{(g)}$
- 11: Run Algorithm 2 with $\boldsymbol{\Pi}^{(g)} = |\tilde{\mathbf{S}}_{c,t^*}^{(g)}|$ and initial phases of $\tilde{\mathbf{S}}_{c,t^*}^{(g)}$ to find $\mathbf{S}_c^{(g)}$ and $\mathbf{S}_{cm}^{(g)}$.

Output: $\mathbf{S}_c^{(g)}$ and $\mathbf{S}_{cm}^{(g)}$

C. CONVERGENCE AND COMPLEXITY ANALYSIS

1) CONVERGENCE ANALYSIS

Alternating minimization approach is employed to obtain both fully and partially connected constrained analog beamformers. Convergence of given algorithms is guaranteed since cost functions, which are bounded below, are monotonically decreasing at each iteration. Considering these two facts, algorithms given in this paper converge to local minima [5], [26].

2) COMPLEXITY ANALYSIS

Constrained beamformer design algorithms that are described in this section are based on GEB found by generalized eigenvalue problem in (16) where matrices with size $M \times M$ are involved. In the literature, Lanczos algorithm and its variations are used to find a few most dominant generalized eigenvectors [46]. Although the exact computational complexity of this algorithm is not clear due to the iterative solutions, its complexity can roughly be taken as $O(M^3)$.

On the other hand, alternating minimization based algorithms require additional computation over GEB. Phase extraction has the same complexity order with GEB as it only requires the phases of GEB. Algorithm 1 and Algorithm 3 have the same solution steps which involve phase extraction of matrix multiplication at one step. Another step includes SVD of a matrix with size $D_g \times D_g$ which has a complexity order of $O(D_g^3)$. Since these computations are repeated several times, complexity of PE-AM and dynamic connection design becomes $O(n_{max} D_g^3)$ where n_{max} is the number of iterations until the convergence. Algorithm 2 only includes phase extraction from a vector multiplication and a LS solution which requires the inverse of a $D_g \times D_g$ matrix with complexity order of $O(D_g^3)$. Complexity of fixed subarray design becomes $O(n_{max} D_g^3)$ due to the alternating minimization method. Algorithm 4, which is used to find the dynamic constrained analog beamformers, involves execution of

TABLE 2. Computational complexity orders of analog beamforming algorithms.

Beamformer Type	Computational Complexity
GEB	$O(M^3)$
PE	$O(M^3)$
PE-AM	$O(M^3 + n_{max}D_g^3)$
Fixed Subarray	$O(M^3 + n_{max}D_g^3)$
Dynamic Subarray	$O(M^3 + (N_{iter} + 1)n_{max}D_g^3)$

Algorithm 2 once and Algorithm 3 N_{iter} times. Therefore, computational complexity of dynamic subarray design becomes $O((N_{iter} + 1)n_{max}D_g^3)$. It can be understood that complexity orders of proposed alternating minimization-based algorithms are not high as number of RF chains D_g is limited in hybrid systems. Moreover, recall that unconstrained analog beamformers are designed based on slowly varying CCMs. Hence, constrained beamformers and dynamic connections are updated infrequently. Computational complexity orders of proposed unconstrained and constrained analog beamformer design algorithms are summarized in Table 2.

V. PERFORMANCE MEASURES

In order to show the effectiveness of proposed constrained design, we should define some performance measures. We firstly introduce ergodic capacity which is the most commonly used performance measure. Then, we describe a channel estimation method in reduced dimension and define the normalized mean square error (nMSE) that shows the accuracy of channel estimation.

A. ERGODIC CAPACITY

Ergodic capacity of user m in group g at the output of i^{th} equalizer iteration is defined as

$$C_i^{(g_m)} = \mathbb{E} \left\{ \left[\mathbf{H}_{eff,l}^{(g,g)} \right]_{l=0}^{L-1} \left\{ \log_2 \left(1 + SINR_i^{(g_m)} \right) \right\} \right\}, \quad (44)$$

where the expectation is taken over different instantaneous channel realizations while keeping the analog beamformer constant, i.e., user locations and CCMs are kept constant. Output SINR of user m in group g at the output of i^{th} equalizer iteration is denoted by $SINR_i^{(g_m)}$. Let the Bussgang decomposition of the soft output for decoding transmitted symbols given in (10) is expressed as [47]

$$\hat{x}_n^{(g_m),i} = a^{(g_m),i} x_n^{(g_m)} + b_n^{(g_m),i}, \quad (45)$$

for $n = 0, 1, \dots, N - 1$ and $m = 1, 2, \dots, K_g$, where $a^{(g_m),i}$ and $b_n^{(g_m),i}$ are the complex amplitude and residual interference of the transmitted symbol of user m in group g , respectively. Then the output SINR of this user for given $\left\{ \mathbf{H}_{eff,l}^{(g,g)} \right\}_{l=0}^{L-1}$ is calculated as

$$SINR_i^{(g_m)} = \frac{|a^{(g_m),i}|^2 \mathbb{E} \{ |x_n^{(g_m)}|^2 \}}{\mathbb{E} \{ |b_n^{(g_m),i}|^2 \}} = \frac{E_s^{(g)}}{K_g} \frac{|a^{(g_m),i}|^2}{\mathbb{E} \{ |b_n^{(g_m),i}|^2 \}}. \quad (46)$$

By using (7), (9) and (10), complex amplitude can be computed as

$$\begin{aligned} a^{(g_m),i} &= \mathbb{E} \{ \hat{x}_n^{(g_m),i} (x_n^{(g_m)})^* \} / \mathbb{E} \{ |x_n^{(g_m)}|^2 \} \\ &= \frac{1}{N} \sum_{k=0}^{N-1} [\mathbf{W}_k^{(g,i)}]_{(:,m)}^H \mathbf{\Lambda}_{eff,k}^{(g,g)} \mathbf{e}_m, \end{aligned} \quad (47)$$

where $\mathbf{e}_m \in \mathbb{Z}^{K_g \times 1}$ is a binary vector whose m^{th} entry is one whereas others are zero. On the other hand, residual interference power can be calculated as $\mathbb{E} \{ |b_n^{(g_m),i}|^2 \} = \mathbb{E} \{ |\hat{x}_n^{(g_m),i}|^2 \} - \frac{E_s^{(g)}}{K_g} |a^{(g_m),i}|^2$ where

$$\begin{aligned} &\mathbb{E} \{ |\hat{x}_n^{(g_m),i}|^2 \} \\ &= \frac{1}{N} \sum_{k=0}^{N-1} [\mathbf{W}_k^{(g,i)}]_{(:,m)}^H \mathbf{R}_{\tilde{\mathbf{y}}_{f,k}^{(g)}} [\mathbf{W}_k^{(g,i)}]_{(:,m)} \\ &\quad + \frac{1}{N} \sum_{k=0}^{N-1} [\mathbf{C}_k^{(g,i)}]_{(:,m)}^H \mathbf{P}_i^{(g)} [\mathbf{C}_k^{(g,i)}]_{(:,m)} \\ &\quad - \frac{2}{N} \sum_{k=0}^{N-1} \text{Re} \left\{ [\mathbf{W}_k^{(g,i)}]_{(:,m)}^H \mathbf{\Lambda}_{eff,k}^{(g,g)} \mathbf{P}_i^{(g)} [\mathbf{C}_k^{(g,i)}]_{(:,m)} \right\}, \end{aligned} \quad (48)$$

which can be obtained by using Parseval's relation for (9) and (10). Covariance matrix of $\tilde{\mathbf{y}}_{f,k}^{(g)}$ given in (20) is used for the calculation above. Since $a^{(g_m),i}$ and $\mathbb{E} \{ |b_n^{(g_m),i}|^2 \}$ do not depend on the time instance n , they can be calculated once for given instantaneous effective channel matrices.

1) ASYMPTOTIC ANALYSIS

In IB-DFE framework, estimates are expected to be more reliable as the number of iterations increases. Asymptotically, if perfect knowledge of intra-group user symbols are obtained, reliability matrices approach to the maximum reliability case that can be expressed as $\mathbf{P}^{(g)} = \frac{E_s^{(g)}}{K_g} \mathbf{I}_{K_g}$. In Appendix E, asymptotic SINR of user m in group g is obtained as

$$SINR^{(g_m)} = \frac{1}{N} \sum_{k=0}^{N-1} \frac{E_s^{(g)}}{K_g} [\mathbf{\Lambda}_{eff,k}^{(g,g)}]_{(:,m)}^H \mathbf{R}_{\tilde{\mathbf{y}}_{f,k}^{(g)}}^{-1} [\mathbf{\Lambda}_{eff,k}^{(g,g)}]_{(:,m)}. \quad (49)$$

It can be understood that the equalizer at each frequency bin becomes a whitening matched filter after perfect intra-group cancellation in frequency domain. Therefore, asymptotic SINR in (49) is called the matched filter bound (MFB) which is the maximum SINR that can be achieved by utilizing IB-DFE.

B. BEAMSPACE-AWARE CHANNEL ESTIMATION

In this paper, channel estimation is realized in reduced dimension. Instantaneous intra-group effective channel matrices $\left\{ \mathbf{H}_{eff,l}^{(g,g)} \right\}_{l=0}^{L-1}$ of each group are estimated during the uplink channel estimation phase in TDD mode where distinct pilot sequences with length T are assigned to each user in a group. Equivalent signal vector of group g at the BS at the end of the

pilot sequence transmission is defined as

$$\begin{aligned} \bar{\mathbf{y}}^{(g)} &\triangleq \left[[\tilde{\mathbf{y}}_0^{(g)}]^H, [\tilde{\mathbf{y}}_1^{(g)}]^H, \dots, [\tilde{\mathbf{y}}_{T-1}^{(g)}]^H \right]^H \\ &= \sum_{g'=1}^G \left(\mathbf{X}^{(g')} \otimes \mathbf{I}_{D_g} \right) \bar{\mathbf{h}}_{eff}^{(g,g')} + \left(\mathbf{I}_T \otimes [\mathbf{S}^{(g)}]^H \right) \bar{\mathbf{n}}, \end{aligned} \quad (50)$$

where $\mathbf{X}^{(g)} \in \mathbb{C}^{T \times K_g L}$ is the equivalent pilot matrix of group g and $\bar{\mathbf{h}}_{eff}^{(g,g')} \in \mathbb{C}^{D_g K_g L \times 1}$ is the overall effective channel vector of group g' after the analog beamformer of group g . Note that equivalent signal vector $\tilde{\mathbf{y}}^{(g)} \in \mathbb{C}^{TD_g \times 1}$ is obtained by vertically concatenating received signal vectors after the analog beamformer $\{\tilde{\mathbf{y}}_n^{(g)}\}_{n=0}^{T-1}$ in (6). Similarly, equivalent noise vector $\bar{\mathbf{n}} \in \mathbb{C}^{TM \times 1}$ is constructed by vertically concatenating noise vectors $\{\mathbf{n}_n\}_{n=0}^{T-1}$. Equivalent pilot matrix is defined as

$$\mathbf{X}^{(g)} \triangleq \left[\mathbf{X}^{(g_1)}, \mathbf{X}^{(g_2)}, \dots, \mathbf{X}^{(g_{K_g})} \right], \quad (51)$$

where $[\mathbf{X}^{(g_m)}]_{(i,j)} \triangleq x_{i-j}^{(g_m)}$, for $i = 0, 1, \dots, T - 1$ and $j = 0, 1, \dots, L - 1$. Pilot sequence entries of user m in group g are denoted by $x_n^{(g_m)}$. Note that entries with negative indices can be taken as cyclic prefix or previously decoded symbols. Overall effective channel vector of group g' for the analog beamformer of group g is defined as

$$\bar{\mathbf{h}}_{eff}^{(g,g')} \triangleq \left[[\bar{\mathbf{h}}_{eff}^{(g,g'_1)}]^H, [\bar{\mathbf{h}}_{eff}^{(g,g'_2)}]^H, \dots, [\bar{\mathbf{h}}_{eff}^{(g,g'_{K_{g'}})}]^H \right]^H, \quad (52)$$

$$\bar{\mathbf{h}}_{eff}^{(g,g'_m)} \triangleq \left[[\mathbf{h}_{eff,0}^{(g,g'_m)}]^H, [\mathbf{h}_{eff,1}^{(g,g'_m)}]^H, \dots, [\mathbf{h}_{eff,L-1}^{(g,g'_m)}]^H \right]^H, \quad (53)$$

where $\mathbf{h}_{eff,l}^{(g,g'_m)} \triangleq [\mathbf{S}^{(g)}]^H \mathbf{h}_l^{(g'_m)}$, or equivalently $\mathbf{h}_{eff,l}^{(g,g'_m)} \triangleq [\mathbf{H}_{eff,l}^{(g,g'_m)}]_{(:,m)}$, for $l = 0, 1, \dots, L - 1$. It is important to note that we are only interested in intra-group channel vectors. In other words, our aim is to estimate $\bar{\mathbf{h}}_{eff}^{(g,g)}$ from (50) as

$$\hat{\bar{\mathbf{h}}}_{eff}^{(g,g)} = [\mathbf{Z}^{(g)}]^H \bar{\mathbf{y}}^{(g)} \quad (54)$$

where $\mathbf{Z}^{(g)} \in \mathbb{C}^{D_g K_g L \times TD_g}$ is the channel estimator matrix. In this paper, LMMSE and LS channel estimators are considered. Expressions for these estimators can be found in Appendix F. In order to show the accuracy of channel estimation, we calculate nMSE which is defined as

$$\begin{aligned} nMSE^{(g)} &\triangleq \frac{\mathbb{E} \left\{ \left\| \bar{\mathbf{h}}_{eff}^{(g,g)} - \hat{\bar{\mathbf{h}}}_{eff}^{(g,g)} \right\|^2 \right\}}{\mathbb{E} \left\{ \left\| \bar{\mathbf{h}}_{eff}^{(g,g)} \right\|^2 \right\}} \\ &= \frac{\text{Tr} \left\{ \mathbf{R}_{\bar{\mathbf{h}}_{eff}^{(g,g)}} \right\} + \text{Tr} \left\{ \mathbf{R}_{\hat{\bar{\mathbf{h}}}_{eff}^{(g,g)}} \right\} - 2\text{Re} \left\{ \text{Tr} \left\{ \mathbf{R}_{\bar{\mathbf{h}}_{eff}^{(g,g)} \hat{\bar{\mathbf{h}}}_{eff}^{(g,g)}} \right\} \right\}}{\text{Tr} \left\{ \mathbf{R}_{\bar{\mathbf{h}}_{eff}^{(g,g)}} \right\}} \end{aligned} \quad (55)$$

where covariance matrix of $\hat{\bar{\mathbf{h}}}_{eff}^{(g,g)}$ and covariance between $\hat{\bar{\mathbf{h}}}_{eff}^{(g,g)}$ and $\bar{\mathbf{h}}_{eff}^{(g,g)}$ can be computed as

$$\mathbf{R}_{\hat{\bar{\mathbf{h}}}_{eff}^{(g,g)}} = \mathbb{E} \left\{ \hat{\bar{\mathbf{h}}}_{eff}^{(g,g)} [\hat{\bar{\mathbf{h}}}_{eff}^{(g,g)}]^H \right\} = [\mathbf{Z}^{(g)}]^H \mathbf{R}_{\bar{\mathbf{y}}^{(g)}} \mathbf{Z}^{(g)}, \quad (56)$$

$$\mathbf{R}_{\bar{\mathbf{h}}_{eff}^{(g,g)} \hat{\bar{\mathbf{h}}}_{eff}^{(g,g)}} = \mathbb{E} \left\{ \hat{\bar{\mathbf{h}}}_{eff}^{(g,g)} [\bar{\mathbf{h}}_{eff}^{(g,g)}]^H \right\} = [\mathbf{Z}^{(g)}]^H \mathbf{R}_{\bar{\mathbf{y}}^{(g)} \bar{\mathbf{h}}_{eff}^{(g,g)}}. \quad (57)$$

Expressions of $\mathbf{R}_{\bar{\mathbf{y}}^{(g)} \bar{\mathbf{h}}_{eff}^{(g,g)}}$, $\mathbf{R}_{\bar{\mathbf{y}}^{(g)}}$ and $\mathbf{R}_{\bar{\mathbf{h}}_{eff}^{(g,g)}}$ are given in (73), (74) and (75) in Appendix F, respectively. There are some concluding remarks for the described channel estimation scheme.

Remark 3: With the assumption that analog beamformer suppresses inter-group interference perfectly, training periods of groups do not have to be synchronized. In this case, training symbols of other groups ($\mathbf{X}^{(g')}$, $g' \neq g$) can be assumed to be random. Furthermore, users in a group should use different training sequences whereas users in other groups can reuse the same sequences with properly designed analog beamformers. Hence, it can be inferred that this channel estimation scheme is robust against pilot contamination.

Remark 4: Analog beamforming method proposed in this paper was shown to be applicable to multiantenna user case in Lemma 1. Similarly, channel estimation scheme described in this subsection can be extended to multiantenna user case by redefining effective channels as in (67) in Appendix B.

VI. NUMERICAL RESULTS

In this section, proposed constrained analog beamformers are compared by using the performance measures in Section V. BS has an ULA with $M = 128$ antennas whereas users have single-antenna elements. Carrier frequency is selected as 30GHz while the antenna spacing of the ULA is set to half the wavelength. A predetermined user-grouping is employed with $G = 4$ groups where each group has $K_g = 2$ users.⁶ Although total number of delays is $L = 32$, users do not have active MPCs at every delay. In the simulations, angular spread $\Delta_l^{(g_m)}$ of all users is set to 2° for every MPC. Power profile function $\rho_l^{(g_m)}(\theta)$ is assumed to be uniform in the angular spread of the MPC and channel gain $\sqrt{\gamma^{(g_m)}}$ is set to 1 for all users. MPC cluster powers of users are assumed to be the same. Signal-to-noise ratio (SNR) of group g is denoted by $SNR^{(g)} = \frac{E_s^{(g)}}{N_0}$.

The scenario used for simulations is given in Table 3. This table shows the active MPC indices of groups and mean AoAs of users related to each active MPC. Group-1 is selected as the mobile group with shifting angle of ϕ which is swept from -45° to 45° with increments of 0.1° for simulation purposes. It is important to note that CCMs are slowly varying parameters. However, the reason to use a mobile group is to obtain results for different angular profiles and observe the effect of overlapping MPCs. Hence, we are only interested in the results of the mobile group which is Group-1. For each ϕ , CCMs and analog beamformer of Group-1 are calculated, then ergodic capacity, BER, and nMSE are obtained. Ergodic capacity of user m in Group-1 and nMSE of Group-1 corresponding to shifting angle ϕ are denoted by $C_\phi^{(1_m)}$ and $nMSE_\phi^{(1)}$, respectively. Ergodic capacity and BER results are

⁶Here, we assume that users come in groups, either by nature or by the use of a proper user grouping algorithm [35], [36].

TABLE 3. Angle-delay profile of groups.

Group	MPC index	Mean AoAs ($\mu_l^{(g_m)}$) of users
1 (Mobile)	0	$\{\phi-15.5^\circ, \phi-14.5^\circ\}$
	5	$\{\phi-2.5^\circ, \phi-1.5^\circ\}$
	11	$\{\phi+16.5^\circ, \phi+17.5^\circ\}$
2	3	$\{40.5^\circ, 41.5^\circ\}$
	9	$\{20.5^\circ, 21.5^\circ\}$
3	8	$\{-10.5^\circ, -9.5^\circ\}$
	17	$\{-20.5^\circ, -19.5^\circ\}$
4	29	$\{-40.5^\circ, -39.5^\circ\}$

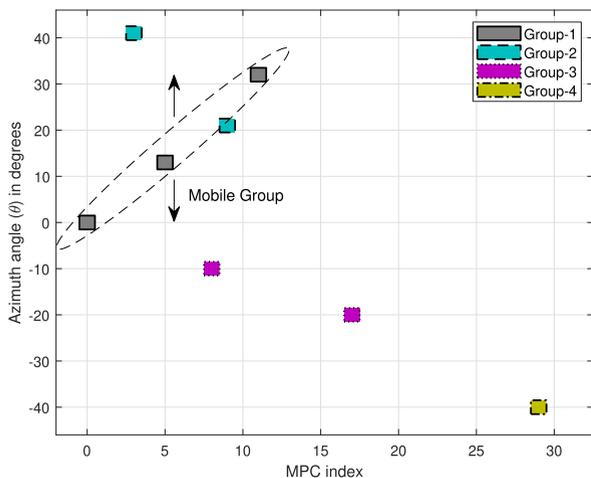


FIGURE 2. Angle-delay map of groups with $\phi = 15^\circ$.

obtained with perfect CSI knowledge. Cardinality of the input alphabet is selected as 4 (i.e., QPSK) for BER simulations, whereas Gaussian alphabets are assumed for ergodic capacity calculations. Therefore, results related to ergodic capacity are shown for ZF and LMMSE type digital beamformers.

Angle-delay map of groups is given in Fig. 2 in order to visualize the scenario given in Table 3. In this figure, ϕ of the mobile group is set to 15° . This is a typical mm-wave scenario where we observe a sparse angle-delay profile.

A. JSJM WITH FULLY CONNECTED ARRAYS

In this section, constrained fully connected array structures are compared. Fully digital (FD) beamformer and unconstrained near-optimal analog beamformer GEB introduced in Section III-A are used as benchmark. In order to make a fair comparison, FD is applied with the perfect knowledge of instantaneous effective intra-group CSI and inter-group CCMs. DFT beamformer, phase extraction and PE-AM in Algorithm 1 are the considered constrained beamformers.

Firstly, we will introduce beampattern metric which shows the power that can be attained at the angle of interest θ for a given analog beamformer. Interference suppression capability of beamformers can be observed by using this metric. Beampattern as a function of θ for given analog beamformer $\mathbf{S}^{(g)}$ can be defined as

$$B(\theta) = \mathbf{u}(\theta)^H \mathbf{S}^{(g)} \left([\mathbf{S}^{(g)}]^H \mathbf{S}^{(g)} \right)^{-1} [\mathbf{S}^{(g)}]^H \mathbf{u}(\theta). \quad (58)$$

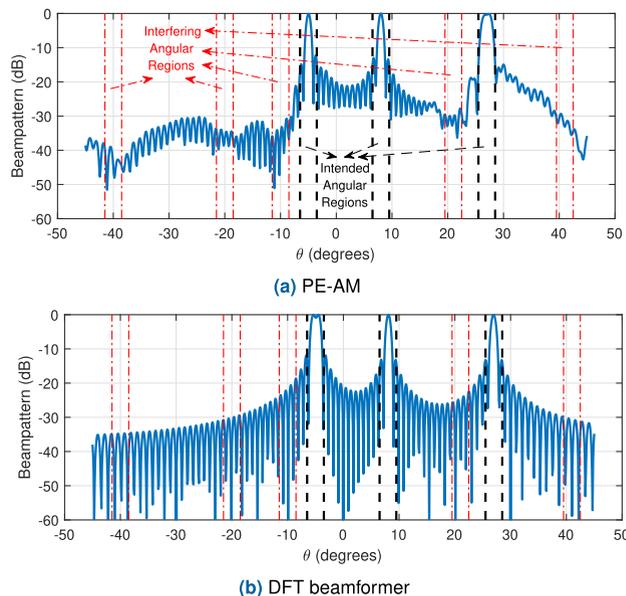


FIGURE 3. Beampatterns of Group-1 for fully connected array structures with $\phi = 10^\circ$, $D_1 = 4$ and $\text{SNR}^{(g)} = 40$ dB for $\forall g$.

Fig. 3 shows the beampatterns of Group-1 for PE-AM and DFT beamformer with $\phi = 10^\circ$ and $D_1 = 4$ where SNR of each group is set to 40 dB. It is seen that both beamformers can form narrow beams. Although PE-AM does not provide deep nulls, we observed that it can provide moderate suppression for MPCs of other groups whereas DFT beamformer does not consider these MPCs. If beampatterns of PE-AM and DFT beamformer are compared, this suppression appears more significant for interfering angular regions that are close to the angular region of intended group.

Fig. 4 shows the average spectral efficiency of Group-1 with changing SNR where SNR of other groups is set to $\text{SNR}^{(g')} = 40$ dB for $g' \neq 1$. Note that average spectral efficiency is found by taking the average of ergodic capacity over both users and shifting angle ϕ , i.e., $C_{avg}^{(1)} = \mathbb{E}_{\phi, m} \{ C_{\phi}^{(1m)} \}$. It can be observed that LMMSE type digital beamformer does not provide a significant improvement over ZF for analog beamformers except DFT beamformer with $D_1 = 2$ since it does not consider inter-group interference. Similarly, there is a significant gap between ZF and LMMSE type beamformers for FD architecture as it does not have any inter-group interference suppression with ZF. However, LMMSE provides improvement for all constrained analog beamformers, especially for DFT beamformer, with $D_1 = 4$ as it can efficiently suppress residual inter-group interference when the number of RF chains is greater than the number of users. It should be noted that LMMSE and ZF type digital beamformers perform the same for GEB at high SNR which shows that there is no residual interference. Furthermore, it is seen that there is a significant gap between DFT beamformer and other constrained beamformers for $D_1 = 2$, whereas the gap between PE and PE-AM is very small. However, there is a gap between PE and PE-AM for $D_1 = 4$ which is expected

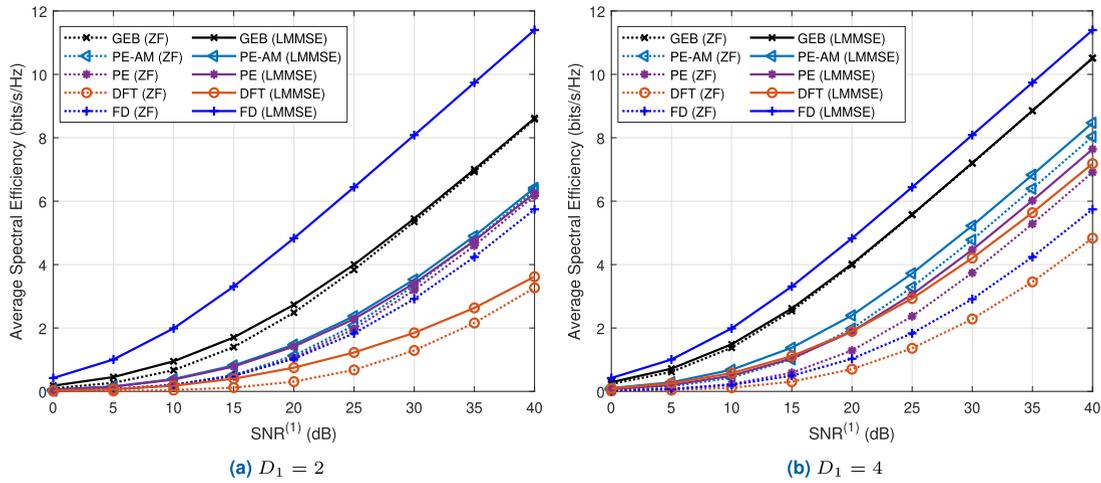


FIGURE 4. Average spectral efficiency of Group-1 vs. SNR for fully connected array structures with $SNR(g') = 40$ dB for $g' \neq 1$.

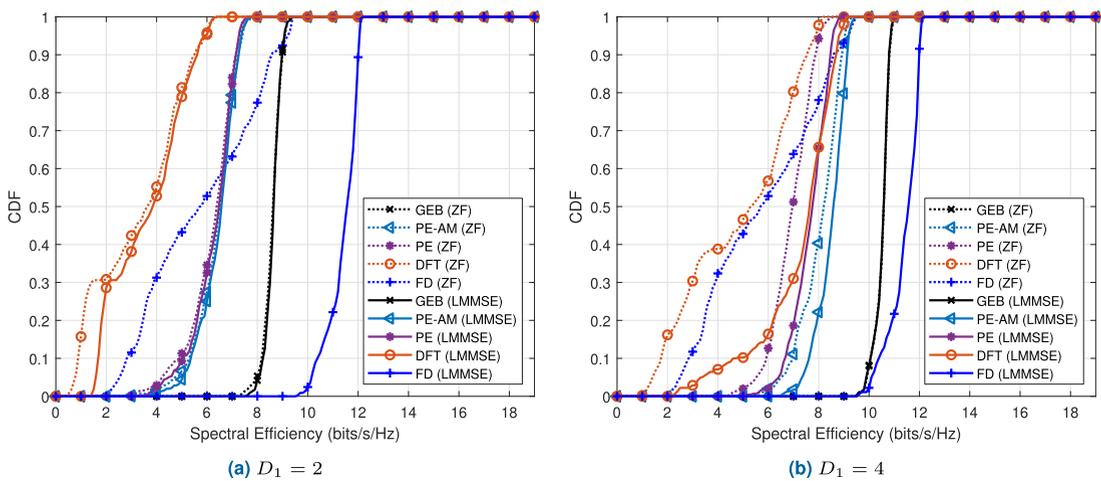
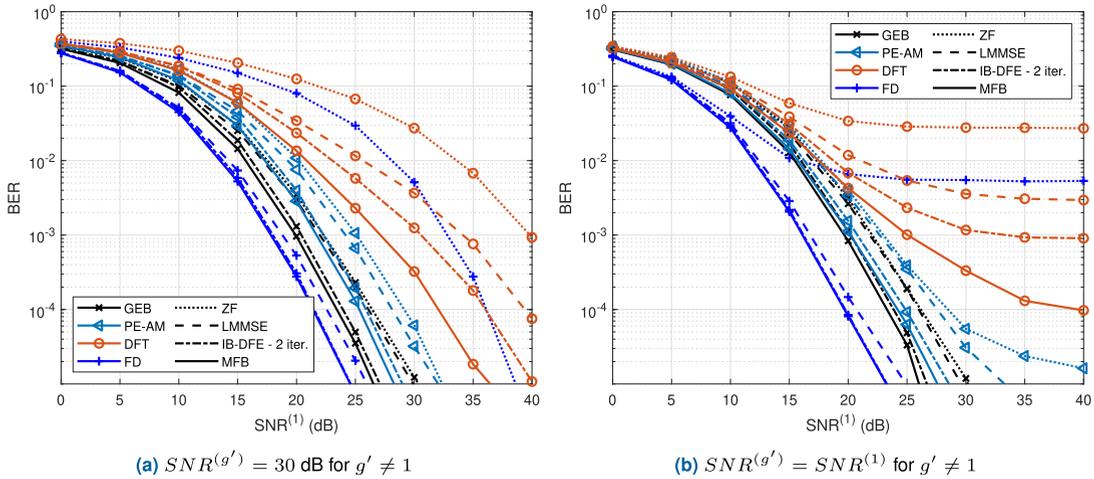
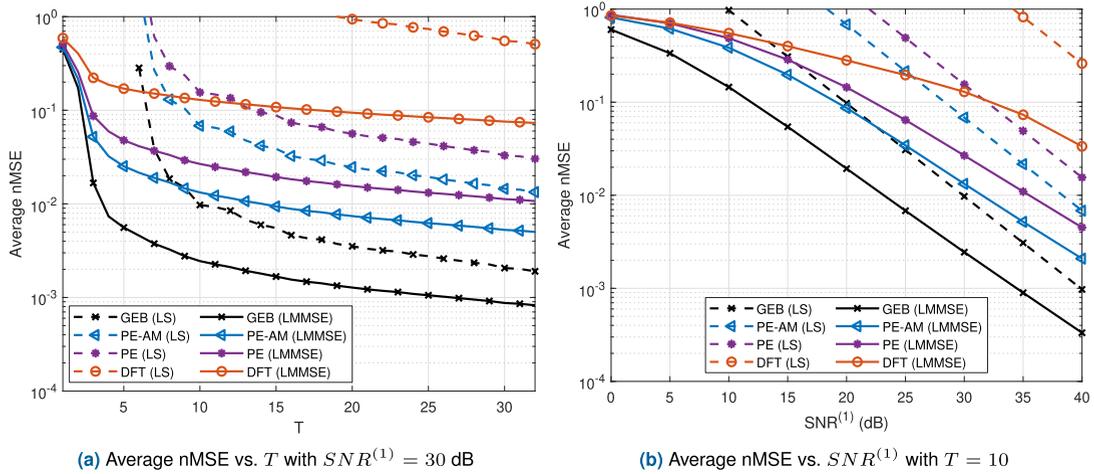


FIGURE 5. CDF of spectral efficiency of Group-1 for fully connected array structures with $SNR(g) = 40$ dB for v_g .

since increase in RF chains corresponds to increase in the dimension of the compensation matrix for PE-AM. LMMSE type digital beamformer makes the DFT beamformer perform almost as good as PE for $D_1 = 4$. Moreover, the gap between FD beamformer and GEB is small for $D_1 = 4$, which shows that GEB can achieve the capacity of optimal FD architecture with limited RF chains.

Previous results show the average performance of analog beamformers in the mobile group. However, angular regions of MPCs belonging to Group-1 overlap with angular regions of MPCs of other groups at certain ϕ values. It is important to observe the degradation in spectral efficiency with different constrained analog beamformers for overlapping situations. In other words, we need to observe the variation in spectral efficiency to obtain the outage capacity. Let the cumulative density function (CDF) of spectral efficiency of Group-1 is defined as $P(C_\phi^{(g)} < c)$ where average spectral efficiency of Group-1 for shifting angle ϕ is denoted by

$C_\phi^{(1)} = \mathbb{E}_m \{C_\phi^{(1_m)}\}$. Then, CDF curves correspond to outage probabilities for given spectral efficiency values. In Fig. 5, CDF curves with different analog beamformers are compared for $D_1 = 2$ and $D_1 = 4$ where SNR of each group is set to 40 dB. It is observed that CDF curves of DFT beamformer are wider compared to others which results in higher outage probability. For example, outage probability of PE-AM with $D_1 = 4$ is close to zero at 7 bits/s/Hz while it is above 0.3 for DFT beamformer. On the other hand, GEB has sharp CDF curves for both $D_1 = 2$ and $D_1 = 4$ cases which means that it is a robust analog beamforming method. If CDF curves of GEB and FD beamformer are compared, it can be observed that GEB (even with a significantly reduced number of RF chains) performs very close to FD at low outage probability levels. Furthermore, it is observed that PE and PE-AM have sharp CDF curves which are overlapped for $D_1 = 2$ as the size of compensation matrix depends on D_1 .


FIGURE 6. Average BER of Group-1 vs. SNR for fully connected array structures with $D_1 = 4$.

FIGURE 7. Average nMSE of Group-1 for fully connected array structures with $D_1 = 4$ and $SNR^{(g')} = 40 \text{ dB}$ for $g' \neq 1$.

Another performance measure is BER, which can be used to observe the contributions of IB-DFE method over linear equalizers. We considered two different scenarios for BER results. Firstly, we obtained average BER of Group-1 with changing SNR for $D_1 = 4$ while keeping SNR of other groups at 30 dB, which is shown in Fig. 6a where MFB denotes the asymptotic case with perfect intra-group cancellation. It can be observed that IB-DFE iterations improve the BER of all beamformers compared to linear equalizers, and BER curves of IB-DFE with 2 iterations are very close to MFB curves for FD beamformer, GEB and PE-AM. However, there is still SNR gap between BER curves of MFB and IB-DFE for DFT beamformer. Furthermore, it can be observed that FD beamformer, GEB and PE-AM perform very close to each other, whereas there is 10 dB SNR gap between PE-AM and DFT beamformer with IB-DFE at BER of 10^{-3} . The second scenario is the case where SNR of every group is varied together, which is shown in Fig. 6b. It can be observed that DFT beamformer has an error floor even for MFB case, which is expected since perfect intra-group cancellation is

utilized with IB-DFE, but DFT beamformer does not consider inter-group interference. Similarly, FD beamformer with ZF does not suppress inter-group interference leading to an error floor. Although PE-AM has an error floor with ZF type digital beamformer, it disappears with IB-DFE iterations. Furthermore, it can be seen that GEB and PE-AM perform very close to each other, which shows that PE-AM preserves the interference suppression capability of GEB. On the other hand, BER results show that interference sensitivity of DFT beamformer could be detrimental although this could not be observed from average spectral efficiency results.

Channel estimation accuracy is another performance measure that is considered in this paper. Fig. 7 shows the average nMSE of Group-1 with $D_1 = 4$ where SNR of other groups is set to $SNR^{(g')} = 40 \text{ dB}$ for $g' \neq 1$. Average of nMSE is taken over shifting angle ϕ , i.e., $nMSE_{avg}^{(1)} = \mathbb{E}_{\phi} \{nMSE_{\phi}^{(1)}\}$. It can be observed that LMMSE estimator outperforms LS estimator especially with short pilot lengths for all analog beamformers. Moreover, it is important to note that nMSE

of DFT beamformer is much higher compared to others even with LMMSE estimator while nMSE of PE-AM is less than nMSE of both PE and DFT beamformer.

In spectral efficiency analysis, DFT beamformer performed close to PE for $D_1 = 4$ with perfect CSI. However, nMSE results show that spectral efficiency performance of DFT would degrade more compared to other ones if the channel estimation errors are considered. Considering the beam-pattern, outage capacity, BER and nMSE results, it can be concluded that DFT beamformer should be avoided for JSDM framework especially when the near-far effect is observed as it does not consider MPCs other groups.⁷ Furthermore, PE-AM yields better performance than PE especially when the number of RF chains is not highly restricted since dimensions of compensation matrix are determined by this number. Consequently, PE-AM should be preferred to obtain fully connected constrained analog beamformers.

B. JSDM WITH PARTIALLY CONNECTED ARRAYS

In this section, partially connected array structures are considered. While SNR of other groups (i.e., inter-group interference) was set to 40 dB for fully connected structures, we consider a moderate interference scenario for partially connected arrays where SNR of interfering groups is set to 20 dB since even constrained analog beamformers with fully connected arrays do not attain the performance of GEB at high interference. Furthermore, we merge Group-1 and Group-2 and denote it by Group-1' which includes 4 users and 5 different MPC clusters. Relative angular regions of MPCs of the merged group is taken as in Fig. 2 and the center of the MPCs is swept with shifting angle ϕ from -45° to 45° with increments of 0.1° as in previous part. We want to serve more users with a single analog beamformer which is the reason for merging groups. We will only be interested in the merged group and design a partially connected array for the analog beamformer of this group.⁸ Interference originated from Group-3 and Group-4 can be considered as any interference with a known covariance matrix. We will consider two different fixed partially connected arrays. The first one is called adjacent or ordered which has a connection matrix $\mathbf{\Pi}'_{ord} = \mathbf{I}_{D_1'} \otimes \mathbf{1}_{M/D_1'}$, where $\mathbf{1}_{M/D_1'} \in \mathbb{Z}^{M/D_1' \times 1}$ is a vector of ones. Second type of fixed partially connected arrays is called interlaced which has a connection matrix $\mathbf{\Pi}'_{int} = \mathbf{1}_{M/D_1'} \otimes \mathbf{I}_{D_1'}$.

In Fig. 8, beampatterns of Group-1' with fixed and dynamic partially connected arrays are given for $D_1' = 8$ where MPCs are located as in Fig. 2. We selected the number of RF chains as 8 since it is the minimum power of two that is larger than the number of MPCs. SNR of Group-1 and Group-2 before merging was set to 30 dB while SNR of other groups is set to 20 dB. We observe that ordered array forms wide beams

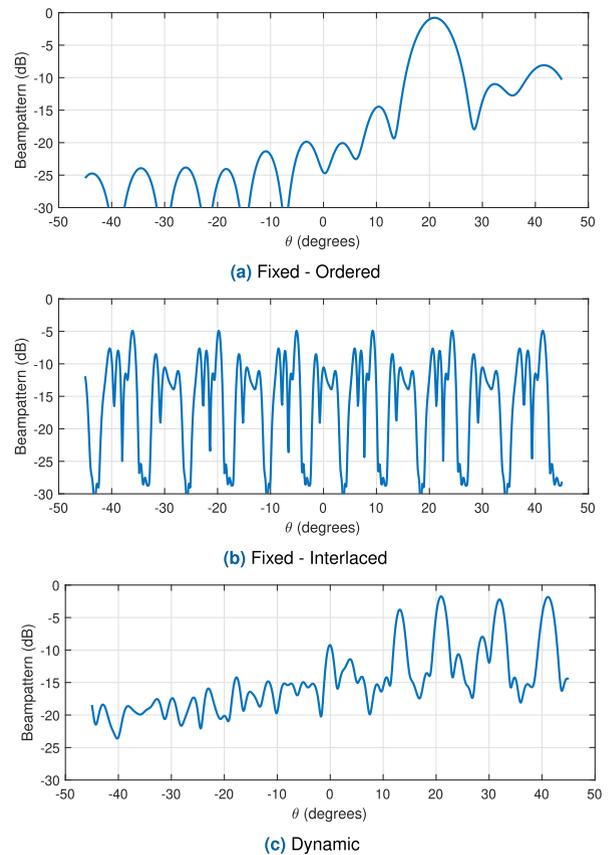


FIGURE 8. Beampatterns of Group-1' for partially connected array structures with $D_1' = 8$, $SNR^{(1)} = SNR^{(2)} = 30$ dB and $SNR^{(3)} = SNR^{(4)} = 20$ dB.

whereas dynamic subarray can form narrow beams located in the angular regions of Group-1'. Hence, dynamic subarray preserved the multipath diversity even though beam power is slightly higher at interfering angular regions. In addition, interference suppression of ordered array would be lost due to wide beams when the MPCs are close to each other in the mobile scenario. On the other hand, interlaced array completely lost interference suppression as it suffers from the grating lobe effect which occurs when the antenna spacing of an array is larger than half the wavelength. It is important to note that the proposed dynamic array is seen to effectively compromise between avoiding grating lobes and forming narrower beams.

Average spectral efficiency with changing SNR of Group-1' is given in Fig. 9 for $D_1' = 8$ where fixed and dynamic subarrays and fully connected PE-AM and GEB are considered while SNR of other groups is set to 20 dB. Firstly, it is important to note that PE-AM performs very close to GEB for this scenario where moderate interference is considered. If we compare partially connected array designs, dynamic subarray outperforms both fixed array structures. Another important point is that the gap between LMMSE and ZF type digital beamformers is smaller for the dynamic subarray which indicates that residual interference after the analog beamformer is smaller compared to fixed subarrays. Moreover, the gap between dynamic and fixed subarrays is

⁷The near-far effect stems from the fact that average received signal strength of different UEs may differ significantly depending on their distance to the BS.

⁸Other groups can be served by using another array at the BS, or by scheduling them in time and frequency resources.

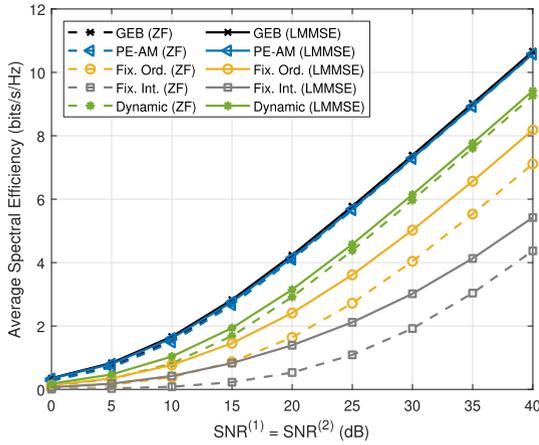


FIGURE 9. Average spectral efficiency of Group-1' vs. SNR for partially connected array structures with $D_{1'} = 8$ and $SNR^{(3)} = SNR^{(4)} = 20$ dB.

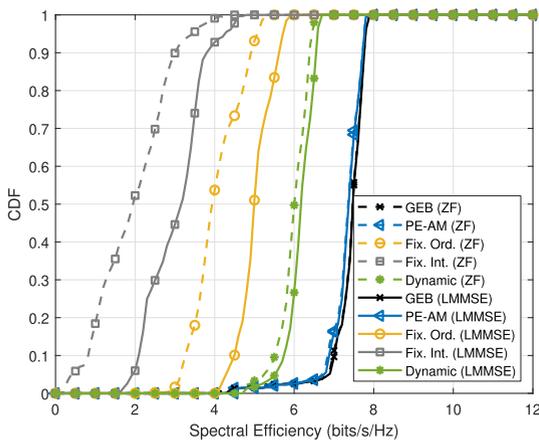


FIGURE 10. CDF of spectral efficiency of Group-1' for partially connected array structures with $D_{1'} = 8$, $SNR^{(1)} = SNR^{(2)} = 30$ dB and $SNR^{(3)} = SNR^{(4)} = 20$ dB.

larger in this paper compared to prior work [24], [28]. On the other hand, ordered array has higher average spectral efficiency than interlaced array as expected. We can make similar comments if we compare CDF of spectral efficiency curves which are given in Fig. 10. We observe that CDF curves of dynamic subarray are parallel to CDF curves of PE-AM and GEB whereas fixed arrays, especially interlaced structure, have wider CDF curves. This result shows the robustness of the dynamic subarray structure.

Fig. 11 shows the average BER of Group-1' with changing SNR for $D_{1'} = 8$ while SNR of other groups is set to 20 dB. It can be noted that BER is improved and MFB can be achieved for all beamformers with IB-DFE method. Furthermore, SNR gap between fully connected beamformers and dynamic subarray is around 3 dB for IB-DFE method at BER of 10^{-3} , while there is 8 dB SNR gap between dynamic and ordered subarrays. It can be seen that performance gap between dynamic and fixed subarrays is larger than the gap observed from spectral efficiency curves. Therefore, BER results showed the superiority of dynamic subarray over fixed ones more clearly. In conclusion, PE-AM attains the

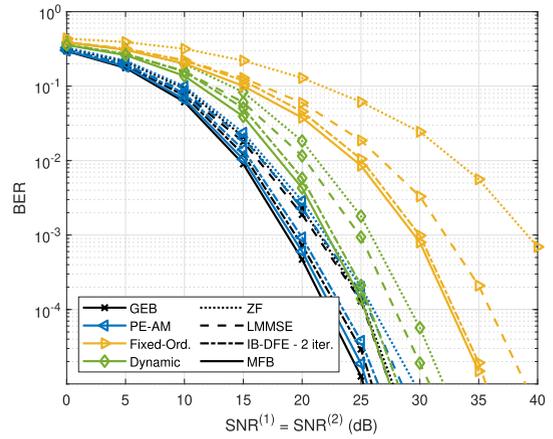


FIGURE 11. Average BER of Group-1' vs. SNR for partially connected array structures with $D_{1'} = 8$ and $SNR^{(3)} = SNR^{(4)} = 20$ dB.

performance of GEB, and proposed dynamic subarray algorithm performs close to fully connected arrays which shows the resilience of proposed algorithms against interference for the moderate interference scenario.

VII. CONCLUSION

In this paper, we proposed an interference-aware two-stage beamforming framework, specifically for JSDM, where interference is considered in both analog and digital beamforming stages. Near-optimal slowly varying statistical analog beamformer (i.e., GEB) is designed where inter-group interference suppression is imposed at this stage. GEB is approximated with both fully and partially connected arrays by considering constant-modulus constraint. Furthermore, an algorithm to find the optimal connection structure is proposed for dynamic subarray design. Low-complexity alternating minimization algorithms are employed for obtaining constrained analog beamformers. Superiority of proposed beamformers is shown by using comprehensive analysis tools. It is observed that PE-AM attains the performance of unconstrained GEB with moderate interference strength. Furthermore, this algorithm outperforms commonly used DFT beamformer and phase of GEB. On the other hand, dynamic subarray design performs close to fully connected ones and outperforms fixed subarrays. Moreover, MMSE criterion based IB-DFE type digital beamformers improve the performance of constrained analog beamformers as they take residual interference into account. This work could be extended to a spatial-wideband system as a future study.

APPENDIX A

COVARIANCE MATRICES IN REDUCED DIMENSION

Covariance matrix of $\tilde{\mathbf{s}}_{f,k}^{(g)}$ is expressed as

$$\begin{aligned} \mathbf{R}_{\tilde{\mathbf{s}}_f}^{(g)} &= \mathbb{E} \left\{ \tilde{\mathbf{s}}_{f,k}^{(g)} [\tilde{\mathbf{s}}_{f,k}^{(g)}]^H \right\} \\ &= \mathbb{E} \left\{ \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} \tilde{\mathbf{s}}_n^{(g)} e^{-j\frac{2\pi}{N}kn} \sum_{n'=0}^{N-1} \frac{1}{\sqrt{N}} [\tilde{\mathbf{s}}_{n'}^{(g)}]^H e^{j\frac{2\pi}{N}kn'} \right\} \end{aligned}$$

$$\begin{aligned}
 &= \frac{1}{N} \sum_{n=0}^{N-1} \sum_{n'=0}^{N-1} \mathbb{E} \left\{ \tilde{\mathbf{s}}_n^{(g)} [\tilde{\mathbf{s}}_{n'}^{(g)}]^H \right\} e^{j \frac{2\pi}{N} k(n'-n)} \\
 &= \frac{1}{N} \sum_{n=0}^{N-1} \sum_{n'=0}^{N-1} [\mathbf{S}^{(g)}]^H \mathbf{R}_s^{(g)} \delta_{nn'} \mathbf{S}^{(g)} e^{j \frac{2\pi}{N} k(n'-n)} \\
 &= [\mathbf{S}^{(g)}]^H \mathbf{R}_s^{(g)} \mathbf{S}^{(g)}, \tag{59}
 \end{aligned}$$

where $\mathbf{R}_s^{(g)}$ is the covariance matrix of intra-group signals of group g whose expression is given in (13). Similarly, covariance matrix of $\boldsymbol{\eta}_{f,k}^{(g)}$ can be expressed as $\mathbf{R}_{\boldsymbol{\eta}_f}^{(g)} = [\mathbf{S}^{(g)}]^H \mathbf{R}_{\boldsymbol{\eta}}^{(g)} \mathbf{S}^{(g)}$ where expression for $\mathbf{R}_{\boldsymbol{\eta}}^{(g)}$, which is the covariance matrix of sum of inter-group signals and noise, is given in (14). It is important to note that these covariance matrices are independent of frequency bin.

**APPENDIX B
PROOF OF LEMMA 1**

If there are multiple antennas at user m in group g , $\mathbf{h}_l^{(gm)}$ in (2) is replaced with $\mathbf{H}_l^{(gm)} \in \mathbb{C}^{M \times M_t^{(gm)}}$ where $M_t^{(gm)}$ is the number of antennas at the user side. Then, intra-group signals of group g in (5) can be rewritten as

$$\mathbf{s}_n^{(g)} = \sum_{m=1}^{K_g} \sum_{l=0}^{L-1} \mathbf{H}_l^{(gm)} \mathbf{x}_{(n-l)N}^{(gm)}, \tag{60}$$

where $\mathbf{x}_n^{(gm)} \in \mathbb{C}^{M_t^{(gm)} \times 1}$ is the precoded data vector of user m in group g with transmit power $\mathbb{E}\{\|\mathbf{x}_n^{(gm)}\|^2\} = \frac{E_s^{(g)}}{K_g}$. Precoded data vector is defined as $\mathbf{x}_n^{(gm)} \triangleq \mathbf{V}^{(gm)} \mathbf{d}_n^{(gm)}$ where $\mathbf{V}^{(gm)} \in \mathbb{C}^{M_t^{(gm)} \times d_t^{(gm)}}$ is the precoding matrix and $\mathbf{d}_n^{(gm)} \in \mathbb{C}^{d_t^{(gm)} \times 1}$ is the data vector with $d_t^{(gm)}$ streams. Transmitted symbols are assumed to be uncorrelated among streams and time indices which can be expressed as $\mathbb{E}\{\mathbf{d}_n^{(gm)} [\mathbf{d}_n^{(gm)}]^H\} = \mathbf{I}_{d_t^{(gm)}} \delta_{nn'}$.

Assuming that channels obey Kronecker channel model, which is a practical model for spatially correlated MIMO [48], the channel matrices of users can be written as

$$\mathbf{H}_l^{(gm)} = [\mathbf{R}_l^{(gm)}]^{1/2} \bar{\mathbf{H}}_l^{(gm)} \left([\boldsymbol{\Psi}_l^{(gm)}]^{1/2} \right)^T, \tag{61}$$

where entries of $\bar{\mathbf{H}}_l^{(gm)} \in \mathbb{C}^{M \times M_t^{(gm)}}$ are independent zero-mean circularly symmetric complex Gaussian random variables with unit variance. In (61), $\mathbf{R}_l^{(gm)}$ defined in (3) expresses the spatial correlation among the receive antenna elements depending on AoAs of MPCs at the BS whereas $\boldsymbol{\Psi}_l^{(gm)} \in \mathbb{C}^{M_t^{(gm)} \times M_t^{(gm)}}$ shows the spatial correlation among transmit antenna elements depending on angle of departures (AoDs) at the user side. The matrix $\boldsymbol{\Psi}_l^{(gm)}$ can be expressed in a similar form as $\mathbf{R}_l^{(gm)}$. Let $\mathbf{H}_l^{(gm)}$ matrices be independent for different MPC clusters as in single-antenna user case. Then, channel statistics for a given MPC can be written as

$$\mathbb{E} \left\{ \text{vec} \left(\mathbf{H}_l^{(gm)} \right) \text{vec} \left(\mathbf{H}_l^{(gm)} \right)^H \right\} = \boldsymbol{\Psi}_l^{(gm)} \otimes \mathbf{R}_l^{(gm)}, \tag{62}$$

which can be obtained after some manipulations starting from (61). Moreover, the channel gain can be written as

$$\begin{aligned}
 &\sum_{l=0}^{L-1} \text{Tr} \left\{ \mathbb{E} \left\{ \mathbf{H}_l^{(gm)} [\mathbf{H}_l^{(gm)}]^H \right\} \right\} \\
 &= \sum_{l=0}^{L-1} \text{Tr} \left\{ \boldsymbol{\Psi}_l^{(gm)} \right\} \text{Tr} \left\{ \mathbf{R}_l^{(gm)} \right\} = \gamma^{(gm)}, \tag{63}
 \end{aligned}$$

where $\text{Tr}\{\boldsymbol{\Psi}_l^{(gm)}\}$ is set to 1. Considering the definitions above, covariance matrix of intra-group signals of group g can be calculated as

$$\begin{aligned}
 \mathbf{R}_s^{(g)} &= \mathbb{E} \left\{ \mathbf{s}_n^{(g)} [\mathbf{s}_n^{(g)}]^H \right\} \\
 &= \sum_{m=1}^{K_g} \sum_{l=0}^{L-1} \mathbb{E} \left\{ \mathbf{H}_l^{(gm)} \mathbf{x}_n^{(g)} [\mathbf{x}_n^{(g)}]^H [\mathbf{H}_l^{(gm)}]^H \right\}. \tag{64}
 \end{aligned}$$

Let the expectation inside the summation in (64) is denoted by $\Phi_l^{(gm)}$. Then, entry of $\Phi_l^{(gm)}$ at i^{th} row and j^{th} column can be expressed as

$$\begin{aligned}
 &[\Phi_l^{(gm)}]_{(i,j)} \\
 &= \mathbb{E} \left\{ [\mathbf{H}_l^{(gm)}]_{(i,:)} \mathbf{x}_n^{(g)} [\mathbf{x}_n^{(g)}]^H [\mathbf{H}_l^{(gm)}]_{(j,:)}^H \right\} \\
 &= \mathbb{E} \left\{ \text{Tr} \left\{ \mathbf{x}_n^{(g)} [\mathbf{x}_n^{(g)}]^H [\mathbf{H}_l^{(gm)}]_{(j,:)}^H [\mathbf{H}_l^{(gm)}]_{(i,:)} \right\} \right\} \\
 &= \text{Tr} \left\{ \mathbb{E} \left\{ \mathbf{x}_n^{(g)} [\mathbf{x}_n^{(g)}]^H \right\} \mathbb{E} \left\{ [\mathbf{H}_l^{(gm)}]_{(j,:)}^H [\mathbf{H}_l^{(gm)}]_{(i,:)} \right\} \right\} \\
 &= \text{Tr} \left\{ \mathbf{V}^{(gm)} [\mathbf{V}^{(gm)}]^H \boldsymbol{\Psi}_l^{(gm)} [\mathbf{R}_l^{(gm)}]_{(i,j)} \right\} \\
 &= \text{Tr} \left\{ [\mathbf{V}^{(gm)}]^H \boldsymbol{\Psi}_l^{(gm)} \mathbf{V}^{(gm)} \right\} [\mathbf{R}_l^{(gm)}]_{(i,j)} \\
 &= c^{(gm)} [\mathbf{R}_l^{(gm)}]_{(i,j)}, \tag{65}
 \end{aligned}$$

where the constant is $c^{(gm)} = \text{Tr}\{[\mathbf{V}^{(gm)}]^H \boldsymbol{\Psi}_l^{(gm)} \mathbf{V}^{(gm)}\}$. With the result above, covariance matrix of intra-group signals can be rewritten as

$$\mathbf{R}_s^{(g)} = \sum_{m=1}^{K_g} \sum_{l=0}^{L-1} \Phi_l^{(gm)} = \sum_{m=1}^{K_g} \sum_{l=0}^{L-1} c^{(gm)} \mathbf{R}_l^{(gm)}, \tag{66}$$

which has a similar form as in (13). The only difference is that the constant $c^{(gm)}$ is dependent on the precoding and slowly varying AoDs at the user side. Furthermore, covariance matrix of sum of inter-group signals and noise $\mathbf{R}_{\boldsymbol{\eta}}^{(g)}$ can be obtained in a similar form as in (14) with the usage of constant $c^{(gm)}$. It is feasible to assume that statistical precoding is utilized at the user side. In this case, the constant $c^{(gm)}$ becomes a slowly varying parameter that can easily be learned at the BS. Hence, the same statistical analog beamforming method proposed in Section III-A can be used for the multi-antenna user case.

Moreover, proposed digital beamforming and channel estimation scheme would remain the same with the modified effective channel definition which is expressed as

$$\mathbf{H}_{\text{eff},l}^{(g, gm)} = [\mathbf{S}^{(g)}]^H \mathbf{H}_l^{(gm)} \mathbf{V}^{(gm)}, \tag{67}$$

where effects of both the beamformer at the BS and the precoder at the user side are considered. Note that the BS does not require the knowledge of the precoder at the user side during channel estimation phase since effective channels in (67) are estimated directly.

APPENDIX C PROOF OF LEMMA 2

Replacing $\mathbf{S}^{(g)}$ with $\mathbf{S}^{(g)}\mathbf{A}$ results in a similarity transformation for the second term in the determinant in (12), i.e., $\mathbf{A}^{-1}([\mathbf{S}^{(g)}]^H \mathbf{R}_\eta^{(g)} \mathbf{S}^{(g)})^{-1}([\mathbf{S}^{(g)}]^H \mathbf{R}_s^{(g)} \mathbf{S}^{(g)})\mathbf{A}$. Similarity transformation does not affect the cost function since generalized eigenvectors diagonalize this term [44]. Hence, alternative representations of GEB can be used without loss of performance.

APPENDIX D EXPECTED SINR IN REDUCED DIMENSION

In reduced dimension, it is possible to find an expected SINR value for each group. Intra-group signal power can be divided by power of sum of inter-group interference and noise in reduced dimension by using (6). With this method, expected SINR of group g can be expressed as

$$\begin{aligned} \overline{\text{SINR}}^{(g)} &= \frac{\mathbb{E}\left\{\left\|\tilde{\mathbf{s}}_{f,k}^{(g)}\right\|^2\right\}}{\mathbb{E}\left\{\left\|\tilde{\boldsymbol{\eta}}_{f,k}^{(g)}\right\|^2\right\}} = \frac{\text{Tr}\left\{\mathbb{E}\left\{\tilde{\mathbf{s}}_{f,k}^{(g)}\left[\tilde{\mathbf{s}}_{f,k}^{(g)}\right]^H\right\}\right\}}{\text{Tr}\left\{\mathbb{E}\left\{\tilde{\boldsymbol{\eta}}_{f,k}^{(g)}\left[\tilde{\boldsymbol{\eta}}_{f,k}^{(g)}\right]^H\right\}\right\}} \\ &= \frac{\text{Tr}\left\{\mathbf{R}_{\tilde{\mathbf{s}}_f}^{(g)}\right\}}{\text{Tr}\left\{\mathbf{R}_{\tilde{\boldsymbol{\eta}}_f}^{(g)}\right\}} = \frac{\text{Tr}\left\{[\mathbf{S}^{(g)}]^H \mathbf{R}_s^{(g)} \mathbf{S}^{(g)}\right\}}{\text{Tr}\left\{[\mathbf{S}^{(g)}]^H \mathbf{R}_\eta^{(g)} \mathbf{S}^{(g)}\right\}}, \quad (68) \end{aligned}$$

where expressions for covariance matrices $\mathbf{R}_{\tilde{\mathbf{s}}_f}^{(g)}$ and $\mathbf{R}_{\tilde{\boldsymbol{\eta}}_f}^{(g)}$ are taken from Appendix A.

APPENDIX E ASYMPTOTIC SINR ANALYSIS

Assuming the perfect knowledge of intra-group user symbols, reliability matrices are taken as $\mathbf{P}^{(g)} = \frac{E_s^{(g)}}{K_g} \mathbf{I}_{K_g}$. In this case, $\mathbf{A}_k^{(g)} = -[\mathbf{D}_k^{(g)}]^{-1}$ relation can be obtained from (23) and (24). Then, $\mathbf{A}_k^{(g)}$ can be rewritten as

$$\mathbf{A}_k^{(g)} = \frac{K_g}{E_s^{(g)}} \mathbf{I}_{K_g} + [\boldsymbol{\Lambda}_{eff,k}^{(g,g)}]^H \mathbf{R}_{\tilde{\boldsymbol{\eta}}_{f,k}^{(g)}}^{-1} \boldsymbol{\Lambda}_{eff,k}^{(g,g)}, \quad (69)$$

if matrix inversion lemma is utilized for (23) with asymptotic reliability matrices. Considering the results above, feedback filters in (21) can be expressed as

$$\begin{aligned} \mathbf{C}_k^{(g)} &= -\mathbf{I}_{K_g} + \mathbf{A}_k^{(g)} \boldsymbol{\Delta}^{(g)} \\ &= -\mathbf{I}_{K_g} + \left(\frac{K_g}{E_s^{(g)}} \mathbf{I}_{K_g} + [\boldsymbol{\Lambda}_{eff,k}^{(g,g)}]^H \mathbf{R}_{\tilde{\boldsymbol{\eta}}_{f,k}^{(g)}}^{-1} \boldsymbol{\Lambda}_{eff,k}^{(g,g)} \right) \boldsymbol{\Delta}^{(g)}. \quad (70) \end{aligned}$$

Furthermore, feedforward filters can be obtained if (70) is substituted in (18). After some manipulations that leverage

matrix inversion lemma, feedforward filters can be written as

$$\mathbf{W}_k^{(g)} = \mathbf{R}_{\tilde{\boldsymbol{\eta}}_{f,k}^{(g)}}^{-1} \boldsymbol{\Lambda}_{eff,k}^{(g,g)} \boldsymbol{\Delta}^{(g)}. \quad (71)$$

It can be observed from (71) that feedforward filters become scaled forms of whitening matched filters for the asymptotic case. That is, feedback filters perfectly cancel the intra-group signals, and the equalizers reduce to optimal whitening matched filters. Finally, it can be shown that asymptotic SINRs of users can be found as in (49), if complex amplitude in (47) and residual interference power from (48) are computed with the asymptotic feedforward and feedback filters that are defined above.

APPENDIX F CHANNEL ESTIMATORS

A. LMMSE TYPE CHANNEL ESTIMATOR

LMMSE type channel estimator of group g can be written as

$$\mathbf{Z}^{(g)} = \mathbf{R}_{\bar{\mathbf{y}}^{(g)}}^{-1} \mathbf{R}_{\bar{\mathbf{y}}^{(g)} \bar{\mathbf{h}}_{eff}^{(g,g)}} \bar{\mathbf{h}}_{eff}^{(g,g)} \quad (72)$$

where covariance matrix between $\bar{\mathbf{y}}^{(g)}$ and $\bar{\mathbf{h}}_{eff}^{(g,g)}$ and covariance matrix of $\bar{\mathbf{y}}^{(g)}$ can be computed as

$$\begin{aligned} \mathbf{R}_{\bar{\mathbf{y}}^{(g)} \bar{\mathbf{h}}_{eff}^{(g,g)}} &= \mathbb{E}\left\{\bar{\mathbf{y}}^{(g)}\left[\bar{\mathbf{h}}_{eff}^{(g,g)}\right]^H\right\} = \left(\mathbf{X}^{(g)} \otimes \mathbf{I}_{D_g}\right) \mathbf{R}_{\tilde{\mathbf{h}}_{eff}^{(g,g)}}, \quad (73) \end{aligned}$$

$$\begin{aligned} \mathbf{R}_{\bar{\mathbf{y}}^{(g)}} &= \mathbb{E}\left\{\bar{\mathbf{y}}^{(g)}\left[\bar{\mathbf{y}}^{(g)}\right]^H\right\} \\ &= \left(\mathbf{X}^{(g)} \otimes \mathbf{I}_{D_g}\right) \mathbf{R}_{\tilde{\mathbf{h}}_{eff}^{(g,g)}} \left(\mathbf{X}^{(g)} \otimes \mathbf{I}_{D_g}\right)^H + \mathbf{I}_T \otimes \mathbf{R}_{\tilde{\boldsymbol{\eta}}_f}^{(g)}, \quad (74) \end{aligned}$$

where covariance matrix of sum of inter-group interference and noise terms in reduced dimension is denoted by $\mathbb{E}\left\{\tilde{\boldsymbol{\eta}}_n^{(g)}\left[\tilde{\boldsymbol{\eta}}_{n'}^{(g)}\right]^H\right\} = \mathbf{R}_{\tilde{\boldsymbol{\eta}}_f}^{(g)} \delta_{nn'}$. This covariance matrix can be expressed as $\mathbf{R}_{\tilde{\boldsymbol{\eta}}_f}^{(g)} = [\mathbf{S}^{(g)}]^H \mathbf{R}_\eta^{(g)} \mathbf{S}^{(g)}$. Other groups are assumed to be in data transmission mode in (74) according to Remark 3. That is, $\mathbf{X}^{(g')}$ is taken as a random matrix for $g' \neq g$ which is why $\mathbf{R}_{\tilde{\boldsymbol{\eta}}_f}^{(g)}$ is used in (74). Covariance matrix of the overall effective channel of group g is expressed as

$$\begin{aligned} \mathbf{R}_{\tilde{\mathbf{h}}_{eff}^{(g,g)}} &= \mathbb{E}\left\{\bar{\mathbf{h}}_{eff}^{(g,g)}\left[\bar{\mathbf{h}}_{eff}^{(g,g)}\right]^H\right\} \\ &= \text{blkdiag}\left\{\sum_{l=0}^{L-1} \mathbf{E}_{L,l+1} \otimes [\mathbf{S}^{(g)}]^H \mathbf{R}_l^{(g,m)} \mathbf{S}^{(g)}\right\}_{m=1}^{K_g} \quad (75) \end{aligned}$$

where $\mathbf{E}_{L,l+1} \triangleq \mathbf{e}_{L,l+1} \mathbf{e}_{L,l+1}^H$ and $\mathbf{e}_{L,l+1} \triangleq [\mathbf{I}_L]_{(:,l+1)}$. It is important to note that due to uncorrelated structure of channel vectors, (75) has a block diagonal form where blocks in the diagonals are covariance matrices of effective channels.

B. LS TYPE CHANNEL ESTIMATOR

LS type channel estimator of group g can be expressed as

$$\mathbf{Z}^{(g)} = \left(\mathbf{X}^{(g)}\left([\mathbf{X}^{(g)}]^H \mathbf{X}^{(g)}\right)^{-1}\right) \otimes \mathbf{I}_{D_g}. \quad (76)$$

However, $\bar{\mathbf{h}}_{\text{eff}}^{(g,g)}$ has a sparse structure which degrades the performance of LS estimator given in (76). We should eliminate the columns of $\mathbf{X}^{(g)}$ corresponding to the inactive MPCs and find the estimator in (76). Then, we should insert zero columns into the estimator to make up for the eliminated columns. These zero columns in $\mathbf{Z}^{(g)}$ leads to estimating zero vectors for channels belonging to inactive MPCs. In this way, the dimension of the inverse in the LS estimator is reduced and the performance of the estimator is increased.

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