In-network Compression for Multiterminal Cascade MIMO Systems

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Abstract

We study the problem of receive beamforming in uplink cascade multiple-input multiple-output (MIMO) systems as an instance of that of cascade multiterminal source coding for lossy function computation. Using this connection, we develop two coding schemes for the second and show that their application leads to beamforming schemes for the first. In the first coding scheme, each terminal in the cascade sends a description of the source that it observes; the decoder reconstructs all sources, lossily, and then computes an estimate of the desired function. This scheme improves upon standard routing in that every terminal only compresses the innovation of its source w.r.t. the descriptions that are sent by the previous terminals in the cascade. In the second scheme, the desired function is computed gradually in the cascade network, and each terminal sends a finer description of it. In the context of uplink cascade MIMO systems, the application of these two schemes leads to *centralized* receive-beamforming and *distributed* receive-beamforming, respectively. Numerical results illustrate the performance of the proposed methods and show that they outperform standard routing.

I. INTRODUCTION

Consider the cascade communication system for function computation shown in Figure 1. Terminal l, l = 1, ..., L, observes, or measures, a discrete memoryless source S_l^n and communicates with Terminal (l+1) over an error-free finite-capacity link of rate R_l . Terminal (L+1) does not observe any source, and plays the role of a decoder which wishes to reconstruct

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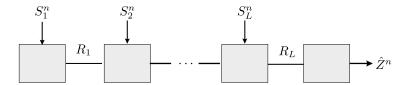


Fig. 1: Multi-terminal cascade source coding for lossy function computation.

a function Z^n lossily, to within some average fidelity level D, where $Z_i = \varphi(S_{1,i},\ldots,S_{L,i})$ for some function $\varphi(\cdot)$. The memoryless sources (S_1^n,\ldots,S_L^n) are arbitrary correlated among them, with joint measure $p_{S_1,\ldots,S_L}(s_1,\ldots,s_L)$. For this communication system, optimal tradeoffs among compression rate tuples (R_1, \ldots, R_L) and allowed distortion level D, captured by the rate-distortion region of the model, are not known in general, even if the sources are independent. For some special cases, inner and outer bounds on the rate-distortion region, that do not agree in general, are known, e.g., in [2] for the case L=2. A related work for the case L=2 has also appeared in [3]. For the general case with $L \geq 2$, although a single-letter characterization of the rate-distortion region seems to be out of reach, one can distinguish essentially two different transmission approaches or modes. In the first mode, each terminal operates essentially as a routing node. That is, each terminal in the cascade sends an appropriate compressed version, or description, of the source that it observes; the decoder reconstructs all sources, lossily, and then computes an estimate of the desired function. In this approach, the computation is performed centrally, at only the decoder, i.e., Terminal (L+1). In the second mode, Terminal $l, l = 1, \ldots, L$, processes the information that it gets from the previous terminal, and then describes it, jointly with its own observation or source, to the next terminal. That is, in a sense, the computation is performed distributively in the network. (See, e.g., [4]–[6], where variants of this approach are sometimes referred to as *in-network processing*).

Consider now the seemingly unrelated uplink multiple-input multiple-output (MIMO) system model shown in Figure 2. In this model, M users communicate concurrently with a common base station (BS), as in standard uplink wireless systems. The base station is equipped with a large number of antennas, e.g., a Massive MIMO BS; and the baseband processing is distributed across a number, say L, of modules or radio remote units (RRUs). The modules are connected each to a small number of antennas; and are concatenated in a line network, through a common

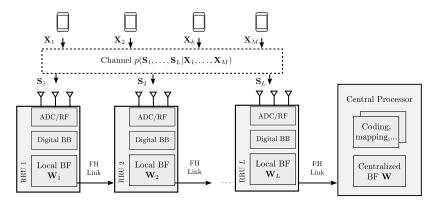


Fig. 2: Chained MIMO architecture for uplink Massive MIMO systems.

fronthaul link that connects them to a central processor (CP) unit. This architecture, sometimes referred to as "chained MIMO" [7] and proposed as an alternative to the standard one in which each RRU has its dedicated fronthaul link to the CP [8]–[17], offers a number of advantages and an additional degree of flexibility if more antennas/modules are to be added to the system. The reader may refer to [18]–[21] where examples of testbed implementations of this novel architecture can be found. For this architecture, depending on the amount of available channel state information (CSI), receive-beamforming operations may be better performed centrally at the CP or distributively across RRUs. Roughly, if CSI is available only at the CP, not at the RRUs, it seems reasonable that beamforming operations be performed only centrally, at the CP. In this case, RRU l, l = 1, ..., L, sends a compressed version \hat{S}_l of its received signal S_l to the CP which first collects the vector $(\hat{S}_l, ..., \hat{S}_L)$ and then performs receive-beamforming on it. In contrast, if local CSI is available or can be acquired at the RRUs, due to the linearity of the receive beamforming (which is a simple matrix multiplication) parts of the receive beamforming operations can be performed distributively at the RRUs (see Section III).

The above shows some connections among the model of Figure 2 and that, more general, of Figure 1. In this paper, we study them using a common framework. Specifically, we develop two coding schemes for the multiterminal cascade source coding problem of Figure 1; and then show that their application to the uplink cascade MIMO system of Figure 1 leads to schemes for receive-beamforming which, depending on the amount of available CSI at the RRUs, are better performed centrally at the CP or distributively across RRUs. In the first coding scheme, each

terminal in the cascade sends a description of the source that it observes; the decoder reconstructs all sources lossily and then computes an estimate of the desired function. This scheme improves upon standard routing in that every terminal only compresses the innovation of its source w.r.t. the descriptions that are sent by the previous terminals in the cascade. In the second scheme, the desired function is computed gradually in the cascade network; and each terminal sends a finer description of it. Furthermore, we also derive a lower bound on the minimum distortion at which the desired function can be reconstructed at the decoder by relating the problem to the Wyner-Ziv type system studied in [22]. Numerical results show that the proposed methods outperform standard compression strategies and perform close to the lower bound in some regimes.

A. Notation

Throughout, we use the following notation. Upper case letters are used to denote random variables, e.g., X; lower case letters used to denote realizations of random variables x; and calligraphic letters denote sets, e.g., \mathcal{X} . The cardinality of a set \mathcal{X} is denoted by $|\mathcal{X}|$. The length-n sequence (X_1,\ldots,X_n) is denoted as X_k^n ; and, for $(1 \le k \le j \le n)$ the sub-sequence (X_k,X_{k+1},\ldots,X_j) is denoted as X_k^j . Boldface upper case letters denote vectors or matrices, e.g., \mathbf{X} , where context should make the distinction clear. For an integer $L \ge 1$, we denote the set of integers smaller or equal L as $\mathcal{L} \triangleq \{1,\ldots,L\}$; and, for $1 \le l \le L$, we use the shorthand notations $\mathcal{L}_l \triangleq \{1,2,\ldots l\}$, $\mathcal{L}_l^c \triangleq \{l+1\ldots L\}$ and $\mathcal{L}/l \triangleq \{1,\ldots,l-1,l+1,\ldots,L\}$. We denote the covariance of a vector \mathbf{X} by $\mathbf{\Sigma}_{\mathbf{x}} \triangleq \mathbf{E}[\mathbf{X}\mathbf{X}^H]$; $\mathbf{\Sigma}_{\mathbf{x},\mathbf{y}}$ is the cross-correlation $\mathbf{\Sigma}_{\mathbf{x},\mathbf{y}} \triangleq \mathbf{E}[\mathbf{X}\mathbf{Y}^H]$, and the conditional correlation matrix of \mathbf{X} given \mathbf{Y} as $\mathbf{\Sigma}_{\mathbf{x}|\mathbf{y}} \triangleq \mathbf{\Sigma}_{\mathbf{x}} - \mathbf{\Sigma}_{\mathbf{x},\mathbf{y}} \mathbf{\Sigma}_{\mathbf{y}}^{-1} \mathbf{\Sigma}_{\mathbf{y},\mathbf{x}}$. The length-N vector with all entries equal zero but the l-th element which is equal unity is denoted as δ_l , i.e., $\delta_l \triangleq [\mathbf{0}_{1 \times l-1}, \mathbf{1}, \mathbf{0}_{1 \times N-l}]$; and the $l \times N$ matrix whose entries are all zeros, but the first l diagonal elements which are equal unity, is denoted by $\mathbf{\bar{I}}_l$, $[\mathbf{\bar{I}}_l]_{i,i} = 1$ for $i \le l$ and 0 otherwise. We also, define $\log^+(\cdot) \triangleq \max\{0,\log(\cdot)\}$.

II. CASCADE SOURCE CODING SYSTEM MODEL

Let $\{S_{1,i}, S_{2,i}, \dots, S_{L,i}\}_{i=1}^n = (S_1^n, \dots, S_L^n)$ be a sequence of n independent and identically distributed (i.i.d.) samples of the L-dimensional source (S_1, S_2, \dots, S_L) jointly distributed as $p(s_1, \dots, s_L)$ over $S_1 \times \dots \times S_L$. For convenience, we denote $S_{L+1}^n \triangleq \emptyset$.

A cascade of (L+1) terminals are concatenated as shown in Figure 1, such that Terminal $l, l=1,\ldots,L$, is connected to Terminal (l+1) over an error-free link of capacity R_l bits per channel use. Terminal (L+1) is interested in reconstructing a sequence Z^n lossily, to within some fidelity level, where $Z_i=\varphi(S_{1,i},\ldots,S_{L,i}), i=1,\ldots,n$, for some function $\varphi:S_1\times\ldots\times S_L\to \mathcal{Z}$. To this end, Terminal $l, l=1,\ldots,L$, which observes the sequence S_l^n and receives message $m_{l-1}\in\mathcal{M}_{l-1}\triangleq\{1,\ldots,M_{l-1}^{(n)}\}$ from Terminal (l-1), generates a message $m_l\in\mathcal{M}_l$ as $m_l=f_l^{(n)}(s_l^n,m_{l-1})$ for some encoding function $f_l^{(n)}:S_l^n\times\mathcal{M}_{l-1}\to\mathcal{M}_l$, and forwards it over the error-free link of capacity R_l to Terminal (l+1). At Terminal (L+1), the message m_L is mapped to an estimate $\hat{Z}^n=g^{(n)}(m_L)$ of Z^n , using some mapping $g^{(n)}:\mathcal{M}_L\to\hat{\mathcal{Z}}^n$. Let $\hat{\mathcal{Z}}$ be the reconstruction alphabet and $d:\mathcal{Z}\times\hat{\mathcal{Z}}\to[0,\infty)$ be a single letter distortion. The distortion between Z^n and the reconstruction \hat{Z}^n is defined as $d(z^n;\hat{z}^n)=\frac{1}{n}\sum_{i=1}^n d(z_i;\hat{z}_i)$.

Definition 1. A tuple $(R_1, ..., R_L)$ is said to achieve distortion D for the cascade multi-terminal source coding problem if there exist L encoding functions $f_l^{(n)}: \mathcal{S}_l \times \mathcal{M}_{l-1} \to \mathcal{M}_l$, l = 1, ..., L, and a function $g^{(n)}: \mathcal{M}_L \to \hat{\mathcal{Z}}$ such that

$$R_l \ge \frac{1}{n} \log M_l^{(n)}, \ l = 1, \dots, L \ \ and \ \ D \ge \frac{1}{n} \mathbb{E}[d(Z^n; \hat{Z}^n)].$$

The rate-distortion (RD) region $\mathcal{R}(D)$ of the cascade multi-terminal source coding problem is defined as the closure of all rate tuples (R_1, \ldots, R_L) that achieve distortion D.

III. SCHEMES FOR CASCADE SOURCE CODING

In this section, we develop two coding schemes for the cascade source coding model of Figure 1 and analyze the RD regions that they achieve.

A. Improved Routing (IR)

A simple strategy which is inspired by standard routing (SR) in graphical networks and referred to as multiplex-and-forward in [5] has Terminal l, l = 1, ..., L, forward a compressed version of its source to the next terminal, in addition to the bit stream received from the previous terminal in the cascade (without processing). The decoder decompresses all sources and then

outputs an estimate of the desired function. In SR, observations are compressed independently and correlation with the observation of the next terminal in the cascade is not exploited.

In this section, we propose a scheme, to which we refer to as "Improved Routing" (IR), which improves upon SR by compressing at each terminal its observed signal S_l^n into a description U_l^n considering the compressed observations from the previous terminals, i.e., $(U_1^n, \ldots, U_{l-1}^n)$ as side information available both at the encoder and the decoder [23]. Thus, each terminal only compresses the innovative part of the observation with respect to the compressed signals from previous terminals (see Section IV-A). In doing so, it uses B_l bits per source sample. Along with the produced compression index of rate B_l , each terminal also forwards the bit stream received from the previous terminal to the next one without processing. The decoder successively decompresses all sources and outputs an estimate of the function of interest.

Theorem 1. The RD region $\mathcal{R}_{IR}(D)$ that is achievable with the IR scheme is given by the union of rate tuples (R_1, \ldots, R_L) satisfying

$$R_l \ge \sum_{i=1}^l I(S_i; U_i | U_1, \dots, U_{i-1}), \quad \text{for} \quad l = 1, \dots, L,$$
 (1)

for some joint pmf $p(s_1, \ldots, s_L) \prod_{l=1}^L p(u_l | s_l, u_1, \ldots, u_{l-1})$ and function g, s.t. $\hat{Z} = g(U_1, \ldots, U_L)$ and $E[d(Z, \hat{Z})] \leq D$.

Remark 1. The auxiliary random variables (U_1, \ldots, U_L) that are involved in (1) satisfy the following Markov Chains

$$U_l - (S_l, U_{\mathcal{L}_{l-1}}) - (S_{\mathcal{L}/l}, U_{\mathcal{L}_l^c}) \quad for \quad l = 1, \dots, L,$$
 (2)

where
$$U_{\mathcal{L}_{l-1}} = (U_1, \dots, U_{l-1})$$
 and $U_{\mathcal{L}_l^c} = (U_{l+1}, \dots, U_L)$.

Outline Proof: Fix $\epsilon > 0$, and a joint pmf $p(s_1, \ldots, s_L, u_1, \ldots, u_L)$ that factorizes as

$$p(s_1, \dots, s_L, u_1, \dots, u_L) = p(s_1, \dots, s_L) \prod_{l=1}^L p(u_l | s_l, u_1, \dots, u_{l-1}),$$
(3)

and a reconstruction function $g(\cdot)$ such that $\mathrm{E}[d(Z;g(U_1,\ldots,U_L))] \leq D/(1+\epsilon)$. Also, fix non-negative R_1,\ldots,R_L such that for $R_l \geq R_1+\cdots+R_{l-1}$, for $l=1,\ldots,L$.

Codebook generation: Let $B_l = R_l - (R_1 + \dots + R_{l-1})$ for $l = 1, \dots, L$. Generate a codebook \mathcal{C}_1 consisting of a collection of 2^{nB_1} codewords $\{u_1^n(i_1)\}$, indexed with $i_1 = 1, \dots, 2^{nB_1}$, where codeword $u_1^n(i_1)$ has its elements generated i.i.d. according to $p(u_1)$. For each index i_1 , generate a codebook $\mathcal{C}_2(i_1)$ consisting of a collection of 2^{nB_2} codewords $\{u_2^n(i_1,i_2)\}$ indexed with $i_2 = 1, \dots, 2^{nB_2}$, where codeword $u_2^n(i_1,i_2)$ is generated independently and i.i.d. according to $p(u_2|u_1)$. Similarly, for each index tuple (i_1, \dots, i_{l-1}) generate a codebook $\mathcal{C}_l(i_1, \dots, i_{l-1})$ of 2^{nB_l} codewords $\{u_l^n(i_1, \dots, i_l)\}$ indexed with $i_l = 1, \dots, 2^{nB_l}$, and where codeword $u_l^n(i_1, \dots, i_l)$ has its elements generated i.i.d. according to $p(u_l|u_1, \dots, u_{l-1})$.

Encoding at Terminal 1: Terminal 1 finds an index i_1 such that $u_1^n(i_1) \in \mathcal{C}_1$ is strongly ϵ -jointly typical with s_1^n , i.e., $(u_1^n(i_1), s_1^n) \in \mathcal{T}_{[U_1S_1]}^{(n)}$. Using standard arguments, this step can be seen to have vanishing probability of error as long as n is large enough and

$$B_1 \ge I(S_1; U_1). \tag{4}$$

Then, it forwards $m_1 = i_1$ to Terminal 2.

Encoding at Terminal $l \geq 2$: Upon reception of m_{l-1} with the indices $m_{l-1} = (i_1, \ldots, i_{l-1})$, Terminal l finds an index i_l such that $(u_1^n(i_1), \ldots, u_l^n(i_1, \ldots, i_l))$ are strongly ϵ -jointly typical with s_l^n , i.e., $(u_1^n(i_1), \ldots, u_l^n(i_1, \ldots, i_l), s_l^n) \in \mathcal{T}^{(n)}_{[U_1, \ldots, U_l, S_l]}$. Using standard arguments, this step can be seen to have vanishing probability of error as long as n is large enough and

$$B_l \ge I(U_l; S_l | U_1, \dots, U_{l-1}).$$
 (5)

Then, it forwards i_l and m_{l-1} to Terminal (l+1) as $m_l = (i_l, m_{l-1})$.

Reconstruction at end Terminal (L+1): Terminal (L+1) collects all received indices as $m_L = (i_1, \ldots, i_L)$, and reconstructs the codewords $(u_1^n(i_1), \ldots, u_L^n(i_1, \ldots, i_L))$. Then, it reconstructs an estimate of Z^n sample-wise as $\hat{Z}_i = g(u_{1,i}(i_1), u_{2,i}(i_1, i_2), \ldots, u_{L,i}(i_1, \ldots, i_L))$, $i = 1, \ldots, n$. Note that in doing so, the average distortion constraint is satisfied.

Finally, substituting $B_l = R_l - (R_1 + \cdots + R_{l-1})$ in (4) and (5), we get (1). This completes the proof of Theorem 1.

Remark 2. In the coding scheme of Theorem 1, the compression rate on the communication hop

between Terminal l and Terminal (l+1), $l=1,\ldots,L$, can be improved further (i.e., reduced) by taking into account sequence S^n_{l+1} as decoder side information, through Wyner-Ziv binning. The resulting strategy, however, is not of "routing type", unless every Wyner-Ziv code is restricted to account for the worst side information ahead in the cascade, i.e., binning at Terminal l accounts for the worst quality side information among the sequences $\{S^n_j, j=l+1, \ldots, L\}$. Also, in this case, since the end Terminal (L+1), or CP, does not observe any side information, i.e., $S^n_{L+1}=\emptyset$, this strategy makes most sense if the Wyner-Ziv codes are chosen such that the last relay terminal in the cascade, i.e., Terminal L, recovers an estimate of the desired function and then sends it using a standard rate-distortion code to the CP in a manner that allows the latter to reconstruct the desired function to within the desired fidelity level.

The above routing scheme necessitates that every terminal l, $l=1,\ldots,L$, reads the compressed bit streams from previous terminals in the cascade prior to the compression of its own source. This is reflected through (U_1,\ldots,U_{l-1}) treated not only as decoder side information but also as encoder side information. From a practical viewpoint, treating previous terminals streams as encoder side information improves rates but generally entails additional delays. The following corollary specializes the result of Theorem 1 to the case in which (U_1,\ldots,U_{l-1}) is treated only as decoder side information, i.e., the auxiliary random variables are restricted to satisfy that $U_l - S_l - (U_1,\ldots,U_{l-1})$ forms a Markov chain. We also present an alternate coding scheme that is based on successive Wyner-Ziv coding [24].

Corollary 1. The RD region $\mathcal{R}_{WZR}(D)$ that is achievable with the WZR scheme is given by the set of rate tuples (R_1, \ldots, R_L) satisfying

$$R_l \ge \sum_{i=1}^l I(S_i; U_i | U_1, \dots, U_{i-1}), \quad \text{for} \quad l = 1, \dots, L,$$
 (6)

for some joint pmf $p(s_1, \ldots, s_L) \prod_{l=1}^L p(u_l|s_l)$ and function g s.t. $E[d(Z, g(U_1, \ldots, U_L))] \leq D$.

Remark 3. The auxiliary random variables (U_1, \ldots, U_L) that are involved in (6) satisfy the following Markov Chains

$$U_l - S_l - (S_{\mathcal{L}/l}, U_{\mathcal{L}/l}), \quad for \quad l = 1, \dots, L,$$
 (7)

where
$$S_{\mathcal{L}/l} = (S_1, \dots, S_{l-1}, S_{l+1}, \dots, S_L)$$
 and $U_{\mathcal{L}/l} = (U_1, \dots, U_{l-1}, U_{l+1}, \dots, U_L)$.

Outline Proof: The proof of Corollary 1 follows by applying successively standard Wyner-Ziv source coding [24]. Hereafter, we only outline the main steps, for the sake of brevity. Fix $\epsilon > 0$ and a joint pmf $p(s_1, \ldots, s_L, u_1, \ldots, u_L)$ that factorizes as

$$p(s_1, \dots, s_L, u_1, \dots, u_L) = p(s_1, \dots, s_L) \prod_{l=1}^L p(u_l | s_l)$$
(8)

and a function $g(\cdot)$ such that $\mathrm{E}[d(Z,g(U_1,\ldots,U_L))] \leq D/(1+\epsilon)$. Also, fix non-negative R_1,\ldots,R_L , such that for $R_l\geq R_1+\cdots+R_{l-1}$ for $l=1,\ldots,L$.

Codebook generation: Let non-negative $\hat{R}_1, \ldots, \hat{R}_L$, and set $B_l = R_l - (R_1 + \cdots + R_{l-1})$ for $l = 1, \ldots, L$. Generate L codebooks $\{\mathcal{C}_l\}$, $l = 1, \ldots, L$, with codebook \mathcal{C}_l consisting of a collection of $2^{n(B_l + \hat{R}_l)}$ independent codewords $\{u_l^n(i_l)\}$ indexed with $i_l = 1, \ldots, 2^{n(B_l + \hat{R}_l)}$, where codeword $u_l^n(i_l)$ has its elements generated i.i.d. according to $p(u_l)$. Randomly and independently assign these codewords into 2^{nB_l} bins $\{\mathcal{B}_{j_l}\}$ indexed with $j_l = 1, \ldots, 2^{nB_l}$, each containing $2^{n\hat{R}_l}$ codewords.

Encoding at Terminal l: Terminal l finds an index i_l such that $u_l^n(i_l) \in \mathcal{C}_l$ is strongly ϵ -jointly typical¹ with s_l^n , i.e., $(u_l^n(i_l), s_l^n) \in \mathcal{T}_{[U_lS_l]}^{(n)}$. Using standard arguments, it is easy to see that this can be accomplished with vanishing probability of error as long as n is large and

$$B_l + \hat{R}_l \ge I(S_l; U_l). \tag{9}$$

Let j_l such that $\mathcal{B}_{j_l} \ni u_l^n(i_l)$. Terminal l then forwards the bin index j_l and the received message $m_{l-1} = (j_1, \dots, j_{l-1})$ to Terminal l as $m_l = (m_{l-1}, j_l)$.

Reconstruction at the end Terminal (L+1): Terminal (L+1) collects all received bin indices as $m_L = (j_1, \ldots, j_L)$, and reconstructs the codewords $u_1^n(i_1), \ldots, u_L^n(i_L)$ successively in this order, as follows. Assuming that codewords $(u_1^n(i_1), \ldots, u_{l-1}^n(i_{l-1}))$ have been reconstructed correctly, it finds the appropriate codeword $u_l^n(i_l)$ by looking in the bin \mathcal{B}_{j_l} for the unique $u_l^n(i_l)$ that is

¹For formal definitions of strongly ϵ -joint typicality, the reader may refer to [23].

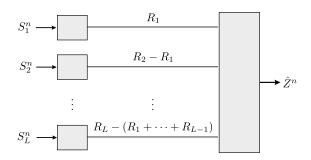


Fig. 3: Distributed source coding model for the routing scheme of Corollary 1.

 ϵ -jointly typical with $(u_1^n(i_1), \dots, u_{l-1}^n(i_{l-1}))$. Using standard arguments, it is easy to see that the error in this step has vanishing probability as long as n is large and

$$\hat{R}_l < I(U_l; U_1, \dots, U_{l-1}). \tag{10}$$

Terminal (L+1) reconstructs an estimate of Z^n sample-wise as $\hat{Z}_i = g(u_{1,i}(i_1), u_{2,i}(i_2), \dots, u_{L,i}(i_L)),$ $i=1,\dots,n$. Note that in doing so the average distortion constraint is satisfied. Finally, substituting $B_l = R_l - (R_1 + \dots + R_{l-1})$ and combining (9) and (10), we get (6); and

this completes the proof of Corollary 1. \Box

Remark 4. Note that the rate constraints in Theorem 1 and Corollary 1 are identical. However, $\mathcal{R}_{WZR}(D) \subseteq \mathcal{R}_{IR}(D)$ since the set of feasible pmfs in IR is larger than that in WZR.

Remark 5. As it can be conveyed from the proof of Corollary 1, since every Terminal l, l = 1, ..., L, uses a part $(R_1 + ... + R_{l-1})$ of its per-sample rate R_l to simply route the bit streams received from the previous terminals in the cascade and the remaining per-sample $B_l = R_l - (R_1 + ... + R_{l-1})$ bits to convey a description of its observed source S_l^n , the resulting scheme can be seen as one for the model shown in Figure 3 in which the terminals are connected through parallel links to the CP. Using this connection, the performance of the above WZR scheme can be further improved by compressing the observations à-la Berger-Tung [25].

Remark 6. In accordance with Remark 2, for the model of Figure 1 yet another natural coding strategy is one in which one decomposes the problem into L successive Wyner-Ziv type problems for function computation, one for each hop. Specifically, in this strategy one sees the

communication between Terminal l and Terminal (l+1), $l=1,\ldots,L$, as a Wyner-Ziv source coding problem with two-sided state information, state information S_l^n at the encoder and state information S_{l+1}^n at the decoder. This strategy, which is not of "routing type", is developed in the next section.

B. In-Network Processing (IP)

In the routing schemes in Section III-A, the function of interest is computed at the destination from the compressed observations, i.e., the terminals have to share the fronthaul to send a compressed version of their observations to Terminal (L+1). We present a scheme to which we refer to as "In-Network Processing" (IP), in which instead, each terminal computes a part of the function to reconstruct at the decoder so that the function of interest is computed along the cascade. To that end, each terminal decompresses the signal received from the previous terminal and jointly compresses it with its observation to generate an estimate of the part of the function of interest, which is forwarded to the next terminal (see Section IV-C). Correlation between the computed part of the function and the source at the next terminal S_{l+1}^n is exploited through Wyner-Ziv coding. Note that by decompressing and recompressing the observations at each terminal, additional distortion is introduced [2].

Theorem 2. The RD region $\mathcal{R}_{IP}(D)$ that is achievable with IP is given by the union of rate tuples (R_1, \ldots, R_L) satisfying

$$R_l \ge I(S_l, U_{l-1}; U_l | S_{l+1}), \quad l = 1, \dots, L,$$
 (11)

for some joint pmf $p(s_1, ..., s_L) \prod_{l=1}^L p(u_l|s_l, u_{l-1})$ and a function g, such that $\hat{Z} = g(U_L)$ and $E[d(Z, \hat{Z})] \leq D$.

Remark 7. The auxiliary random variables (U_1, \ldots, U_L) that are involved in (11) satisfy the following Markov Chains

$$U_l \rightarrow (S_l, U_{l-1}) \rightarrow (S_{\mathcal{L}/l}, U_{\mathcal{L}/\{l-1,l\}}) \quad for \quad l = 1, \dots, L,$$
 (12)

where
$$S_{\mathcal{L}/l} = (S_1, \dots, S_{l-1}, S_{l+1}, \dots, S_L), \ U_{\mathcal{L}/\{l-1,l\}} = (U_1, \dots, U_{l-2}, U_{l+1}, \dots, U_L).$$

Outline Proof: Fix $\epsilon > 0$ and a joint pmf $p(s_1, \ldots, s_L, u_1, \ldots, u_L)$ that factorizes as

$$p(s_1, \dots, s_L, u_1, \dots, u_L) = p(s_1, \dots, s_L) \prod_{l=1}^L p(u_l | s_l, u_{l-1}),$$
(13)

and a function $g(\cdot)$ such that $\mathrm{E}[d(Z;g(U_L))] \leq D/(1+\epsilon)$. Also fix non-negative R_1,\ldots,R_L . Codebook generation: Let non-negative $\hat{R}_1,\ldots,\hat{R}_L$. Generate L codebooks $\{\mathcal{C}_l\}$, $l=1,\ldots,L$, with codebook \mathcal{C}_l consisting of a collection of $2^{n(R_l+\hat{R}_l)}$ independent codewords $\{u_l^n(i_l)\}$, indexed with $i_l=1,\ldots,2^{n(R_l+\hat{R}_l)}$, where codeword $u_l^n(i_l)$ has its elements generated randomly and independently i.i.d. according to $p(u_l)$. Randomly and independently assign these codewords into 2^{nR_l} bins $\{\mathcal{B}_{j_l}\}$ indexed with $j_l=1,\ldots,2^{nR_l}$, each containing $2^{n\hat{R}_l}$ codewords.

Encoding at Terminal 1: Terminal 1 finds an index i_1 such that $u_1^n \in \mathcal{C}_1$ is strongly ϵ -jointly typical with s_1^n , i.e., $(u_1^n(i_1), s_1^n) \in \mathcal{T}^{(n)}_{[U_1S_1]}$. Using standard arguments, it is easy to see that this can be accomplished with vanishing probability of error as long as n is large and

$$R_1 + \hat{R}_1 > I(S_1; U_1).$$
 (14)

Let j_1 such that $\mathcal{B}_{j_1} \ni u_1^n(i_1)$. Terminal 1 then forwards the index j_1 to Terminal 2. Decompression and encoding at Terminal $l \ge 2$: Upon reception of the bin index $m_{l-1} = j_{l-1}$ from Terminal (l-1), Terminal l finds $u_{l-1}^n(i_{l-1})$ by looking in the bin \mathcal{B}_{j_l} for the the unique $u_{l-1}^n(i_l)$ that is ϵ -jointly typical with s_l^n . Using standard arguments, it can be seen that this can be accomplished with vanishing probability of error as long as n is large enough and

$$\hat{R}_{l-1} < I(U_{l-1}; S_l). \tag{15}$$

Then, Terminal l finds an index i_l such that $u_l^n(i_l) \in \mathcal{C}_l$ is strongly ϵ -jointly typical with $(s_l^n, u_{l-1}^n(i_{l-1}))$, i.e., $(s_l^n, u_{l-1}^n(i_{l-1}), u_l^n(i_l)) \in \mathcal{T}_{[SU_{l-1}U_l]}^{(n)}$. Using standard arguments, it can be seen that this can be accomplished with vanishing probability of error as long as n is large and

$$R_l + \hat{R}_l \ge I(S_l, U_{l-1}; U_l).$$
 (16)

Let j_l such that $\mathcal{B}_{j_l} \ni u_l^n(i_l)$. Terminal l forwards the bin index to Terminal (l+1) as $m_l = j_l$. Reconstruction at end Terminal (L+1): Terminal (L+1) collects the bin index $m_L = j_L$ and

reconstructs the codeword $u_L^n(i_L)$ by looking in the bin \mathcal{B}_{j_L} . Since Terminal (L+1) does not have available any side information sequence, from (15), successful recovery of the unique $u_L^n(i_L)$ in the bin \mathcal{B}_{j_L} requires $\hat{R}_L = 0$. That is, each bin contains a single codeword and $j_L = i_L$. Then, Terminal (L+1) reconstructs an estimate of Z^n sample-wise as $\hat{Z}_i = g(u_{L,i}(i_L))$, $i = 1, \ldots, n$. In doing so, the average distortion constraint is satisfied.

Finally, combining (14), (15) and (16), we get (11). This completes the proof of Theorem 2. \Box

Remark 8. It is shown in [2] that for L=2, in general none of the IR and IP schemes outperform the other; and a scheme combining the two strategies is proposed.

IV. CENTRALIZED AND DISTRIBUTED BEAMFORMING IN CHAINED MIMO SYSTEMS

In this section, we apply the cascade source coding model to study the achievable distortion in a Gaussian uplink MIMO system with a chained MIMO architecture (C-MIMO) in which M single antenna users transmit over a Gaussian channel to L RRUs as shown in Figure 2. The signal received at RRU l, l = 1, ..., L, equipped with K antennas, $\mathbf{S}_l \in \mathbb{C}^{K \times 1}$, is given by

$$\mathbf{S}_l = \mathbf{H}_l \mathbf{X} + \mathbf{N}_l, \tag{17}$$

where $\mathbf{X} = [X_1, \dots, X_M]^T$ is the signal transmitted by the M users and $X_m \in \mathbb{C}$, $m = 1, \dots, M$ is the signal transmitted by user m. We assume that each user satisfies an average power constraint $\mathrm{E}[|X_m|^2] \leq \mathrm{snr}, \ m = 1, \dots, M$, where $\mathrm{snr} > 0$; $\mathbf{H}_l \in \mathbb{C}^{K \times M}$ is the channel between the M users and RRU l and $\mathbf{N}_l \in \mathbb{C}^{K \times 1} \sim \mathcal{CN}(\mathbf{0}, \mathbf{I})$ is the additive ambient noise.

The transmitted signal by the M users is assumed to be distributed as $\mathbf{X} \sim \mathcal{CN}(\mathbf{0}, \mathrm{snr}\mathbf{I})$ and we denote the observations at the L RRUs as $\mathbf{S} = [\mathbf{S}_1^T, \dots, \mathbf{S}_L^T]^T$. Thus, we have $\mathbf{S} \sim \mathcal{CN}(\mathbf{0}, \mathbf{\Sigma_s})$, where $\mathbf{\Sigma_s} = \mathrm{snr}\mathbf{H}_{\mathcal{L}}\mathbf{H}_{\mathcal{L}}^H + \mathbf{I}$ and $\mathbf{H}_{\mathcal{L}} \triangleq [\mathbf{H}_1^T, \dots, \mathbf{H}_L^T]^T$.

In traditional receive-beamforming, a beamforming filter $\mathbf{W} \in \mathbb{C}^{M \times L \cdot K}$ is applied at the decoder on the received signal \mathbf{S} to estimate the channel input \mathbf{X} with the linear function

$$\mathbf{Z} \triangleq \mathbf{WS}.$$
 (18)

In C-MIMO, the decoder (the CP) is interested in computing the receive beamforming signal (18) with minimum distortion, although S is not directly available at the CP but remotely observed at the terminals. Depending on the available CSI, receive-beamforming computation may be better performed centrally at the CP or distributively across the RRUs:

Centralized Beamforming: If CSI is available only at the CP, not at the RRUs, it seems reasonable that beamforming operations are performed only centrally at the CP. In this case, RRU l, l = 1, ..., L, sends a compressed version $\hat{\mathbf{S}}_l$ of its output signal \mathbf{S}_l to the CP, which first collects the vector $(\hat{\mathbf{S}}_l, ..., \hat{\mathbf{S}}_L)$, and then performs receive-beamforming on it.

Distributed Beamforming: If local CSI is available at the RRUs, or can be acquired, receive beamforming operations can be performed distributively along the cascade. Due to linearity the joint beamforming operation (18) can be expressed as a function of the received source as

$$\mathbf{Z} = \mathbf{W}\mathbf{S} = \mathbf{W}_1\mathbf{S}_1 + \dots + \mathbf{W}_L\mathbf{S}_L,\tag{19}$$

where $\mathbf{W}_l \in \mathbb{C}^{M \times K}$ corresponds to blocks of K columns of \mathbf{W} such that $[\mathbf{W}_1, \dots, \mathbf{W}_L] = \mathbf{W}$. In this case, the receive beamforming signal can be computed gradually in the cascade network, by letting the RRUs compute a part of the desired function, e.g., as proposed in Section IV-C, RRU $l, l = 1, \dots, L$ computes an estimate of $\mathbf{W}_1\mathbf{S}_1 + \dots + \mathbf{W}_l\mathbf{S}_l$.

The distortion between ${\bf Z}$ and the reconstruction of the beamforming signal $\hat{{\bf Z}}$ at the CP is measured with the sum-distortion

$$d(\mathbf{Z}, \hat{\mathbf{Z}}) \triangleq \operatorname{Tr}\{(\mathbf{Z} - \hat{\mathbf{Z}})(\mathbf{Z} - \hat{\mathbf{Z}})^{H}\}.$$
 (20)

For a given fronthaul tuple (R_1, \ldots, R_L) in the RD region $\mathcal{R}(D)$, the minimum achievable average distortion D is characterized by the distortion-rate function² given by

$$D(R_1, \dots, R_L) \triangleq \min\{D \ge 0 : (R_1, \dots, R_L) \in \mathcal{R}(D)\}.$$
(21)

²This formulation is equivalent to the rate-distortion framework considered in Section III; here we consider the distortion-rate formulation for convenience.

Next, we study the distortion-rate function in a Gaussian C-MIMO model under centralized and distributed beamforming with the schemes proposed for the cascade source coding problem.

A. Centralized Beamforming with Improved Routing

In this section, we consider distortion-rate function of the IR scheme in Section III-A applied for centralized beamforming. Each RRU forwards a compressed version of the observation to the CP, which estimates the receive-beamforming signal **Z** from the decompressed observations. While the optimal test channels are in general unknown, next theorem gives the distortion-rate function of IR for centralized beamforming for the C-MIMO setup under jointly distributed Gaussian test channels.

Theorem 3. The distortion-rate function for the IR scheme under jointly Gaussian test channels is given by

$$D_{\rm IR}(R_1,\ldots,R_L) = \min_{\mathbf{K}_1,\ldots,\mathbf{K}_L} \operatorname{Tr}\{\mathbf{\Sigma}_{\mathbf{z}} - \mathbf{\Sigma}_{\mathbf{z},\mathbf{u}_{\mathcal{L}}} \mathbf{\Sigma}_{\mathbf{u}_{\mathcal{L}}}^{-1} \mathbf{\Sigma}_{\mathbf{z},\mathbf{u}_{\mathcal{L}}}^{H}\}$$
(22)

s.t.
$$R_l \ge B_1 + \ldots + B_l$$
, $l = 1, \ldots, L$, (23)

$$B_l \triangleq \log |\mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}} + \mathbf{K}_l|/|\mathbf{K}_l|, \tag{24}$$

where
$$\Sigma_{\mathbf{s}_{l}|\mathbf{u}_{\mathcal{L}_{l-1}}} = \Sigma_{\mathbf{s}_{l}} - \Sigma_{\mathbf{s}_{l},\mathbf{u}_{\mathcal{L}_{l-1}}} \Sigma_{\mathbf{u}_{\mathcal{L}_{l-1}}}^{-1} \Sigma_{\mathbf{s}_{l},\mathbf{u}_{\mathcal{L}_{l-1}}}^{H}$$
 and $\Sigma_{\mathbf{s}_{l}} = \delta_{l} \Sigma_{\mathbf{s}} \delta_{l}^{T}$, $\Sigma_{\mathbf{s}_{l},\mathbf{u}_{\mathcal{L}_{l-1}}} = \delta_{l} \Sigma_{\mathbf{s}} \bar{\mathbf{I}}_{l}^{T}$, $\Sigma_{\mathbf{u}_{\mathcal{L}_{l-1}}} = \delta_{l} \Sigma_{\mathbf{s}} \bar{\mathbf{I}}_{l}^{T}$

Proof: We evaluate Theorem 1 by considering jointly Gaussian sources and test channels $(\mathbf{S}_1, \dots, \mathbf{S}_L, \mathbf{U}_1, \dots, \mathbf{U}_L)$ satisfying $p(\mathbf{s}_1, \dots, \mathbf{s}_L) \prod_{l=1}^L p(\mathbf{u}_l | \mathbf{s}_l, \mathbf{u}_1, \dots, \mathbf{u}_{l-1})$ and the minimum mean square error (MMSE) estimator $\hat{\mathbf{Z}} = \mathrm{E}[\mathbf{Z}|\mathbf{U}_{\mathcal{L}}]$ as reconstruction function g, where we define $\mathbf{U}_{\mathcal{L}_{l-1}} \triangleq [\mathbf{U}_1, \dots, \mathbf{U}_{l-1}]$ and $\mathbf{U}_{\mathcal{L}} \triangleq \mathbf{U}_{\mathcal{L}_L}$. Note that MMSE reconstruction is optimal under (20), while considering jointly Gaussian test channels might be suboptimal in general. First we derive a lower bound on the achievable distortion. We have

$$B_l \triangleq I(\mathbf{S}_l; \mathbf{U}_l | \mathbf{U}_{\mathcal{L}_{l-1}}) \tag{25}$$

$$=I(\mathbf{J}_l;\mathbf{U}_l|\mathbf{U}_{\mathcal{L}_{l-1}}) \tag{26}$$

$$=I(\mathbf{J}_l;\mathbf{U}_{\mathcal{L}_{l-1}},\mathbf{U}_l) \tag{27}$$

$$\geq I(\mathbf{J}_l; \mathbf{U}_l),$$
 (28)

where in (26) we define the MMSE error $\mathbf{J}_l \triangleq \mathbf{S}_l - \mathrm{E}[\mathbf{S}_l | \mathbf{U}_{\mathcal{L}_{l-1}}]$, which is Gaussian distributed as $\mathbf{J}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{\Sigma}_{\mathbf{s}_l | \mathbf{u}_{\mathcal{L}_{l-1}}})$; (27) follows due to the orthogonality principle [23], and due to the fact that for Gaussian random variables, orthogonality implies independence of \mathbf{J}_l and $\mathbf{U}_{\mathcal{L}_{l-1}}$.

For the fixed test channels, let us choose matrix $\mathbf{K}_l \succeq \mathbf{0}$ such that for $l = 1, \dots, L$

$$cov(\mathbf{J}_{l}|\mathbf{U}_{l}) = \mathbf{\Sigma}_{\mathbf{s}_{l}|\mathbf{u}_{\mathcal{L}_{l-1}}}^{1/2} \mathbf{K}_{l}^{1/2} (\mathbf{\Sigma}_{\mathbf{s}_{l}|\mathbf{u}_{\mathcal{L}_{l-1}}} + \mathbf{K}_{l})^{-1} \mathbf{K}_{l}^{1/2} \mathbf{\Sigma}_{\mathbf{s}_{l}|\mathbf{u}_{\mathcal{L}_{l-1}}}^{1/2}.$$
(29)

Note that such \mathbf{K}_l always exists since $\mathbf{0} \leq \mathrm{cov}(\mathbf{J}_l|\mathbf{U}_l) \leq \mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}}$, and can be found explicitly as follows. After some straightforward algebraic manipulations, (29) can be written as $\mathbf{K}_l^{1/2}\mathbf{A}\mathbf{K}_l^{1/2,H} = \mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}}$, where $\mathbf{A} \triangleq \mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}}^{1/2} \mathrm{cov}^{-1}(\mathbf{J}_l|\mathbf{U}_l)\mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}}^{1/2} - \mathbf{I}$. Note that $\mathbf{A} \succeq \mathbf{0}$, and let $\mathbf{A} = \mathbf{V}\mathbf{\Lambda}_A\mathbf{V}^H$ and $\mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}} = \mathbf{V}'\mathbf{\Lambda}_{\Sigma_l}\mathbf{V}'^H$. Then, it follows that \mathbf{K}_l is given by $\mathbf{K}_l = \mathbf{K}_l^{1/2}\mathbf{K}_l^{1/2,H}$, where $\mathbf{K}_l^{1/2} = \mathbf{V}'\mathbf{\Lambda}_K^{1/2}\mathbf{V}^H$ and $\mathbf{\Lambda}_K = \mathbf{\Lambda}_{\Sigma_l}/\mathbf{\Lambda}_A$. Then, from (28), we have

$$B_l \ge h(\mathbf{J}_l) - h(\mathbf{J}_l|\mathbf{U}_l) \tag{30}$$

$$\geq \log |\mathbf{\Sigma}_{\mathbf{s}_{l}|\mathbf{u}_{\mathcal{L}_{l-1}}}| - \log |\operatorname{cov}(\mathbf{J}_{l}|\mathbf{U}_{l})| \tag{31}$$

$$= \log |\mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}} + \mathbf{K}_l| - \log |\mathbf{K}_l|. \tag{32}$$

The distortion is lower bounded as

$$D \ge \operatorname{Tr}\{E[(\mathbf{Z} - E[\mathbf{Z}|\mathbf{U}_{\mathcal{L}}])(\mathbf{Z} - E[\mathbf{Z}|\mathbf{U}_{\mathcal{L}}])^{H}]\} = \operatorname{Tr}\{\Sigma_{\mathbf{z}} - \Sigma_{\mathbf{z},\mathbf{u}_{\mathcal{L}}}\Sigma_{\mathbf{u}_{\mathcal{L}}}^{-1}\Sigma_{\mathbf{z},\mathbf{u}_{\mathcal{L}}}^{H}\},$$
(33)

where (33) follows due to the linearity of the MMSE estimator for jointly Gaussian variables.

The lower bound given by (32) and (33) is achievable by letting $\mathbf{U}_l = \mathbf{S}_l + \mathbf{Q}_l$, with $\mathbf{Q}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{K}_l)$, and independent of all other variables, as follows

$$B_l = I(\mathbf{S}_l; \mathbf{U}_l | \mathbf{U}_{\mathcal{L}_{l-1}}) \tag{34}$$

$$= h(\mathbf{U}_l|\mathbf{U}_{\mathcal{L}_{l-1}}) - h(\mathbf{U}_l|\mathbf{U}_{\mathcal{L}_{l-1}}, \mathbf{S}_l)$$
(35)

$$= h(\mathbf{S}_l - \mathrm{E}[\mathbf{S}_l | \mathbf{U}_{\mathcal{L}_{l-1}}] + \mathbf{Q}_l | \mathbf{U}_{\mathcal{L}_{l-1}}) - h(\mathbf{Q}_l)$$
(36)

$$= h(\mathbf{S}_l - \mathbf{E}[\mathbf{S}_l | \mathbf{U}_{\mathcal{L}_{l-1}}] + \mathbf{Q}_l) - h(\mathbf{Q}_l)$$
(37)

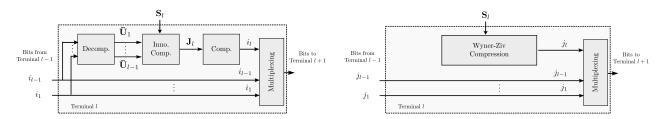


Fig. 4: Improved Routing scheme for C-MIMO. Fig. 5: Wyner-Ziv Routing scheme for C-MIMO.

$$= \log |\mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}} + \mathbf{K}_l| - \log |\mathbf{K}_l|, \tag{38}$$

where (36) follows since $\mathbf{U}_l = \mathbf{S}_l + \mathbf{Q}_l$ and (37) is due to the orthogonality principle. In the case $\text{cov}(\mathbf{J}_l|\mathbf{U}_l) = \mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_{l-1}}}$, we have $B_l = 0$ which can be trivially achieved by letting $\mathbf{U}_l = \emptyset$. Optimizing over the positive semidefinite covariance matrices $\mathbf{K}_1, \dots, \mathbf{K}_L \succeq \mathbf{0}$ gives the

desired minimum distortion D in Theorem 3. This completes the proof. \Box The IR scheme in Section III-A requires joint compression at RRU l of the observed source \mathbf{S}_l and the previous compression codewords $\mathbf{U}_1, \ldots, \mathbf{U}_{l-1}$ to generate the compression codeword

 \mathbf{U}_l . However, for the Gaussian C-MIMO, it is shown next that the sum-distortion D_{IR} in Theorem 3 can also be achieved by applying at each RRU separate decompression of the previous compression codewords, the innovation sequence computation \mathbf{J}_l , followed by independent compression of \mathbf{J}_l into a codeword $\bar{\mathbf{U}}_l$, which is independent of the previous compression codewords $\bar{\mathbf{U}}_1, \ldots, \bar{\mathbf{U}}_{l-1}$, as follows. See Figure 4. At RRU l:

- Upon receiving bits m_{l-1} , decompress $\bar{\mathbf{U}}_1,\ldots,\bar{\mathbf{U}}_{l-1}$.
- Compute the innovation sequence $\mathbf{J}_l \triangleq \mathbf{S}_l \mathrm{E}[\mathbf{S}_l|\bar{\mathbf{U}}_1,\ldots,\bar{\mathbf{U}}_{l-1}].$
- Compress \mathbf{J}_l at B_l bits per sample independently of $\bar{\mathbf{U}}_1, \dots, \bar{\mathbf{U}}_{l-1}$ using a codeword $\bar{\mathbf{U}}_l$, where $\bar{\mathbf{U}}_l = \mathbf{J}_l + \mathbf{Q}_l$, with $\mathbf{Q}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{K}_l)$ independent of each other.

Note that \mathbf{J}_l corresponds to the MMSE error of estimating \mathbf{S}_l from $(\bar{\mathbf{U}}_1, \dots, \bar{\mathbf{U}}_{l-1})$, and is an i.i.d. zero-mean Gaussian sequence distributed as $\mathbf{J}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{\Sigma}_{\mathbf{j},l})$.

Proposition 1. For the Gaussian C-MIMO model, separate decompression, innovation computation and independent innovation compression achieves the minimum distortion $D_{IR}(R_1, \ldots, R_L)$ characterized by the distortion-rate function in Theorem 3.

Proof: We show that any distortion D achievable for a pmf $p(\mathbf{s}_1, \dots, \mathbf{s}_L) \prod_{l=1}^L p(\mathbf{u}_l | \mathbf{s}_l, \mathbf{u}_1, \dots, \mathbf{u}_{l-1})$ and the corresponding (B_1, \dots, B_L) in Theorem 3 is also achievable with separate decompression, innovation computation and compression as detailed above. From standard arguments, compressing \mathbf{J}_l at B_l bits requires

$$B_l \ge I(\mathbf{J}_l; \bar{\mathbf{U}}_l) \tag{39}$$

$$= h(\bar{\mathbf{U}}_l) - h(\bar{\mathbf{U}}_l|\mathbf{J}_l) \tag{40}$$

$$= \log |\mathbf{\Sigma}_{\mathbf{s}_l|\mathbf{u}_{\mathcal{L}_l}} + \mathbf{K}_l| - \log |\mathbf{K}_l|, \tag{41}$$

where (41) follows since $\Sigma_{\mathbf{j},l} = \Sigma_{\mathbf{s}_l | \mathbf{u}_{\mathcal{L}_{l-1}}}$, which follows since RRU l can compute $\mathbf{U}_{l'} = \bar{\mathbf{U}}_{l'} + \mathrm{E}[\mathbf{S}_{l'} | \mathbf{U}_1, \dots, \mathbf{U}_{l'-1}]$ for $l' = 1, \dots, l$, which is distributed as the test channels $\mathbf{U}_{l'} = \mathbf{S}_{l'} + \mathbf{Q}_{l'}$ and thus $\mathrm{E}[\mathbf{S}_l | \mathbf{U}_{\mathcal{L}_{l-1}}] = \mathrm{E}[\mathbf{S}_l | \bar{\mathbf{U}}_{\mathcal{L}_{l-1}}]$.

The distortion between ${\bf Z}$ and its estimation from ${ar {\bf U}}_{\cal L}$ satisfies

$$D \ge \operatorname{Tr}\{\mathbb{E}[(\mathbf{Z} - \mathbb{E}[\mathbf{Z}|\bar{\mathbf{U}}_{\mathcal{L}}])(\mathbf{Z} - \mathbb{E}[\mathbf{Z}|\bar{\mathbf{U}}_{\mathcal{L}}])^{H}]\}$$
(42)

$$= \operatorname{Tr}\{ \operatorname{E}[(\mathbf{Z} - \operatorname{E}[\mathbf{Z}|\mathbf{U}_{\mathcal{L}}])(\mathbf{Z} - \operatorname{E}[\mathbf{Z}|\mathbf{U}_{\mathcal{L}}])^{H}] \}$$
(43)

$$= \operatorname{Tr} \{ \mathbf{\Sigma}_{\mathbf{z}} - \mathbf{\Sigma}_{\mathbf{z}, \mathbf{u}_{\mathcal{L}}} \mathbf{\Sigma}_{\mathbf{u}_{\mathcal{L}}}^{-1} \mathbf{\Sigma}_{\mathbf{z}, \mathbf{u}_{\mathcal{L}}}^{H} \}. \tag{44}$$

Thus, any achievable distortion D for given $p(\mathbf{s}_1, \dots, \mathbf{s}_L) \prod_{l=1}^L p(\mathbf{u}_l | \mathbf{s}_l, \mathbf{u}_1, \dots, \mathbf{u}_{l-1})$ and fixed (B_1, \dots, B_L) in Theorem 3 is achievable by separate decompression, innovation computation and independent compression of the innovation. This completes the proof.

Determining the optimal covariance matrices $(\mathbf{K}_1, \dots, \mathbf{K}_L)$ achieving $D_{\mathrm{IR}}(R_1, \dots, R_L)$ in Theorem 3 requires a joint optimization, which is generally not simple. Next, we propose a method to successively obtain a feasible solution $(\mathbf{K}_1^*, \dots, \mathbf{K}_L^*)$ and the corresponding minimum distortion $D_{\mathrm{IR}-\mathrm{S}}^*(R_1, \dots, R_L)$ for given (R_1, \dots, R_L) :

- 1) For a given fronthaul tuple (R_1, \ldots, R_L) , fix non-negative B_1, \ldots, B_L , satisfying $R_l \ge B_1 + \ldots + B_l$, for $l = 1, \ldots, L$.
- 2) For such (B_1, \ldots, B_L) , sequentially find \mathbf{K}_l^* from RRU 1 to RRU L as the \mathbf{K}_l minimizing the distortion between the innovation \mathbf{J}_l and its reconstruction as follows. At RRU l, for

given $\mathbf{K}_1^*, \dots, \mathbf{K}_{l-1}^*$ and B_l , \mathbf{K}_l^* is found from the covariance matrix \mathbf{K}_l minimizing

$$D_{l}(B_{l}) \triangleq \min_{\mathbf{K}_{l}} \operatorname{Tr} \{ \operatorname{E}[\mathbf{J}_{l} - \operatorname{E}[\mathbf{J}_{l} | \bar{\mathbf{U}}_{l}]) (\mathbf{J}_{l} - \operatorname{E}[\mathbf{J}_{l} | \bar{\mathbf{U}}_{l}])^{H}] \}$$

$$\text{s.t. } B_{l} \geq \log |\mathbf{\Sigma}_{\mathbf{s}_{l} | \mathbf{u}_{\mathcal{L}_{l-1}}} + \mathbf{K}_{l} | / |\mathbf{K}_{l}|.$$

$$(45)$$

Note that (45) corresponds to the distortion-rate problem of compressing a Gaussian vector source $\mathbf{J}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{\Sigma}_{\mathbf{j},l})$ at B_l bits and its solution is given below in Proposition 2.

- 3) Compute the achievable distortion $D_{\mathrm{IF}}^{\mathrm{S}}(B_1,\ldots,B_L)$ by evaluating $\mathrm{Tr}\{\boldsymbol{\Sigma}_{\mathbf{z}}-\boldsymbol{\Sigma}_{\mathbf{z},\mathbf{u}_{\mathcal{L}}}\boldsymbol{\Sigma}_{\mathbf{u}_{\mathcal{L}}}^{-1}\boldsymbol{\Sigma}_{\mathbf{z},\mathbf{u}_{\mathcal{L}}}^{H}\}$ as in Theorem 1 with the chosen covariance matrices $(\mathbf{K}_{1}^{*},\ldots,\mathbf{K}_{L}^{*})$.
- 4) Compute $D_{\mathrm{IR-S}}^*(R_1,\ldots,R_L)$ as the minimum $D_{\mathrm{IF}}^{\mathrm{S}}(B_1,\ldots,B_L)$ over (B_1,\ldots,B_L) satisfying the fronthaul constraints $R_l \geq B_1 + \cdots + B_l$ for $l = 1,\ldots,L$.

The solution for the distortion-rate problem in (45) is standard and given next for completeness.

Proposition 2. Given $\mathbf{K}_1^*, \dots, \mathbf{K}_{l-1}^*$, let $\Sigma_{\mathbf{j},l} = \Sigma_{\mathbf{s}_l | \mathbf{u}_{\mathcal{L}_{l-1}}} = \mathbf{V}_l \mathbf{\Lambda}_J \mathbf{V}_l^H$, where $\mathbf{V}_l^H \mathbf{V}_l = \mathbf{I}$ and $\mathbf{\Lambda}_J \triangleq diag[\lambda_1^J, \dots, \lambda_K^J]$. The optimal distortion (45) is $D_l = \sum_{k=1}^K \min\{\lambda, \lambda_k^J\}$ where $\lambda > 0$ is the solution to

$$B_l = \sum_{k=1}^K \log^+ \left(\frac{\lambda_k^J}{\lambda}\right),\tag{46}$$

and is achieved with $\mathbf{K}_l^* = \mathbf{V}_l \mathbf{\Lambda} \mathbf{V}_l^H$, where $\mathbf{\Lambda} = diag[\lambda_1^Q, \dots, \lambda_K^Q]$ and $\lambda_k^Q = \min\{\lambda, \lambda_k^J\}/(\lambda_k^J - \min\{\lambda, \lambda_k^J\})^3$.

Outline Proof: The minimization of the RD problem in (45) is standard, e.g. [26], and well known to be achieved by uncorrelating the vector source \mathbf{J}_l into K uncorrelated components as $\mathbf{J}_l' = \mathbf{V}^H \mathbf{J}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{\Lambda}_J)$. Then, the available B_l bits are distributed over the K parallel source components \mathbf{J}_l' by solving the reverse water-filling problem

$$D_l = \min_{d_1, \dots, d_K \ge 0} \sum_{k=1}^K d_k \qquad \text{s.t. } B_l = \sum_{k=1}^K \log^+ \left(\frac{\lambda_k^J}{d_k}\right),$$

³Note the slight abuse of notation. If for the k-th uncorrelated components we have $\lambda \leq \lambda_k^J$, in the achievability we have $\lambda_k^Q = \infty$. It should be understood that the k-th component is not assigned any bit for compression. This is in line with (46) as the number of bits assigned for the k-th component is given by $\log^+\left(\lambda_k^J/\lambda\right) = 0$.

The solution to this problem is given by $d_k = \min\{\lambda_k^J, \lambda\}$, where $\lambda > 0$ satisfies (46). The optimality of \mathbf{K}_l^* follows since D_l is achieved with \mathbf{K}_l^* as stated in Proposition 2 [26], [27].

B. Centralized Beamforming with Successive Wyner-Ziv

In this section, we consider the distortion-rate function of the WZR scheme in Corollary 1 for centralized beamforming. Similarly to IR, each RRU forwards a compressed version of its observation to the CP, which estimates the receive-beamforming signal Z from the decompressed observations. Next theorem shows that WZR achieves the same distortion-rate function as the IR scheme under jointly Gaussian test channels.

Theorem 4. The distortion-rate function of the WZR scheme $D_{WZR}(R_1, ..., R_L)$ with jointly Gaussian test channels, is the same as the distortion-rate function of the IR scheme with Gaussian test channels in Theorem 3, i.e., $D_{WZR}(R_1, ..., R_L) = D_{IR}(R_1, ..., R_L)$

Outline Proof: Since $\mathcal{R}_{\text{WZR}}(D) \subseteq \mathcal{R}_{\text{IR}}(D)$, we only need to show that any distortion D achievable with IR in Theorem 3 is also achievable with WZR. For fixed (B_1, \ldots, B_L) and $p(\mathbf{s}_1, \ldots, \mathbf{s}_L) \prod_{l=1}^L p(\mathbf{u}_l | \mathbf{s}_l, \mathbf{u}_1, \ldots, \mathbf{u}_{l-1})$ with IR in Theorem 3, the minimum distortion is achieved by considering a test channel $\mathbf{U}_l = \mathbf{S}_l + \mathbf{Q}_l$. Since this test channel is also in the class of test channels $\prod_{l=1}^L p(\mathbf{u}_l | \mathbf{s}_l)$ of WZR, it follows that any achievable distortion D for fixed (B_1, \ldots, B_L) and $p(\mathbf{s}_1, \ldots, \mathbf{s}_L) \prod_{l=1}^L p(\mathbf{u}_l | \mathbf{s}_l, \mathbf{u}_1, \ldots, \mathbf{u}_{l-1})$ in Theorem 3 is achievable with WZR.

C. In-Network Processing for Distributed Beamforming

In this section, we study the distortion-rate function of the IP scheme in Section III-B for distributed beamforming. At each RRU, the received signal from the previous terminal is jointly compressed with the observation and forwarded to the next RRU. While the optimal joint compression per RRU along the cascade remains an open problem, even for independent observations [23], we propose to gradually compute the desired function \mathbf{Z} by reconstructing at each RRU parts of \mathbf{Z} . In particular, compression at RRU l-1 is designed such that RRU l reconstructs from \mathbf{S}_l and the received bits an estimate of the part of the function:

$$\mathbf{Z}_l \triangleq \mathbf{W}_1 \mathbf{S}_1 + \dots + \mathbf{W}_l \mathbf{S}_l. \tag{47}$$

The design of the compression is done successively. Assuming $\mathbf{U}_{\mathcal{L}_{l-1}}^* \triangleq (\mathbf{U}_1^*, \dots, \mathbf{U}_{l-1}^*)$ are fixed, at RRU l, \mathbf{U}_l^* , is obtained as the solution to the following distortion-rate problem:

$$D_l(R_l) = \min_{p(\mathbf{u}_l|\mathbf{s}_l, \mathbf{u}_{l-1}^*)} \operatorname{Tr}\{\mathbb{E}[(\mathbf{Z}_l - \hat{\mathbf{Z}}_l)(\mathbf{Z}_l - \hat{\mathbf{Z}}_l)^H]\}$$
(48)

s.t.
$$R_l \ge I(\mathbf{S}_l, \mathbf{U}_{l-1}^*; \mathbf{U}_l | \mathbf{S}_{l+1}).$$
 (49)

Problem (48)-(49) corresponds to the distortion-rate function of the Wyner-Ziv type source coding problem of lossy reconstruction of function \mathbf{Z}_l as $\hat{\mathbf{Z}}_l$, which is a function of the encoder observation \mathbf{S}_l , \mathbf{U}_{l-1}^* when side information \mathbf{S}_{l+1} , is available at the decoder [22]. Proposition 3 given below characterizes the optimal test channel at RRU l given $\mathbf{U}_{\mathcal{L}_{l-1}}^*$, i.e., \mathbf{U}_l^* , and shows that it is Gaussian distributed as,

$$\mathbf{U}_l^* = \mathbf{P}_l[\mathbf{U}_{l-1}^*; \mathbf{W}_l \mathbf{S}_l]^H + \mathbf{Q}_l, \tag{50}$$

where $\mathbf{P}_l = [\mathbf{P}_l^U, \mathbf{P}_l^S]$ and $\mathbf{Q}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{K}_l)$.

Remark 9. Operationally, (50) indicates that the optimal codeword \mathbf{U}_{l}^{*} at RRU l can be obtained by compressing a linear combination of the decompressed signal \mathbf{U}_{l-1}^{*} and a beamformed version of the observation $\mathbf{W}_{l}\mathbf{S}_{l}$ with the linear combination \mathbf{P}_{l} . This is exploited below in Proposition 4.

On the other hand, let $\Pi^U_{l,l'} = \mathbf{P}^U_l \cdots \mathbf{P}^U_{l'}$ and the quantization noises $\bar{\mathbf{Q}} = [\mathbf{Q}^T_1, \dots, \mathbf{Q}^T_L]^T$,

$$\bar{\mathbf{P}}_{l}^{S} \triangleq \left[\mathbf{\Pi}_{l,2}^{U} \mathbf{P}_{1}^{S} \mathbf{W}_{1}, \dots, \mathbf{\Pi}_{l,l}^{U} \mathbf{P}_{l-1}^{S} \mathbf{W}_{l-1}, \mathbf{P}_{l}^{S} \mathbf{W}_{l}, \mathbf{0}, \dots, \mathbf{0} \right],$$

$$\bar{\mathbf{P}}_{l}^{Q} \triangleq \left[\mathbf{\Pi}_{l,2}^{U} \mathbf{P}_{1}^{S}, \mathbf{\Pi}_{l,3}^{U} \mathbf{P}_{2}^{S}, \dots, \mathbf{P}_{l}^{U} \mathbf{P}_{l-1}^{S}, \mathbf{0}, \mathbf{0}, \dots, \mathbf{0} \right].$$
(51)

Then, we can write \mathbf{U}_l^* in (50) as $\mathbf{U}_l^* = \mathbf{R}_l + \mathbf{Q}_l$ where

$$\mathbf{R}_{l} \triangleq [\mathbf{P}_{l}^{U}, \mathbf{P}_{l}^{S}][\mathbf{U}_{l-1}^{*}; \mathbf{W}_{l}\mathbf{S}_{l}]^{H} = \bar{\mathbf{P}}_{l}^{S}\mathbf{S} + \bar{\mathbf{P}}_{l}^{Q}\bar{\mathbf{Q}}_{l}.$$
(52)

Remark 10. Equation (52) highlights that due to the successive decompression and recompression performed at each RRU, the quantization noises \mathbf{Q}_l propagate throughout the cascade. The linear combination of the locally beamformed signal $\tilde{\mathbf{S}}_l = \mathbf{W}_l \mathbf{S}_l$ and the decompressed signal \mathbf{U}_{l-1}^* can be seen as a noisy observation of the remote sources \mathbf{S} , through an additive channel

with channel coefficients $\bar{\mathbf{P}}_{l}^{S}$ and correlated noise $\bar{\mathbf{P}}_{l}^{Q}\bar{\mathbf{Q}}_{l}$. This noisy signal is used as an estimate of the partial beamformed signal (47) to be reconstructed at the next RRU.

Next proposition characterizes the optimal test channel \mathbf{U}_l^* in (50) with \mathbf{P}_l^* and \mathbf{K}_l^* for given test channels $\mathbf{U}_{\mathcal{L}_{l-1}}^*$ with their corresponding $\mathbf{P}_1^*, \dots, \mathbf{P}_{l-1}^*$ and $\mathbf{K}_1^*, \dots, \mathbf{K}_{l-1}^*$.

Proposition 3. Let $\mathbf{F}_l \triangleq [\mathbf{U}_{l-1}^*; \mathbf{S}_l]$, $\mathbf{T}_{\mathbf{z}_l, \mathbf{f}_l} \mathbf{\Sigma}_{\mathbf{f}_l | \mathbf{s}_{l+1}} \mathbf{T}_{\mathbf{\bar{z}}_l, \mathbf{f}_l}^H = \mathbf{V}_l \mathbf{\Lambda}_l^D \mathbf{V}_l^H$, where $\mathbf{\Lambda}^D = diag[\lambda_1^D, \dots, \lambda_M^D]$ and $\mathbf{T}_{\mathbf{\bar{z}}_l, \mathbf{f}_l} = \mathbf{\Sigma}_{\mathbf{\bar{z}}_l, \mathbf{f}_l} \mathbf{\Sigma}_{\mathbf{f}_l}^{-1}$, and $\mathbf{\Sigma}_{\mathbf{\bar{z}}_l, \mathbf{f}_l} = \mathbf{\bar{W}}_l \mathbf{\Sigma}_{\mathbf{s}} \mathbf{\bar{P}}_l^{S, H}$, where $\mathbf{\bar{W}}_l \triangleq [\mathbf{W}_1, \dots, \mathbf{W}_l, \mathbf{0}, \dots, \mathbf{0}]$. The minimum distortion (48) is $D_l(R_l) = \sum_{m=1}^M \min\{\lambda, \lambda_m^D\} + \text{Tr}\{\mathbf{\Sigma}_{\mathbf{\bar{z}}_l | \mathbf{f}_l \mathbf{s}_{l+1}}\}$ and $\lambda > 0$ satisfies

$$R_l = \sum_{m=1}^{M} \log^+ \left(\frac{\lambda_m^D}{\lambda}\right),\tag{53}$$

where $\Sigma_{\bar{\mathbf{z}}_{l}|\mathbf{f}_{l}\mathbf{s}_{l+1}} = \Sigma_{\bar{\mathbf{z}}_{l}} - \Sigma_{\bar{\mathbf{z}}_{l},\mathbf{f}_{l}\mathbf{s}_{l+1}} \Sigma_{\mathbf{f}_{l}\mathbf{s}_{l+1}}^{-1} \Sigma_{\bar{\mathbf{z}}_{l},\mathbf{f}_{l}\mathbf{s}_{l+1}}^{H}$, $\Sigma_{\bar{\mathbf{z}}_{l},\mathbf{f}_{l}\mathbf{s}_{l+1}} = \bar{\mathbf{W}}_{l}\Sigma_{\mathbf{s}}[\bar{\mathbf{P}}_{S},\boldsymbol{\delta}_{l+1}]^{H}$, $\Sigma_{\hat{\mathbf{z}}_{l}\mathbf{s}_{l+1}} = [\Sigma_{\mathbf{f}_{l}},\bar{\mathbf{P}}_{l}^{S}\Sigma_{\mathbf{s}}\boldsymbol{\delta}_{l+1}^{H};\boldsymbol{\delta}_{l+1}^{H}\Sigma_{\mathbf{s}}\bar{\mathbf{P}}_{l}^{S,T},\boldsymbol{\Sigma}_{\mathbf{s}_{l+1}}]$, In addition, the minimum distortion in $D_{l}(R_{l})$ is achieved with $\mathbf{K}_{l}^{*} = \mathbf{V}_{l}\boldsymbol{\Lambda}^{Q}\mathbf{V}_{l}^{H}$ and $\mathbf{P}_{l}^{*} = \mathbf{T}_{\bar{\mathbf{z}}_{l},\hat{\mathbf{z}}_{l}}$, where $\boldsymbol{\Lambda}^{Q} = diag[\lambda_{1}^{Q},\ldots,\lambda_{M}^{Q}]$ is a diagonal matrix, with the m-th diagonal element $\lambda_{m}^{Q} = \min\{\lambda,\lambda_{m}^{D}\}/(\lambda_{m}^{D} - \min\{\lambda,\lambda_{m}^{D}\})$.

Outline Proof: The proof is similar to that of the remote Wyner-Ziv source coding problem for source reconstruction in [27]. We consider lossy function reconstruction. For simplicity, we drop the RRU index l in this proof and define $\mathbf{F} \triangleq [\mathbf{U}_{l-1}^*, \mathbf{W}_l \mathbf{S}_l], \mathbf{Y} \triangleq \mathbf{S}_{l+1}, \mathbf{\bar{Z}} \triangleq \mathbf{Z}_l$ and $\mathbf{U} = \mathbf{U}_l$.

First, we obtain a lower bound on the achievable distortion. Let us define the MMSE filters

$$\mathbf{T}_{\mathbf{f},\mathbf{y}} = \boldsymbol{\Sigma}_{\mathbf{f},\mathbf{y}} \boldsymbol{\Sigma}_{\mathbf{y}}^{-1}, \quad \text{and} \quad \left[\mathbf{T}_{\mathbf{\bar{z}},\mathbf{f}} \; \mathbf{T}_{\mathbf{\bar{z}},\mathbf{y}}\right] = \left[\boldsymbol{\Sigma}_{\mathbf{\bar{z}},\mathbf{f}} \boldsymbol{\Sigma}_{\mathbf{\bar{z}},\mathbf{y}}\right] \begin{bmatrix} \boldsymbol{\Sigma}_{\mathbf{f}} & \boldsymbol{\Sigma}_{\mathbf{f},\mathbf{y}} \\ \boldsymbol{\Sigma}_{\mathbf{f},\mathbf{y}}^{H} & \boldsymbol{\Sigma}_{\mathbf{y}} \end{bmatrix}^{-1}. \tag{54}$$

We have from the MMSE estimation of Gaussian vector sources [23],

$$\mathbf{F} = \mathbf{T}_{\mathbf{f},\mathbf{y}}\mathbf{Y} + \mathbf{N}_1,\tag{55}$$

$$\bar{\mathbf{Z}} = \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{f}} \mathbf{F} + \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{y}} \mathbf{Y} + \mathbf{N}_2 = (\mathbf{T}_{\bar{\mathbf{z}}, \mathbf{f}} \mathbf{T}_{\mathbf{f}, \mathbf{y}} + \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{y}}) \mathbf{Y} + \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{f}} \mathbf{N}_1 + \mathbf{N}_2, \tag{56}$$

where N_1 and N_2 correspond to the MMSE errors and are zero-mean jointly Gaussian random vectors independent of each other, N_1 is independent of Y and N_2 is independent of F, Y and have the covariance matrices given by $\Sigma_{N_1} \triangleq \Sigma_{\mathbf{f}|\mathbf{y}}$ and $\Sigma_{N_2} \triangleq \Sigma_{\bar{\mathbf{z}}|\mathbf{f},\mathbf{y}}$.

On the other distortion side, we have

$$D \triangleq \mathrm{E}[(\bar{\mathbf{Z}} - \hat{\mathbf{Z}})(\bar{\mathbf{Z}} - \hat{\mathbf{Z}})^H] \tag{57}$$

$$\geq E[(\bar{\mathbf{Z}} - E[\bar{\mathbf{Z}}|\mathbf{U}, \mathbf{Y}])(\bar{\mathbf{Z}} - E[\bar{\mathbf{Z}}|\mathbf{U}, \mathbf{Y}])^{H}]$$
(58)

$$= \operatorname{Tr}\{E[(\mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}}\mathbf{N}_1 + \mathbf{N}_2 - \hat{\mathbf{N}}_1)(\mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}}\mathbf{N}_1 + \mathbf{N}_2 - \hat{\mathbf{N}}_1)^H]\}$$
(59)

$$= \operatorname{Tr}\{\mathbb{E}[(\mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}}\mathbf{N}_{1} - \hat{\mathbf{N}}_{1})(\mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}}\mathbf{N}_{1} - \hat{\mathbf{N}}_{1})^{H}]\} + \operatorname{Tr}\{\boldsymbol{\Sigma}_{\bar{\mathbf{z}}|\mathbf{f},\mathbf{y}}\}$$
(60)

where (59) follows from (56) and where we have defined $\hat{\mathbf{N}}_1 \triangleq \mathrm{E}[\bar{\mathbf{Z}}|\mathbf{U}] - (\mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}}\mathbf{T}_{\mathbf{f},\mathbf{y}} + \mathbf{T}_{\bar{\mathbf{z}},\mathbf{y}})\mathbf{Y};$ (60) follows from the independence of \mathbf{N}_2 from \mathbf{N}_1 , \mathbf{Y} and \mathbf{F} .

Next, let us define $\mathbf{N}' = \mathbf{V}^H \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{f}} \mathbf{N}_1$ and $\hat{\mathbf{N}}' \triangleq \mathbf{V}^H \hat{\mathbf{N}}_1$, where \mathbf{V} follows from the eigenvalue decomposition

$$\mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}} \mathbf{\Sigma}_{\mathbf{N}_1} \mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}}^H = \mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}} \mathbf{\Sigma}_{\mathbf{f}|\mathbf{y}} \mathbf{T}_{\bar{\mathbf{z}},\mathbf{f}}^H = \mathbf{V} \mathbf{\Lambda}^D \mathbf{V}^H.$$
(61)

Note that N' has independent components of variance Λ^{D} . Therefore, from (60) we have

$$D = \operatorname{Tr}\{E[(\mathbf{N}' - \hat{\mathbf{N}}')(\mathbf{N}' - \hat{\mathbf{N}}')^{H}]\} + \operatorname{Tr}\{\Sigma_{\bar{\mathbf{z}}|\mathbf{f},\mathbf{y}}\}$$
(62)

$$= \sum_{m=1}^{M} \mathrm{E}[(N'_m - \hat{N}'_m)^2] + \mathrm{Tr}\{\boldsymbol{\Sigma}_{\bar{\mathbf{z}}|\mathbf{f},\mathbf{y}}\}, \tag{63}$$

where (62) follows due to the orthonormality of V.

On the other hand, we have

$$R_l \ge I(\mathbf{U}^*, \mathbf{S}; \mathbf{U}|\mathbf{Y})$$
 (64)

$$\geq I(\mathbf{V}^H \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{f}} \mathbf{F}; \mathbf{U} | \mathbf{Y}) \tag{65}$$

$$= h(\mathbf{V}^H \mathbf{T}_{\bar{\mathbf{z}}.\mathbf{f}} \mathbf{F} | \mathbf{Y}) - h(\mathbf{V}^H \mathbf{T}_{\bar{\mathbf{z}}.\mathbf{f}} \mathbf{F} | \mathbf{Y}, \mathbf{U})$$
(66)

$$= h(\mathbf{V}^H \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{f}} \mathbf{N}_1) - h(\mathbf{V}^H \mathbf{T}_{\bar{\mathbf{z}}, \mathbf{f}} \mathbf{N}_1 | \mathbf{Y}, \mathbf{U})$$
(67)

$$= \sum_{m=1}^{K} h(N'_m) - h(N'_m | \mathbf{Y}, \mathbf{U}, \mathbf{N}_{1,1}^{',m-1})$$
(68)

$$\geq \sum_{m=1}^{M} h(N'_m) - h(N'_m | \mathbf{Y}, \mathbf{U})$$

$$(69)$$

$$=\sum_{m=1}^{M}I(N_m';\mathbf{Y},\mathbf{U})\tag{70}$$

$$\geq \sum_{m=1}^{M} I(N'_m; \hat{N'_m}) \tag{71}$$

where (65) follows due to the data processing inequality; (67) is due to (55) and the orthogonality principle of the MMSE estimator; (68) is due to the definition of $\mathbf{N}' = (N'_1, \dots, N'_M)$; (69) follows since conditioning reduces entropy; and (71) is due to the data processing inequality and since \hat{N}_m , where $\hat{\mathbf{N}} = (\hat{N}'_1, \dots, \hat{N}'_M)$, is a function of \mathbf{Y}, \mathbf{U} .

It follows from (63) and (71) that D_l is lower bounded by the sum-distortion D of compressing $N_m' \sim \mathcal{CN}(0, \lambda_m^D)$, $m = 1, \ldots, M$, given as a reverse water filling problem with a modified distortion $\tilde{D} \triangleq D - \text{Tr}\{\Sigma_{\bar{\mathbf{z}}|\mathbf{f},\mathbf{y}}\}$, so that N_m is reconstructed with distortion $d_m \triangleq \mathrm{E}[(N_m' - \hat{N}_m')^2]$,

$$\tilde{D} = \min_{d_1, \dots, d_M > 0} \sum_{m=1}^{M} d_m \quad \text{s.t. } R(D) = \sum_{m=1}^{M} \log \left(\frac{\lambda_m^D}{d_m} \right).$$
 (72)

Note that if $D < \text{Tr}\{\Sigma_{\bar{\mathbf{z}}|\mathbf{f},\mathbf{y}}\}$, then $R(D) = \infty$ and if $D > \text{Tr}\{\Sigma_{\bar{\mathbf{z}}|\mathbf{y}}\}$, then R(D) = 0. The minimum is found with $d_m = \min\{\lambda, \lambda_m^D\}$, for $\lambda > 0$ satisfying (53) [26].

The achievability of the derived lower bound follows by considering the set of tuples in $\mathcal{R}_{IP}(D)$ in Theorem 2 for \mathbf{U}_l satisfying the additional Markov chain $\mathbf{U}_l \oplus \mathbf{R}_l \oplus (\mathbf{U}_{l-1}^*, \mathbf{S}_l) \oplus \mathbf{S}_{l+1}$, which is included in $\mathcal{R}_{IP}(D)$, as $\mathbf{U}_l^* = \mathbf{R}_l + \mathbf{Q}_l$, with $\mathbf{R}_l = \mathbf{P}_l^*[\mathbf{U}_{l-1}^*, \mathbf{W}_l\mathbf{S}_l]^H$ and $\mathbf{Q}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{K}_l^*)$ and where $\mathbf{K}_l^* = \mathbf{V}_l \Lambda_l^Q \mathbf{V}_l^H$ and $\mathbf{P}_l^* = \mathbf{T}_{\mathbf{\bar{z}},\mathbf{f}}$.

The distortion-rate function of the proposed IP scheme in Gaussian C-MIMO is given next.

Theorem 5. Given $U_{\mathcal{L}}^*$ with K_1^*, \dots, K_L^* and P_1^*, \dots, P_L^* successively obtained as in Proposition 3, the distortion-rate of the proposed IP scheme function is given as

$$D_{\text{IP}}(R_1, \dots, R_L) = \text{Tr}\{\boldsymbol{\Sigma}_{\mathbf{z}} - \boldsymbol{\Sigma}_{\mathbf{z}, \mathbf{u}_L^*} \boldsymbol{\Sigma}_{\mathbf{u}_L^*}^{-1} \boldsymbol{\Sigma}_{\mathbf{z}, \mathbf{u}_L^*}^H\},$$
(73)

where $\Sigma_{\mathbf{z},\mathbf{u}_L^*} = \mathbf{W} \Sigma_{\mathbf{s}} \bar{\mathbf{P}}_L^{S,H}$; $\Sigma_{\mathbf{f}_l} = \bar{\mathbf{P}}_l^S \Sigma_{\mathbf{s}} \bar{\mathbf{P}}_l^{S,H} + \bar{\mathbf{P}}_L^Q diag[\mathbf{K}_{\mathcal{L}}^*] \bar{\mathbf{P}}_L^Q$; $\Sigma_{\mathbf{u}_L^*} = \Sigma_{\mathbf{f}_L} + \mathbf{K}_L^*$, and $\Sigma_{\mathbf{f}_l,\mathbf{s}_{l+1}} = \bar{\mathbf{P}}_l^S \Sigma_{\mathbf{s}} \delta_{l+1}$.

Proof: Achievability follows from Theorem 2 with $U_{\mathcal{L}}^*$ obtained as in Proposition 3.

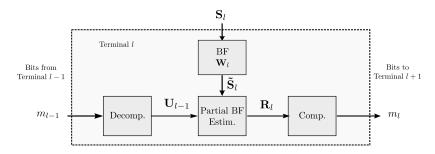


Fig. 6: In-network Processing scheme for C-MIMO.

The IP scheme in Section III-B requires joint compression at each RRU. However, for the Gaussian C-MIMO, it is shown next that the distortion-rate function $D_{\text{IP}}(R_1, \ldots, R_L)$ in Theorem 5 can be achieved by applying at each RRU separate decompression, partial function estimation followed by compression, as shown in Figure 6. At RRU l:

- Upon receiving m_{l-1} , decompress U_{l-1} .
- Apply local beamforming as $\tilde{\mathbf{S}}_l = \mathbf{W}_l \mathbf{S}_l$.
- Linearly combine \mathbf{U}_{l-1} , $\tilde{\mathbf{S}}_l$ to compute an estimate $\mathbf{R}_l = \mathbf{P}_l^*[\mathbf{U}_{l-1}, \tilde{\mathbf{S}}_l]^H$ of the partial function up to Terminal l:

$$\mathbf{Z}_l = \mathbf{W}_1 \mathbf{S}_1 + \dots + \mathbf{W}_l \mathbf{S}_l, \tag{74}$$

• Forward a compressed version of \mathbf{R}_l to Terminal (l+1) using Wyner-Ziv compression considering \mathbf{S}_{l+1} as side information and the test channel $\mathbf{U}_l = \mathbf{R}_l + \mathbf{Q}_l$, $\mathbf{Q}_l \sim \mathcal{CN}(\mathbf{0}, \mathbf{K}_l^*)$. Terminal (L+1) reconstructs \mathbf{Z} using an MMSE estimator as $\hat{\mathbf{Z}} = \mathrm{E}[\mathbf{Z}|\mathbf{U}_L]$.

Proposition 4. For the C-MIMO model, separate decompression, partial function estimation and Wyner-Ziv compression achieves the distortion-rate function $D_{\text{IP}}(R_1, \dots, R_L)$ in Theorem 5.

Proof: The proof follows by showing that at any RRU l, the minimum distortion $D_l(R_l)$ and the test channel \mathbf{U}_l^* in Proposition 3 can also be obtained with separate decompression, partial function estimation and compression. RRU l decompresses \mathbf{U}_{l-1}^* and computes $\tilde{\mathbf{S}}_l = \mathbf{W}_l \mathbf{S}_l$ and $\mathbf{R}_l = \mathbf{P}_l^* [\mathbf{U}_{l-1}, \tilde{\mathbf{S}}_l]^H$. From standard arguments, it follows that compressing à-la Wyner-Ziv with

 S_{l+1} as decoder side information requires

$$R_l \ge I(\mathbf{R}_l; \mathbf{U}_l | \mathbf{S}_{l+1}) \tag{75}$$

$$= I(\mathbf{V}_l^H \mathbf{R}_l; \mathbf{U}_l | \mathbf{S}_{l+1}) \tag{76}$$

$$= I(\mathbf{V}^H \mathbf{T}_{\overline{\mathbf{z}}_1, \mathbf{f}_l} \mathbf{F}_l; \mathbf{U}_l | \mathbf{S}_{l+1}), \tag{77}$$

where (76) follows since V_l is orthonormal.

Following from (66), and by noting that the distortion achievable by estimating $\bar{\mathbf{Z}}_l$ from $\bar{\mathbf{U}}_l^*$ and \mathbf{S}_{l+1} corresponds to D_l in Proposition 3, it follows that any achievable distortion D_l is also achievable with separate decompression, partial function estimation and compression.

V. A LOWER BOUND

In this section, we obtain an outer bound on the RD region $\mathcal{R}(D)$ using a Wyner-Ziv type system in which the decoder is required to estimate the value of some function Z of the input at the encoder X and the side information Y [22]. We use the following notation from [28]. Define the minimum average distortion for Z given Q as $\mathcal{E}(Z|Q) \triangleq \min_{f:Q\to Z} \mathrm{E}[d(Z,f(Q))]$, and the Wyner-Ziv type RD function for value Z, encoder input X and side information Y available at the decoder, as [22]

$$R_{Z,X|Y}^{\text{FWZ}}(D) \triangleq \min_{p(u|x):\mathcal{E}(Z|U,Y) \le D} I(X;U|Y). \tag{78}$$

An outer bound can be obtained using the rate-distortion Wyner-Ziv type function in (78).

Theorem 6. The RD region $\mathcal{R}(D)$ is contained in the region $\mathcal{R}^{o}(D)$, given by the union of tuples (R_1, \ldots, R_L) satisfying

$$R_l \ge R_{Z,S_{\mathcal{L}_l}|S_{\mathcal{L}_l^c}}^{\mathrm{FWZ}}(D), \quad l = 1,\dots, L.$$
 (79)

Outline Proof: The outer bound is obtained by the RD region of L network cuts, such that for the l-th cut, S_{l+1}^n, \ldots, S_L^n acts as side information at the decoder. See Appendix I.

In the Gaussian C-MIMO model, Theorem 6 can be used to explicitly write a lower bound on the achievable distortion for a given fronthaul tuple (R_1, \ldots, R_L) as given next.

Proposition 5. Given the fronthaul tuple $(R_1, ..., R_L)$, the achievable distortion in a C-MIMO system is lower bounded by $D_{LB}(R_1, ..., R_L) = \max_{l=1,...,L} D_l$, where

$$D_{l} = \min_{d_{l,1},\dots,d_{l,M}>0} \sum_{m=1}^{M} d_{l,m} \text{ s.t. } R_{l} = \sum_{m=1}^{M} \log^{+} \left(\frac{\lambda_{l,m}^{D}}{d_{l,m}}\right), \tag{80}$$

with $d_{l,m} = \min\{\lambda_l, \lambda_{l,m}^D\}$, for $\lambda_l > 0$ and $\lambda_{l,m}^D, m = 1, ..., M$ are the eigenvalues of $\Sigma_{\mathbf{z}|\mathbf{s}_{\mathcal{L}_l^c}} = \mathbf{W} \Sigma_{\mathbf{s}|\mathbf{s}_{\mathcal{L}_l^c}} \mathbf{W}^H = \mathbf{V}_l \mathbf{\Lambda}_l^D \mathbf{V}_l^H$, with $\mathbf{\Lambda}_l^D = diag[\lambda_{l,1}^D, ..., \lambda_{l,M}^D]$.

Outline Proof: The poof follows by computing explicitly the Wyner-Ziv RD type function in (78) for Gaussian vector sources for each network cut, which follows in the same lines as the proof of Proposition 3 with $\mathbf{F} = \mathbf{S}_{\mathcal{L}_l}$, $\mathbf{Y} = \mathbf{S}_{\mathcal{L}_l^c}$ and $\mathbf{Z} = \mathbf{WS}$. Note that in this case $\Sigma_{\mathbf{z}|\mathbf{f},\mathbf{y}} = \mathbf{0}$.

VI. NUMERICAL RESULTS

In this section, we provide numerical examples to illustrate the average sum-distortion obtained using IR and IP as detailed in Section IV. We consider several C-MIMO examples, with K users and L RRUs, each equipped with M antennas under different fronthaul capacities. The CP wants to reconstruct the receive-beamforming signal using the Zero-Forcing weights given by

$$\mathbf{W} = (\mathbf{H}_{\mathcal{L}}^H \mathbf{H}_{\mathcal{L}})^{-1} \mathbf{H}_{\mathcal{L}}.$$
 (81)

The channel coefficients are distributed as $h_{l,k} \sim \mathcal{CN}(0,1)$. We also consider the SR scheme of [5]. The schemes are compared among them, and to the lower bound in Theorem 6. Note that WZR achieves the same distortion-rate function as IR as shown in Theorem 4, and is omitted. Figure 7 depicts the sum-distortion in a C-MIMO network with K=15 users and L=4 RRUs, each equipped with M=7 antennas for equal fronthaul capacity per link $R_1=\ldots=R_L=KB$, as a function of the average number of bits per user B. As it can be seen from the figure, the scheme IP based on distributed beamforming outperforms the other centralized beamforming schemes, and performs close to the lower bound. For centralized beamforming, the scheme IF

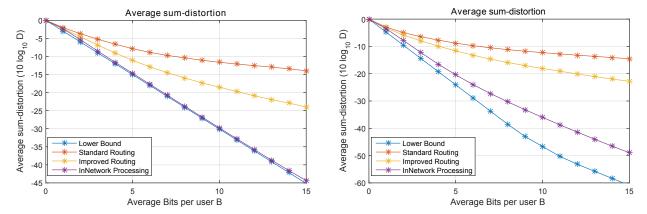


Fig. 7: Average sum-distortion D for M=15, Fig. 8: Average sum-distortion D for M=15, L=4, K=7 vs. average bits per user for B=L=4, K=7 vs. average bits per user for $0,\ldots,15$ for balanced FH capacities $R_l=KB$. $B=0,\ldots,15$ for increasing FH $R_l=lKB$.

performs significantly better than SR, as it reduces the required fronthaul by only compressing the innovation at each RRU.

Figure 8 shows the sum-distortion in a C-MIMO network with K=15 users and L=4 RRUs, each equipped with M=7 antennas, with increasing fronthaul capacity per link $R_l=lKB,\ l=1,\ldots,L$ as a function of the average number of bits per user B. In this case, the IP scheme using distributed beamforming also achieves the lowest sum-distortion among the proposed schemes.

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APPENDIX I

PROOF OF THEOREM 6

Suppose there exist $f_l^{(n)}$, $l=1,\ldots,L$ and $g^{(n)}$ such that for (R_1,\ldots,R_L) , $\frac{1}{n}\sum_{i=1}^n \mathrm{E}[d(Z_i;\hat{Z}_i)] \leq D+\epsilon$, where $\epsilon\to 0$ as $n\to\infty$. Define $U_{l,i}\triangleq (m_l,S_{\mathcal{L}_l^c}^{-i})$, for $l=1,\ldots,L$, where $S_{\mathcal{L}_l^c}^{-i}\triangleq (S_{\mathcal{L}_l^c,1}^{i-1},S_{\mathcal{L}_l^c,i+1}^n)$ and note the Markov chain relation

$$S_{\mathcal{L}_{l,i}^{c}} - S_{\mathcal{L}_{l,i}} - U_{l,i}, \quad l = 1, \dots, L.$$
 (82)

For the l-th cut we have,

$$nR_l \ge H(m_l) \tag{83}$$

$$\geq I(m_l; S_{\mathcal{L}}^n | S_{\mathcal{L}_s^n}^n) \tag{84}$$

$$= \sum_{i=1}^{n} H(S_{\mathcal{L},i}|S_{\mathcal{L}_{l}^{c},i}) - H(S_{\mathcal{L}_{l},i}|S_{\mathcal{L}_{l}^{c}}^{n}, m_{l}, S_{\mathcal{L},1}^{i-1})$$
(85)

$$\geq \sum_{i=1}^{n} H(S_{\mathcal{L},i}|S_{\mathcal{L}_{l}^{c},i}) - H(S_{\mathcal{L},i}|S_{\mathcal{L}_{l}^{c},i}, m_{l}, S_{\mathcal{L}_{l}^{c}}^{-i})$$
(86)

$$= \sum_{i=1}^{n} I(S_{\mathcal{L},i}; U_{l,i} | S_{\mathcal{L}_{l}^{c},i})$$
(87)

$$\geq \sum_{i=1}^{n} R_{Z,S_{\mathcal{L}_l}|S_{\mathcal{L}_l^c}}^{\text{FWZ}} (\mathcal{E}(Z_i|U_{l,i},S_{\mathcal{L}_l^c,i})), \tag{88}$$

where (85) follows since $S_{\mathcal{L}}^n$ is i.i.d., (86) follows since conditioning reduces entropy. On the other hand, we have

$$\mathcal{E}(Z_i|U_{l,i}, S_{\mathcal{L}_i^c, i}) = \mathcal{E}(Z_i|m_l, S_{\mathcal{L}_i^c}^n)$$
(89)

$$= \mathcal{E}(Z_i|m_l, m_{l+1}, \dots, m_L, S_{\mathcal{L}_i^c}^n)$$
(90)

$$\leq \mathcal{E}(Z_i|m_L) \tag{91}$$

where (90) follows since m_l is a deterministic function of m_{l-1}, S_l^n , i.e., $m_l = f_l(m_{l-1}, S_l^n)$, (91) follows since reducing the information can only increase the distortion, Then,

$$nR_l \ge \sum_{i=1}^n R_{Z,S_{\mathcal{L}_l}|S_{\mathcal{L}_l^c}}^{\text{FWZ}}(\mathcal{E}(Z_i|m_L))$$
(92)

$$\geq \sum_{i=1}^{n} R_{Z,S_{\mathcal{L}_{l}}|S_{\mathcal{L}_{l}^{c}}}^{\text{FWZ}}(\text{E}[d(Z_{i},\hat{Z}_{i})]$$
(93)

$$\geq nR_{Z,S_{\mathcal{L}_l}|S_{\mathcal{L}_l^c}}^{\text{FWZ}}(D+\epsilon),\tag{94}$$

where (92) follows since $R_{Z,S_{\mathcal{L}_l}|S_{\mathcal{L}_l^c}}^{\mathrm{FWZ}}(D)$ is monotonic in D, (93) is due to \hat{Z}_i being a function of m_L , and (94) follows as $R_{Z,S_{\mathcal{L}_l}|S_{\mathcal{L}_l^c}}^{\mathrm{FWZ}}(D)$ is convex and monotone in D. This completes the proof.