Diversity Combining via Universal Dimension-Reducing Space—Time Transformations

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Abstract—Receiver diversity combining methods play a key role in combating the detrimental effects of fading in wireless communication and other applications. A novel diversity combining method is proposed, where a universal, i.e., channel independent, orthogonal dimension-reducing space-time transformation is applied prior to quantization of the signals. The scheme may be considered as the counterpart of Alamouti modulation, and more generally of orthogonal space-time block codes.

Index Terms—Diversity methods, analog-digital conversion, relays, MIMO systems.

I. INTRODUCTION

N WIRELESS communication, diversity methods play a central role in combating the detrimental effects of severe channel variation (fading). Of the many techniques that have been developed over the years with this goal, an important class involves the use of multiple receive antennas. With sufficient separation between the antennas, each antenna may be viewed as a branch receiving the transmitted signal multiplied by an approximately independent fading coefficient. Diversity is achieved as the probability that the signal is severely affected by fading on all branches simultaneously is greatly reduced. The number of such (roughly) independent branches is commonly referred to as the diversity order.

Several methods of receive diversity combining are well known, most notably maximum-ratio combining (MRC), selection combining (SC), and equal-gain combining. All of these amount to performing a linear *dimension-reducing* operation. For MRC, this operation is optimal in the sense of producing sufficient statistics, whereas in other methods some information loss is incurred in order to reduce some of the implementation prices inherent to MRC.

The dimension reduction aspect of the combining operation may serve several important goals. One important design goal is to reduce power consumption of a communication device. Most modern communication systems operate in the digital domain. Therefore, such a system must have at least one

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analog-to-digital conversion (ADC) unit that usually consumes a significant amount of power. Some diversity combining methods, most notably SC, can serve to reduce the number of ADCs and thus result with power savings.

Another goal, that is also very relevant to modern communication systems, is reducing the bit rate of the digital interface between different digital blocks. For example, in a centralized (cloud) radio access network setting, each terminal (or relay) needs to be connected via a fronthaul link to the cloud. The combining operation can be used to reduce the required bit rate when communicating over rate-constrained links.

In yet a different scenario, which may be viewed as a special instance of compressive sensing, the dimension-reduction operation can serve to allow time-domain sub-Nyquist sampling.¹ Specifically, when the desired signal is known to posses some additional structure, beyond the frequency band it occupies, the sampling rate can be reduced significantly with a limited loss of information (due to noise accumulation, assuming the signal is contaminated by some noise); see, e.g., [1] and references therein. We will observe that certain scenarios of sub-Nyquist sampling can be recast as an equivalent multiple-antenna problem, and hence dimension-reducing diversity methods can play a role in sub-Nyquist sampling.

As noted above, MRC is optimal in the sense of producing sufficient statistics. As recalled in more detail in the sequel, in MRC the outputs of different antennas are multiplied by channel-dependent weights. Since the combining depends on the specific channel realization, usually, this multiplication is carried out in the digital domain, in which case it requires sampling of all receive antennas, and employing an ADC for each.

Selection combining is an effective method to reap most of the benefits of MRC while reducing the number of ADCs. Namely, in SC, only the strongest receive antenna (or in hybrid schemes, a subset of antennas) is sampled. This allows using less ADC modules, thus consuming less power, while paying only a small price in terms of performance. A classical survey of receive diversity techniques is [2]. More recent accounts that also consider multiple-input multiple-output (MIMO) channels are [3] and [4].

While SC is a practically appealing diversity-combining method in many applications, in common with MRC, its

¹We note that the connection between time-domain sampling and multiple-antenna receive combining is well known, e.g., the analogy between MRC and sampled matched filtering is clear.

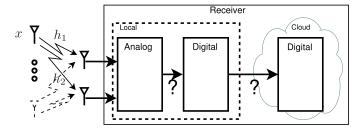


Fig. 1. Basic scenario: receiver architecture for a 2×1 SIMO channel.

implementation requires knowledge (albeit, limited) of the channel in the selection phase. This requires implementing estimation and decision mechanisms that in certain scenarios may be prone to errors and add latency to the system. More importantly, as will be shown in the sequel, SC is ill-suited to scenarios where multiple desired signals are received, as will be most prominently demonstrated for the case of a multiple-access channel.

Another classical combining method is that of equal-gain combining where only the phases of the different antennas are aligned (via phase shifters). While the implementation of the latter is more complex than that of SC, it closes some of the gap to MRC [2]. For a recent survey and comparison of these methods, in the context of mmWave receivers, we refer the reader to [5] and [6]. Since this method is also ill-suited to scenarios where multiple desired signals are received at a single antenna, we take SC as a benchmark in our analysis.

We introduce a new linear diversity-combining scheme utilizing orthogonal space-time block codes. The key difference between the proposed scheme and traditional linear combining schemes is that it is *universal*. That is, the combining weights (in the proposed scheme, the space-time transformation) do not depend on the channel realization. As will be shown, in scenarios involving multi-user detection, universal combining has significant benefits over known linear combining schemes.

The rest of this paper is organized as follows. Section II describes the proposed method in the context of a wireless communication scenario with one transmit and two receive antennas. Section III provides a performance comparison with known methods for both single-user and multi-user scenarios. Section IV describes an application of the proposed method for relaying in a cloud radio access scenario. Section V provides an extension of the method to more than two receive antennas. Section VI outlines the relation between the multiple-antenna scenario considered and sub-Nyquist sampling and demonstrates how the proposed method can be applied to the latter problem.

II. DESCRIPTION OF THE SCHEME FOR TWO RECEIVE ANTENNAS

Consider a 2×1 single-input multiple-output (SIMO) channel, with channel coefficients h_1 and h_2 , as depicted in Figure 1. The signal received at antenna i=1,2, at discrete time t, is

$$s_i(t) = h_i x(t) + n_i(t). \tag{1}$$

We assume that the noise $n_i(t)$ is i.i.d. over space and time with samples that are circularly-symmetric complex Gaussian random variables with unit variance. We further assume the transmitted symbols are subject to the power constraint $\mathbb{E}(|x|^2) = P$.

The scheme works on batches of two time instances and for our purposes, it will suffice to describe it for time instances t=1,2. Let us stack the four complex samples received over T=2 time instances, two over each antenna, into an 8×1 real vector:

$$\mathbf{s} = [s_{1R}(1)s_{1I}(1)s_{2R}(1)s_{2I}(1)s_{1R}(2)s_{1I}(2)s_{2R}(2)s_{2I}(2)]^{T},$$
(2)

where x_R and x_I denote the real and imaginary parts of a complex number x. We similarly define the stacked noise vector \mathbf{n} . Likewise, we define

$$\mathbf{x} = [x_R(1) \, x_I(1) \, x_R(2) \, x_I(2)]^T. \tag{3}$$

Next, we form a 4-dimensional real vector y by applying to the vector s the transformation y = Gs where

$$\mathbf{G} = \frac{1}{\sqrt{2}} \begin{bmatrix} 1 & 0 & 0 & 0 & 0 & 0 & 1 & 0 \\ 0 & 1 & 0 & 0 & 0 & 0 & 0 & -1 \\ 0 & 0 & 1 & 0 & -1 & 0 & 0 & 0 \\ 0 & 0 & 0 & 1 & 0 & 1 & 0 & 0 \end{bmatrix}. \tag{4}$$

Note that unlike conventional diversity-combining schemes, here the combining matrix **G** is *universal*, i.e., it does not depend on the channel coefficients.

Remark 1: We note that the transpose of G is precisely the description of the linear operation performed by Alamouti modulation [7] when expressed over the reals.

It is readily shown that the following holds

$$\mathbf{y} = \frac{\|\mathbf{h}\|}{\sqrt{2}} \mathbf{U}(h_1, h_2) \mathbf{x} + \mathbf{G} \mathbf{n}$$
$$= \frac{\|\mathbf{h}\|}{\sqrt{2}} \mathbf{U}(h_1, h_2) \mathbf{x} + \mathbf{n}', \tag{5}$$

where

$$\mathbf{U}(h_1, h_2) = \frac{1}{\|\mathbf{h}\|} \begin{bmatrix} h_{1R} & -h_{1I} & h_{2R} & -h_{2I} \\ h_{1I} & h_{1R} & -h_{2I} & -h_{2R} \\ h_{2R} & -h_{2I} & -h_{1R} & h_{1I} \\ h_{2I} & h_{2R} & h_{1I} & h_{R1} \end{bmatrix}.$$
(6)

A key observation is that $U(h_1, h_2)$ is an orthonormal matrix for any h_1, h_2 :

$$\mathbf{U}^{T}(h_{1}, h_{2})\mathbf{U}(h_{1}, h_{2}) = \mathbf{I}, \tag{7}$$

where I is the identity matrix. Further, since the rows of G are orthonormal, it follows that \mathbf{n}' is i.i.d. and Gaussian with variance 1/2.²

We may reconstruct (up to additive noise) the original samples by forming

$$\hat{\mathbf{x}} = \mathbf{U}^{T}(h_{1}, h_{2}) \cdot \mathbf{y}$$

$$= \frac{\|\mathbf{h}\|}{\sqrt{2}} \mathbf{x} + \mathbf{n}''$$
(8)

 $^{^2}$ The variance is 1/2 as we chose above to normalize the complex noise to have unit power.

where \mathbf{n}'' is also i.i.d. Gaussian with variance 1/2.

Since the dimension (over the reals) of y is four rather than eight, as is the dimension of the received signal s, we obtained a universal dimension-reducing combining scheme.

We note that in order to perform the reconstruction, the channel gains must of course be estimated as in true for any scheme, e.g, via the use of pilots. The difference is that in the proposed scheme, the estimation process occurs after the combining phase, i.e., using the effective MIMO channel (5).

III. APPLICATION TO ANALOG-TO-DIGITAL CONVERSION

In this section we demonstrate the applicability of the scheme to analog-to-digital conversion for power-limited receivers of narrowband signals. Similarly to SC, it may be used to achieve maximal diversity order with a single radio-frequency (RF) chain and ADC.

We start by analyzing the performance in a scenario where a terminal with two antennas receives the signal transmitted from a single user equipped with a single antenna. In this scenario, setting aside hardware limitations, optimal SC outperforms the proposed method. We then show, in contrast, that when the number of transmitting users increases, the new method is beneficial compared to all known methods that make use of a single RF chain.

A. Single-User Scenario

Consider again the scenario of a 2×1 SIMO system as depicted in Figure 1 and described in the previous section. We note that as the fading coefficients are constants (rather than impulse responses), the model assumed is that of frequency-flat fading.

The best performance may be attained by quantizing (at sufficient resolution) the output of each antenna and then using MRC. Applying MRC amounts to forming

$$y_{\text{MRC}} = \frac{1}{\|\mathbf{h}\|} \begin{bmatrix} h_1^* & h_2^* \end{bmatrix} \begin{bmatrix} s_1 \\ s_2 \end{bmatrix}$$

$$= \frac{\|h_1\|^2 + \|h_2\|^2}{\|\mathbf{h}\|} x + \frac{h_1^* n_1 + h_2^* n_2}{\|\mathbf{h}\|}$$

$$= \underbrace{\|\mathbf{h}\|}_{\mathbf{h}_{\text{eff,MRC}}} x + n, \tag{9}$$

where n is white and Gaussian with unit variance. This constitutes a sufficient statistic. In particular, it is well known [2] that when h_1 and h_2 are independent, we obtain a diversity order of 2. The major downside of such a system is that two RF chains and ADCs are needed.

A classic alternative to MRC that requires only one RF chain and ADC is the method of selection combining. Here, rather than choosing the antenna arbitrarily, we choose the one with the higher signal-to-noise ratio (SNR). Thus the effective channel becomes

$$y_{SC} = \underbrace{\max(|h_1|, |h_2|)}_{h_{\text{eff,SC}}} x + n,$$
 (10)

where again n is Gaussian noise with unit variance. While the performance does not reach that of MRC, it does attain a

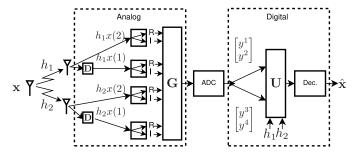


Fig. 2. Proposed receiver front end employing a universal orthogonal space-time diversity transformation. The sampling rate of the ADC element is equal to that of SC.

diversity order of 2. The precise performance under independent Rayleigh fading of SC is well known and may be found, e.g., in [2].

Alternatively, we may apply the space-time diversity combining method described in the previous section to the problem of ADC as follows. Since the processing matrix **G** is fixed for all channels, it is possible to implement its operation in the analog domain (i.e., prior to quantization), requiring only delay, summation and negation elements.

We note that the implementation of the transformation in the analog domain requires four adders (each summing two real signals), two negation elements and two (complex) analog delay lines. As the number of these elements and their complexity (and also their power consumption) does not scale with the resolution of ADCs, it is reasonable to expect that for high-resolution systems, the savings afforded by reducing the number of ADCs will outweigh the additional associated complexity.

As depicted in Figure 2, the received signals are first passed through the dimension-reducing transformation ${\bf G}$ to obtain the vector ${\bf y}=[y^1\,y^2\,y^3\,y^4]^T$ as defined in (5) and (6). Then, a (component-wise) scalar uniform quantizer $Q(\cdot)$ is applied to ${\bf y}$ to obtain ${\bf y}_q=Q({\bf y})$. Note that for T=2 as considered, the dimension of the output vector ${\bf y}$ is four and is equal to the number of source symbol. see (3). Therefore the sampling rate is equal to that of single-antenna (or SC) processing.

We denote the quantization error vector by

$$\mathbf{e} = \mathbf{y} - \mathbf{y}_q$$

$$= \mathbf{y} - Q(\mathbf{y}). \tag{11}$$

The sequence of quantized samples is used to reconstruct an estimation of the source vector $\hat{\mathbf{x}} = [\hat{x}_R(1) \, \hat{x}_I(1) \, \hat{x}_R(2) \, \hat{x}_I(2)]^T$ by applying the transformation:

$$\hat{\mathbf{x}} = \mathbf{U}(h_1, h_2)^T \mathbf{y}_q. \tag{12}$$

Using (5) and (11), we have

$$\hat{\mathbf{x}} = \mathbf{U}(h_1, h_2)^T (\mathbf{y} - \mathbf{e}) \tag{13}$$

$$= \mathbf{U}(h_1, h_2)^T \left(\frac{\|\mathbf{h}\|}{\sqrt{2}} \mathbf{U}(h_1, h_2) \mathbf{x} + \mathbf{n}' - \mathbf{e} \right)$$
 (14)

$$= \frac{\|\mathbf{h}\|}{\sqrt{2}} \mathbf{x} + \mathbf{n}'' - \mathbf{e}', \tag{15}$$

where n'' has the same distribution as n.

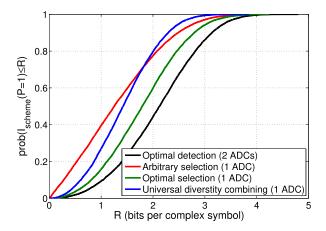


Fig. 3. Cumulative distribution function of the mutual information attained by different combining methods, for the case of single transmitter with a single antenna and a receiver with two receive antennas; i.i.d. Rayleigh fading is assumed with P=1.

As for the quantization error e and its transformed variant e', we may invoke the standard assumption, that may be justified using subtractive dithered quantization, that it is independent of the signal (and hence of x) and is white (i.e., its covariance matrix is the scaled identity).

We conclude that the input/output relationship of the proposed diversity combiner is identical to that of MRC, except for a power loss of a factor of two. In other words, we attain full diversity but no array gain, precisely as in the case of Alamouti space-time diversity transmission. In comparison with SC (without taking into account implementation losses), there is a loss in the achieved SNR whereas an advantage is that no estimation of channel quality in the analog front end nor switching is required.

To compare the performance of the different schemes, we first note that for all channel realizations

$$|h_{\text{eff,Ala}}|^2 \le |h_{\text{eff,SC}}|^2 \le |h_{\text{eff,MRC}}|^2 \tag{16}$$

as these quantities correspond to the average, maximal, and sum of the (squared) channel gains, respectively.

We next compare the mutual information attained by each of the schemes in an i.i.d Rayleigh fading environment. The mutual information is given by

$$I_{\text{scheme}}(P) = \log \left(1 + P h_{\text{eff,scheme}}^2\right),$$
 (17)

where $h_{\rm eff,scheme}$ is the effective scalar channel defined by either (9), (10) or (15). As $h_{\rm eff,scheme}$ is a random variable, so is the resulting mutual information.

Figure 3 depicts the cumulative distribution function (CDF) of the mutual information attained by the three methods, for P=1 (0 dB). MRC combining is denoted as "optimal detection", SC is denoted as "optimal selection" and the new transformation is denoted as "universal diversity combining". As another figure of merit, we also added another (very simple) method of dimension reduction in which the receiver arbitrarily chooses one of the antennas, which we refer to as "arbitrary selection". As can be seen, Alamouti combining outperforms arbitrary antenna selection for "bad" channels

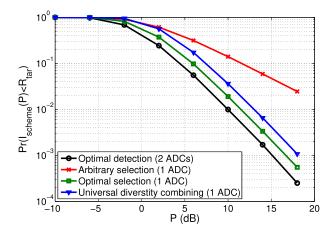


Fig. 4. Comparison of outage probability of different combining methods for an optimally encoded user transmitting over a 2×1 i.i.d. Rayleigh fading channel, with a target rate of $R_{\rm tar}=2$ bits per complex symbol.

but falls short of the performance of SC as expected in light of (16).

Figure 4 depicts the outage probability (assuming perfect coding) for a target rate of 2 bits per complex symbol. As expected, Alamouti combining has a fixed ~ 3 dB gap (factor of two power loss) from MRC, while the gap of SC is smaller.

B. Multi-User Scenario

We consider now the scenario of a $2 \times N$ MIMO-MAC system where N users, each equipped with a single antenna, transmit to a common receiver that is equipped with two antennas. Again, it is assumed that only a single RF chain is to be used at the receiver. As a figure of merit for performance, we now use the symmetric capacity (recalled below). We note that unlike in the case of a single user, sophisticated multi-user detection methods need to be applied in order to approach the symmetric capacity.

We first recall the more general channel model of a MIMO-MAC with N users, where each transmitter has N_t antennas and the receiver has N_r antennas. The input/output relation can be expressed as

$$\mathbf{y} = \sum_{i=1}^{N} \mathbf{H}_i \mathbf{x}_i + \mathbf{n} \tag{18}$$

where \mathbf{H}_i is the channel matrix between user i and the receiver. We assume isotropic ("white") transmission by each user and that all users are subject to the same power constraint P.

Define a subset of users by $S \subseteq \{1, 2, ..., N\}$. Then, the capacity region of the channel is given by (see, e.g., [8]) all rate vectors $(R_1, ..., R_N)$ satisfying

$$\sum_{i \in S} R_i \le C(S)$$

$$\triangleq \log \det \left(\mathbf{I} + P \sum_{i \in S} \mathbf{H}_i \mathbf{H}_i^H \right), \tag{19}$$

for all subsets S in the power set of $\{1, 2, ..., N\}$.

If we impose the constraint that all users transmit at the same rate, then the maximal achievable rate is given by substituting $R_i = C_{\rm sym}/N$ in (19), from which it follows that the symmetric capacity is dictated by the bottleneck:

$$C_{\text{sym}}(P) = \min_{S \subseteq \{1, 2, \dots, N\}} \frac{N}{|S|} \log \det \left(\mathbf{I} + P \sum_{i \in S} \mathbf{H}_i \mathbf{H}_i^H \right). \tag{20}$$

As all the combining methods considered involve only linear operations, we may obtain the associated symmetric capacity for each by computing (20) for the respective effective channel. We next derive explicitly the symmetric capacity associated with each method for the case of two users.

We first consider the unrestricted symmetric capacity, i.e., the symmetric capacity for a system employing optimal reception (two RF chains). We note that we may rewrite (18) as

$$\mathbf{y} = \mathbf{h}_1 \mathbf{x}_1 + \mathbf{h}_2 \mathbf{x}_2 + \mathbf{n} \tag{21}$$

where

$$\mathbf{h}_i = \left[h_{1i} \ h_{2i} \right]^T. \tag{22}$$

Denoting $\mathbf{H}_{comb} = [\mathbf{h}_1 \ \mathbf{h}_2]$, (20) can be written as

$$C_{\text{sym,opt}}(P) = \min \{ C_{\text{opt}}(\{1\}), C_{\text{opt}}(\{2\}), C_{\text{opt}}(\{1,2\}) \}$$

= \text{min} \{ 2 \log(1 + P \|\mathbf{h}_1\|^2), 2 \log(1 + P \|\mathbf{h}_2\|^2), \log \deta(1 + P \cdot \mathbf{H}_{\text{comb}} \mathbf{H}_{\text{comb}}^H) \}. (23)

Similarly, for SC, the symmetric capacity can be expressed as

$$C_{\text{sym,SC}}(P) = \max_{j} \min \left\{ C_{\text{SC}}(\{1\}), C_{\text{SC}}(\{2\}), C_{\text{SC}}(\{1,2\}) \right\}$$

$$= \max_{j} \min \left\{ 2 \log \left(1 + P |h_{1,j}|^2 \right), 2 \log \left(1 + P |h_{2,j}|^2 \right), \log \det \left(1 + P (|h_{1,j}|^2 + |h_{2,j}|^2) \right) \right\}. \tag{24}$$

We now turn to the case of the proposed method. By (5) and (18), the output is given by

$$\mathbf{y} = \sum_{i=1}^{2} \frac{\|\mathbf{h}_i\|}{\sqrt{2}} \mathbf{U}(h_{1i}, h_{2i}) \mathbf{x}_i + \mathbf{n}', \tag{25}$$

where $U(h_{1i}, h_{2i})$ is given by (6). Recalling that these matrices are orthonormal, we obtain

$$C_{\text{sym,Ala}}(P) = \min \{ C_{\text{Ala}}(\{1\}), C_{\text{Ala}}(\{2\}), C_{\text{Ala}}(\{1, 2\}) \}$$

$$= \min \left\{ 2 \log \left(1 + \frac{P}{2} \|\mathbf{h}_{1}\|^{2} \right),$$

$$2 \log \left(1 + \frac{P}{2} \|\mathbf{h}_{2}\|^{2} \right),$$

$$\log \left(1 + \frac{P}{2} \left(\|\mathbf{h}_{1}\|^{2} + \|\mathbf{h}_{2}\|^{2} \right) \right) \right\}.$$
 (26)

A comparison of the CDF of the symmetric capacity achieved by the different methods, in an i.i.d. Rayleigh fading environment, is shown in Figures 5 and 6.

Figure 5 depicts the CDF of all three methods for N=8 users, each with power constraint P=1. As can be seen, Alamouti combining clearly outperforms SC.

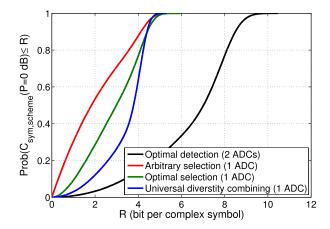


Fig. 5. CDF of the symmetric capacity associated with the different considered methods, in an i.i.d. Rayleigh fading channel environment with eight transmitters, each equipped with a single antenna and where the receiver is equipped with two antennas.

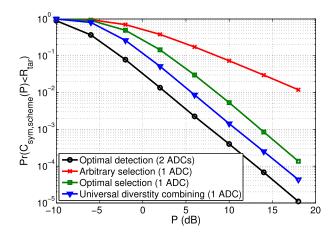


Fig. 6. Comparison of the outage probability associated with the different methods, in an i.i.d. Rayleigh fading environment with eight transmitters, each equipped with a single antenna, and a common receiver equipped with two antennas. All users transmit at an equal rate $R_{\rm tar}$ such that $8R_{\rm tar}=2$ bits per complex symbol.

Figure 6 depicts the outage probability as a function of the SNR, where all users transmit at a common target rate of 2 bits per complex channel use. While the universal combining scheme maintains the ~ 3 dB gap from the MRC, SC suffers from a larger gap. In fact, it is easy to see that the (asymptotic in SNR) gap of SC becomes arbitrarily large as the number of users grows.

We may formalize the asymptotic performance of the universal diversity combining scheme in the form of a theorem.

Theorem 1: For a Rayleigh fading $2 \times N$ MIMO-MAC, when all users transmit at the same rate, for any fixed (symmetric) target rate R, at asymptotic high SNR, the outage probability achieved when applying the universal combining scheme suffers a power penalty factor no greater than 2 with respect to the optimal receiver, i.e.,

$$\lim_{P \to \infty} \frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} = \lim_{P \to \infty} \frac{\Pr\left(C_{\text{sym,opt}}(P/2) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)}.$$
(27)

We give the proof of the theorem for the case of two users in Appendix where it is also explained how the general claim follows along similar lines.

Intuitively, for a large number of users, antenna selection becomes ineffective since the chances that a single receive antenna is good for all users diminishes. Similar reasoning holds also for other channel-dependent reduction methods such as equal gain combining.

IV. APPLICATION TO "DUMB" RELAYING FOR MULTI-USER LINEAR DETECTION AT A REMOTE DESTINATION

Another potential application of the proposed scheme is to be employed as part of a "dumb" relay. By a "dumb" relay we mean a relay (equipped with multiple antennas) that can only apply channel-independent linear processing to the antenna outputs followed by scalar quantization, the output of which is fed into a rate-constrained bit pipe. An example for such an architecture is a cloud radio access network (C-RAN) in which the radio function units and baseband processing units are separated. Since this architecture is designed to support a very large number of users, there has been considerable effort in developing methods to reduce the processing complexity at the cloud (see, e.g., [11]). In this section, we employ the proposed universal transformation to better utilize the rate-constrained bit pipe to the cloud, thus addressing a different bottleneck in this architecture.

Unlike in the previous section, the scheme we present now operates purely in the digital domain. A further difference is that we no longer assume frequency-flat fading. Rather, we will assume that after analog-to-digital conversion, a DFT operation is applied, so that we are working in the frequency domain. In other words, the static channel we will consider is to be understood to apply to a single tone. The "time" index t will correspondingly refer to subsequent uses of the same tone, or in a practical setting could apply to adjacent tones as these typically have very similar channel coefficients.

We demonstrate the application to "dumb" relaying in the context of the system described in Figure 7. Here, two single-antenna users communicate with a central receiver via two relays, each equipped with two antennas, where the medium between the users and relays is a Rayleigh fading wireless channel, whereas the relays are connected to the central receiver via bit pipes.

The signal received at relay i=1,2 and antenna j=1,2 is given by

$$s_j^i(t) = h_{j1}^i \cdot x_1(t) + h_{j2}^i \cdot x_2(t) + n_j^i(t), \tag{28}$$

and the corresponding channel matrix of relay i is

$$\mathbf{H}^{i} = \begin{bmatrix} h_{11}^{i} & h_{12}^{i} \\ h_{21}^{i} & h_{22}^{i} \end{bmatrix}. \tag{29}$$

The question now arises as to how best to utilize the finite number of bits available per sample in quantizing the output

³This definition is similar to the definition of an instantaneous relay (see, e.g., [9] and [10]), with the additional requirement of linearity while allowing a small delay at the relay.

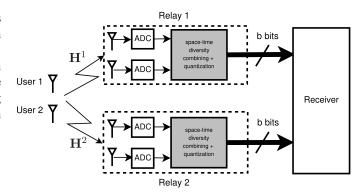


Fig. 7. Two-user virtual MIMO system formed by two-antenna relays connected to a receiver via rate-constrained fronthaul links.

of the two antennas. We note that MRC and selection diversity depend on the use of channel state information (CSI) (which is precluded by the definition of a "dumb" relay). Further, due to the distributed nature of the problem, both MRC and selection combining are also ineffective as the base station is interested in recovering both signals. Other methods such as naive local selection based on received power at a relay's antenna or equal gain combining are also futile.

More specifically, in order to perform useful CSI-dependent combining/selection at the relays, one would need to employ global channel state information, taking into account the channels from all users to all relays. For instance, to employ SC, the antenna selection at the relays would need to be performed jointly, the methods proposed in [12] (where linear equalization at the receiver was considered) being directly applicable. In the present context, such a process would need to take place in the cloud which would then notify each relay what combining/selection operation to employ. Such an approach, considering more general projection operations at the relays, has indeed been explored in the literature; we refer the reader to [13] and references therein.

We now demonstrate that while keeping the bit rate fixed, one can benefit (albeit, not to the extent as with full CSI) from additional antennas at the relays even without exploiting any channel state information at the relays. Specifically, each relay can provide diversity gains to *both users* using the proposed diversity combining method, precisely since it makes no use of CSI at the linear combining stage, rather only in the reconstruction stage.

Assuming both relays use the proposed space-time diversity combining scheme, the signal passed to the cloud from relay i is given by

$$\mathbf{y}^{i} = \mathbf{U}(h_{11}^{i}, h_{21}^{i})\mathbf{x}_{1} + \mathbf{U}(h_{12}^{i}, h_{22}^{i})\mathbf{x}_{2} + \mathbf{n'}^{i},$$

where \mathbf{x}_j represents the real representation of the signal transmitted by user j over the two time instances according to the notation in (3). Thus, at the cloud we obtain the effective channel

$$\begin{bmatrix} \mathbf{y}^1 \\ \mathbf{y}^2 \end{bmatrix} = \underbrace{\begin{bmatrix} \mathbf{U}(h_{11}^1, h_{12}^1) \, \big| \mathbf{U}(h_{21}^1, h_{22}^1)}{\mathbf{U}(h_{21}^2, h_{22}^2) \, \big| \mathbf{U}(h_{21}^2, h_{22}^2)} \end{bmatrix}}_{\mathcal{G}} \begin{bmatrix} \mathbf{x}_1 \\ \mathbf{x}_2 \end{bmatrix} + \begin{bmatrix} \mathbf{n'}^1 \\ \mathbf{n'}^2 \end{bmatrix}.$$

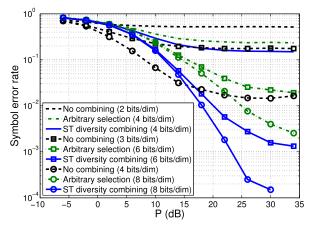


Fig. 8. Symbol error rate achieved using "dumb" relaying using the proposed diversity-combing scheme and comparison to baseline relaying schemes for uncoded 16-QAM transmission over a Rayleigh fading environment, where the receiver employs MMSE equalization.

Note that the effective matrix \mathcal{G} has the desirable property that each of the four submatrices is orthogonal. Thus, it is expected that applying linear equalization to the effective channel followed by a slicer (or in general, a decoder) will exhibit some diversity gain.

The performance of the proposed scheme is demonstrated in Figure 8 for a simple scenario where the users transmit uncoded 16-QAM symbols.4 We assume a simple receiver architecture that consists of linear equalization followed by single-user decoding. Employing such an architecture is reasonable since in our example, the effective channel \mathcal{G} is square (and well conditioned with high probability). As discussed in the previous section, when the number of transmitted streams is larger than the dimension of the received signal, multi-user detection techniques need be employed. As a baseline for comparison, we consider a relay that quantizes and forwards the output of an arbitrary antenna; or alternatively, a relay that quantizes and forwards the output of both antennas but with half the number of bits allocated to each quantizer.⁵ The latter is referred to as "no combining" in Figure 8. Since the output of both antennas is (albeit, coarsely) received, it enjoys an array (power) gain and thus performs better in the low SNR regime.

Substantial improvement may be seen with respect to the baseline schemes when a low bit error probability is desired, where we have considered quantization rates of 4, 6 and 8 bits per sample for each relay.⁶

V. EXTENSIONS TO MORE THAN TWO RECEIVE ANTENNAS

As in the case of space-time modulation for channel coding, extension of the scheme to more receive antennas is possible, albeit with some loss.

⁴We chose to simulate uncoded transmission to avoid the burden of computing the mutual information corresponding to quantized outputs. We believe that similar gains will be manifested in coded transmission.

⁵Since we assume "dumb" relays, the quantization of the inputs to the receiver was performed using a fixed (SNR independent) loading factor, taken as three times the standard deviation of the noise-free input to the quantizer.

⁶As the gains are more pronounced at high SNR, we chose to demonstrate the performance of 16-OAM rather than OPSK transmission.

A natural approach is to try utilizing the theory of orthogonal designs. It should be noted however that it is well known that the decoding delay (number of time instances stacked together) grows exponentially with the number of antennas (see, e.g., [14] and [15]). Another possible avenue is to try to follow the approach of quasi-orthogonal space-time codes as developed in [16]–[18]. We demonstrate both approaches.

Attempting to apply orthogonal designs, one immediately confronts a basic obstacle due to the fact that rate-1 complex orthogonal designs do not exist beyond the case of two antennas [19]. We next demonstrate the problem that arises and also show how it may be resolved by judiciously combining balanced rate-1/2 orthogonal designs [20] (which include the four basic OSTBCs described in [19] for 2-8 antennas) with repeated quantization used in conjunction with multiplicative dithering. For the sake of concreteness and ease of exposition, we demonstrate the method for the case of a SIMO system with M=4 receive antennas.

The received signals are given by (1) where now $i=1,\ldots,M$ (with M=4). We proceed by stacking T=8 time instances of the received signal from the 4 antennas and build an effective real-valued vector by decomposing each entry into its real and imaginary components, just as is done in (2). This yields for M=4, a vector s of dimension $2\times 4\times 8=64$.

By reinterpreting the rate-1/2 orthogonal design of 4 transmit antennas (see [19]), we arrive at a 8×64 transformation matrix \mathbf{G} . Next, we form a 8×1 real vector \mathbf{y} by applying to the effective received vector \mathbf{s} , formed in the manner described in (2), the transformation $\mathbf{y} = \mathbf{G}\mathbf{s}$.

It can be shown that the following holds

$$\mathbf{y} = \frac{\sqrt{2}\|\mathbf{h}\|}{\sqrt{8}} \mathbf{U}(h_1, h_2, h_3, h_4) \mathbf{x} + \mathbf{U}\mathbf{n}$$
$$= \frac{\|\mathbf{h}\|}{2} \mathbf{U}(h_1, h_2, h_3, h_4) \mathbf{x} + \mathbf{n}', \tag{30}$$

where $U(h_1, h_2, h_3, h_4)$ is an 8×16 matrix with orthonormal rows.⁸ Here, the vector \mathbf{x} is the 16-dimensional real representation of the transmitted signal over T=8 time instances, formed analogously to (3).

Since the rows of $U(h_1, h_2, h_3, h_4)$ are orthonormal, it follows that \mathbf{n}' is white (and Gaussian with variance 1/2).

The problem with using a non-rate 1 orthogonal design now becomes clear. Unlike $\mathbf{U}(h_1,h_2)$ (see (6)) which is square, $\mathbf{U}(h_1,h_2,h_3,h_4)$ is non-square and hence is non-invertible. We overcome this obstacle by passing the same observation vector \mathbf{s} via a "dithered" version of \mathbf{G} , such that another set of 8 mutually orthogonal measurement rows is attained. Specifically, let us define a 4-dimensional vector $\mathbf{d} = [d_1 \, d_2 \, d_3 \, d_4]^T$ where each d_i is a complex number of unit magnitude (pure phase). We form a dithered version of the antenna outputs as

$$\tilde{s}_i(t) = d_i \cdot s_i(t), \tag{31}$$

where d_i does not depend on t. We assume that the d_i are drawn at random as i.i.d. uniform phases.

⁷The specific form of G can be found in [21, eq. (36)].

⁸The specific form of $U(h_1, h_2, h_3, h_4)$ is given in [21, eq. (29)].

We may associate with $\tilde{s}_i(t)$, $t=1,\ldots,T=8$, the effective 64-dimensional real vector $\tilde{\mathbf{s}}$. Next, we obtain another 8-dimensional real vector $\tilde{\mathbf{u}}$ by applying to the vector $\tilde{\mathbf{s}}$ the transformation $\tilde{\mathbf{y}} = \mathbf{G}\tilde{\mathbf{s}}$. We therefore obtain

$$\tilde{\mathbf{y}} = \frac{\|\mathbf{h}\|}{2} \mathbf{U}(d_1 h_1, d_2 h_2, d_3 h_3, d_4 h_4) \mathbf{x} + \mathbf{n}'', \tag{32}$$

where n'' is distributed as n'.

Note that the dithers (31) may be absorbed in G, thus defining a "dithered" combining matrix $G_{\rm dith}$. Combining (30) and (32), we have

$$\underbrace{\begin{bmatrix} \mathbf{y} \\ \mathbf{\tilde{y}} \end{bmatrix}}_{\mathbf{y}_{\text{eff,dith}}} = \underbrace{\frac{\|\mathbf{h}\|}{2} \left[\frac{\mathbf{U}(h_1, h_2, h_3, h_4)}{\mathbf{U}(d_1 h_1, d_2 h_2, d_3 h_3, d_4 h_4)} \right]}_{\mathcal{F}_{\text{dith}}} \mathbf{x} + \begin{bmatrix} \mathbf{n}' \\ \mathbf{n}'' \end{bmatrix}.$$
(33)

Finally, we apply component-wise quantization to obtain

$$\mathbf{y}_q = Q(\mathbf{y}_{\text{eff,dith}}).$$
 (34)

We may then recover an estimate of x by applying the inverse of \mathcal{F} to y or a linear MMSE estimator.

As mentioned above, another approach to extend the basic scheme to more antennas is to borrow ideas from quasi-orthogonal space-time codes. As an example for a quasi-orthogonal space-time linear combining matrix, we construct a matrix $G_{\rm quasi}$ by taking half of the columns of ${\bf G}$, specifically columns 1-16 and 49-64, scaling by $\sqrt{2}$ to maintain power normalization. This results in

$$\mathbf{y}_{\text{eff,quasi}} = \underbrace{\frac{\|\mathbf{h}\|}{2} \mathbf{U}_{\text{quasi}}(h_1, h_2, h_3, h_4)}_{\mathcal{F}_{\text{quasi}}} \mathbf{x} + \mathbf{n}'$$
 (35)

where $U_{\text{quasi}}(h_1, h_2, h_3, h_4)$ is given by [21, eq. (35)].

We tested the performance attained with both combining matrices in the scenario considered in Section III. Specifically, Figure 9 depicts the mutual information achieved when using different linear-combining schemes, for the case of a Rayleigh fading 4×1 SIMO channel.

We observe that both $G_{\rm dith}$ and $G_{\rm quasi}$ achieve similar performance with some advantage to $G_{\rm quasi}$. Note that, in addition, $G_{\rm quasi}$ utilizes only four consecutive symbols whereas $G_{\rm dith}$ uses eight and hence induces less latency. On the other hand, the construction of $G_{\rm dith}$ can be readily extended to more antennas.

We further observe that as both variants of space-time diversity combining do not achieve orthogonality, the gap is from MRC is larger than the minimal gap one could hope for (had orthogonality been possible) which is a factor of four in power loss (\sim 6 dB). Specifically, for $G_{\rm quasi}$ the gap from MRC is roughly 7.5 dB in the SNR range simulated. Similarly to the case of a two-antenna system, the gap of SC to MRC is roughly half (in dB) that of the universal combining scheme.

VI. TIME-DOMAIN SUB-NYQUIST INTERPRETATION/APPLICATION

In this section we build on the well-known analogy between combining techniques for multiple-antenna arrays and those

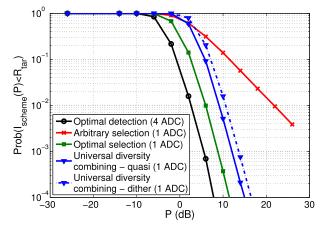


Fig. 9. Comparison of outage probability of different combining methods for an optimally encoded user transmitting over a 4×1 i.i.d. Rayleigh fading channel, with a target rate of $R_{\rm tar}=2$ bits per complex symbol.

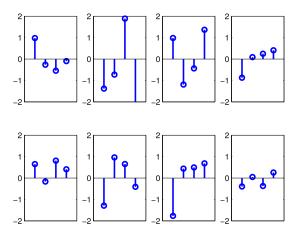


Fig. 10. Example of a pulse shape dictionary.

applied for time-domain signals. For example, the relation between maximal-ratio combining for antenna arrays and the sampled matched filter as an optimal front end (i.e., producing sufficient statistics) for a pulse-amplitude modulated time-domain signal is well recognized (see, e.g., [22]).

We describe how the developed diversity-combining technique may be leveraged to arrive at a sub-Nyquist signal acquisition system. Such systems may offer significant power/cost reductions as demonstrated in [23].

Applying universal diversity combining to sub-Nyquist signal acquisition is somewhat reminiscent of the compressive sensing problem (see, e.g., [24], [25]) where a universal dimension-reducing operation is also applied to reduce the sampling rate for signal acquisition. The difference lies in the channel/signal model assumptions. Whereas in compressive sensing the signal is assumed to satisfy sparsity constraints, in the scenario we consider, the signal is general whereas the channel is assumed to be fixed over several time instances. This assumption in turn corresponds to considering block-diagonal channel matrices with constant vector elements appearing on the diagonal (see below).

We demonstrate the method in the context of detection of pulse-amplitude modulated signals where the pulse shape plays the role of the channel vector. Specifically, suppose we

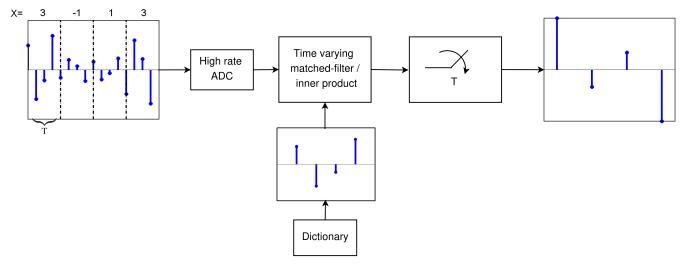


Fig. 11. Optimal analog-to-digital conversion via a time-varying sampled matched filter.

observe a (discrete-time) signal that is known to be of the from

$$\mathbf{s}(t) = \mathbf{h}x(t) + \mathbf{n}(t), \quad t = 1, \dots, K,$$
(36)

where $\mathbf{n}(t)$ is i.i.d. circularly-symmetric complex Gaussian noise of unit power and all vectors are column vectors of some dimension N. In other words, we know that the signal is sparse and lies in the K-dimensional subspace of \mathcal{C}^{NK} spanned by the vectors of the form $[\mathbf{0},\ldots,\mathbf{0},\mathbf{h}^T,\mathbf{0},\ldots,\mathbf{0}]^T$, as is the case in (the discrete-time representation of) pulse-amplitude modulation (PAM). We may view \mathbf{h} as a pulse shape of length N. The pulse shape used may change after NK time instants.

For example, for the case of K=3, the basis assumes the form

$$\begin{bmatrix} \mathbf{h} & 0 & 0 \\ 0 & \mathbf{h} & 0 \\ 0 & 0 & \mathbf{h} \end{bmatrix}. \tag{37}$$

An example for a scenario where the assumed model may be applicable to is sub-Nyquist detection of a frequency hopping signal. In a frequency hopping system, which is an effective method to combat jamming, the signal carrier is being chosen (based on a pseudo-random sequence) from a signal dictionary and is being changed at predefined symbol intervals.

As a concrete example, we may envision that h is a (complex) four-tap carrier signal in a PAM transmission system that is chosen pseudo randomly from a "dictionary". Figure 10 depicts the real part of such a possible dictionary consisting (in this example) of eight possible pulse shapes. The pulse shape chosen is kept constant for several symbols.

For simplicity of exposition, we assume that the receiver is synchronized in the sense that it knows when each pulse shape starts and ends and also knows which member of the dictionary is being used. Nonetheless, the proposed scheme may be advantageous in scenarios where h is unknown at the time of signal acquisition (sampling) and is revealed to the receiver end (or is estimated by it) subsequently.

Figure 11 depicts the optimal method for detection of the transmitted data. The signal is sampled at full rate. Specifically, assuming the pulse shape occupies T seconds in continuous

time, sampling at full rate means sampling at a rate of $\frac{N}{T}$ Hz (in the example, N=4). Then, a matched filter is applied and its output is sampled (at a rate of 1/T Hz) for data recovery.

We note that it is possible to implement the matched filter in the analog domain and sample at a rate of 1/T Hz. However, this requires implementing an analog filter bank, whose size should match that of the pulse shape dictionary and hence it is feasible only for small dictionaries.

The application of the proposed universal diversity method is demonstrated in Figure 12. In line with the duality mentioned above, the example considered where the pulse shape consists of four taps corresponds to a SIMO system with four receive antennas.

As was described in Section V, universal linear diversity combining is performed by applying an 8×32 precoding matrix $G_{\rm quasi}$ (assuming quasi-orthogonal precoding is used). The precoding matrix is applied to the stacked data of the real-valued representation of four consecutive PAM modulated complex symbols. We note that this requires implementing delay as well as summation and negation elements in the analog domain. After sampling, the data is recovered, e.g., by applying linear MMSE equalization with respect to $\mathcal{F}_{\rm quasi}$ followed by decoding.

We further note that one could also use tap selection (in analogy to antenna selection) to reduce the sampling rate, i.e., sample at the time corresponding to the strongest tap of the pulse shape to achieve even better performance than that of universal diversity combining. Nonetheless, whereas in the latter, the analog front end does not vary in time, optimal selection translates to applying a shift in the sampling time, every time the pulse shape changes.

Finally, we note that whereas when one considers an antenna array in a wireless fading environment, it is usually hard to expect that the channel coefficients remain constant over many symbols, thus precluding the use of space-time block codes when the number of antennas is large (as the needed coherence time grows exponentially with the number of antennas). In contrast, when considering the application to a time-domain signal, it is very reasonable to assume that the pulse shape remains constant over a long period of time and

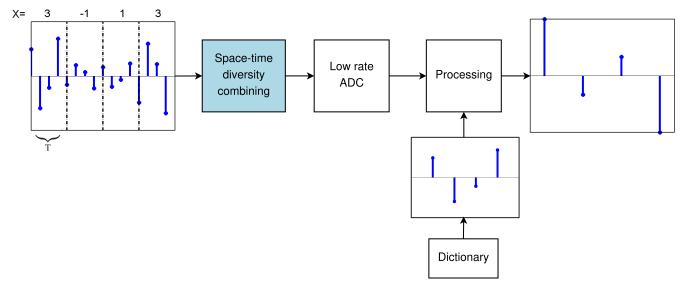


Fig. 12. Sub-Nyquist sampling via space-time diversity combining.

as a consequence, one can apply the proposed scheme to pulse shapes consisting of many taps (at the expense of considerable processing complexity in the digital domain).

APPENDIX

We start by establishing the limit

$$\lim_{P \to \infty} \frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} = 16,\tag{38}$$

where fixed R is any (fixed) target rate and where we consider the case of two users. We then observe that such an increase in outage probability corresponds to no more than a factor of two in terms of power penalty.

With some abuse of notation we now make the dependence of C(S) on P (as defined in (19)) explicit and denote it by C(S) = C(S, P).

To show that (38) holds, we show that

$$\limsup_{P \to \infty} \frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} \le 16, \tag{39}$$

and

$$\liminf_{P \to \infty} \frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,ont}}(P) < R\right)} \ge 16. \tag{40}$$

We start by showing the former inequality. By (23), for any rate R,

$$\Pr(C_{\text{sym,opt}}(P) < R) = \Pr(\min \{C_{\text{opt}}(\{1\}, P), C_{\text{opt}}(\{2\}, P), C_{\text{opt}}(\{1, 2\}, P)\} < R)$$

$$\stackrel{\text{(a)}}{\geq} \Pr\left(\bigcup_{k=1,2} \{C_{\text{opt}}(\{k\}, P) < R\}\right)$$

$$\stackrel{\text{(b)}}{=} \sum_{k=1}^{2} \Pr(C_{\text{opt}}(\{k\}, P) < R), \tag{41}$$

where (a) follows since we are taking into account only the events in which a single-user constraint constitutes the bottleneck and (b) follows since the fading coefficients are independent and hence so are the corresponding single-user constraint events.

As for the outage probability of diversity combining, invoking (26), we may upper bound it by applying the union bound:

$$\begin{split} & \Pr\left(C_{\text{sym,Ala}}(P) < R\right) \\ & = \Pr\left(\min\left\{C_{\text{Ala}}(\{1\}, P), C_{\text{Ala}}(\{2\}, P), \right. \\ & \left. C_{\text{Ala}}(\{1, 2\}, P)\right\} < R\right) \\ & \leq \sum_{k=1}^{2} \Pr\left(C_{\text{Ala}}(\{k\}, P) < R\right) + \Pr\left(C_{\text{Ala}}(\{1, 2\}, P) < R\right). \end{split} \tag{42}$$

For a general $N_r \times N_t$ i.i.d. Rayleigh fading channel between a user and receiver, the Frobenius norm squared $\|\mathbf{H}_k\|_F^2$ is chi-square distributed with $2 N_t N_r$ degrees of freedom. Further, it is readily shown that for small ϵ

$$\Pr\left(\|\mathbf{H}_k\|_F^2 < \epsilon\right) = c(N_t N_r) \cdot \epsilon^{N_t N_r} + o\left(\epsilon^{N_t N_r}\right), \quad (43)$$

where $o(\epsilon)/\epsilon \to 0$ as $\epsilon \to 0$ and where $c(N_t N_r)$ is a constant that depends only on $N_t N_r$.

In the case considered of 1×2 channel matrices, $\|\mathbf{H}_k\|_F^2 = \|\mathbf{h}_k\|^2$ and $N_t N_r = 2$. Therefore,

$$\Pr\left(C_{\text{opt}}(\{k\}, P) < R\right) = \Pr\left(\log\left(1 + P\|\mathbf{h}_k\|^2\right) < R\right)$$
$$= \Pr\left(\|\mathbf{h}_k\|^2 < \frac{2^R - 1}{P}\right)$$
$$= c(4)\left(\frac{2^R - 1}{P}\right)^4 + o\left(\frac{1}{P^4}\right). \tag{44}$$

⁹It can be shown, e.g., that for $N_t N_r = 4$, $c(4) = \frac{1}{384}$.

Similarly,

$$\Pr\left(C_{\text{Ala}}(\{k\}, P) < R\right) = c(4) \left(\frac{2^R - 1}{P/2}\right)^4 + o\left(\frac{1}{P^4}\right)$$
(45)

and

$$\Pr\left(C_{\text{Ala}}(\{1,2\},P) < R\right) = c(8) \left(\frac{2^R - 1}{P/2}\right)^8 + o\left(\frac{1}{P^8}\right). \tag{46}$$

Combining these asymptotics with (41) and (42) yields

$$\frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)}$$

$$\leq \frac{\sum_{k=1}^{2} \Pr\left(C_{\text{Ala}}(\{k\}, P) < R\right) + \Pr\left(C_{\text{Ala}}(\{1, 2\}, P) < R\right)}{\sum_{k=1}^{2} \Pr\left(C_{\text{opt}}(\{k\}, P) < R\right)}$$

$$= \frac{2c(4)\left(\frac{2^{R}-1}{P/2}\right)^{4} + o\left(\frac{1}{P^{4}}\right) + c(8)\left(\frac{2^{R}-1}{P/2}\right)^{8} + o\left(\frac{1}{P^{8}}\right)}{2c(4)\left(\frac{2^{R}-1}{P}\right)^{4} + o\left(\frac{1}{P^{4}}\right)}$$

$$\xrightarrow{P \to \infty} 16. \tag{47}$$

Hence

$$\limsup_{P \to \infty} \frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} \le 16. \tag{48}$$

We turn now to establishing (40). Applying the same arguments as in (41) to $C_{\text{sym,Ala}}(P)$, we get

$$\Pr\left(C_{\text{sym,Ala}}(P) < R\right) \ge \sum_{k=1}^{2} \Pr\left(C_{\text{Ala}}(\{k\}, P) < R\right).$$
 (49)

Applying (42) to $C_{\text{sym,opt}}(P)$, we get

$$\Pr(C_{\text{sym,opt}}(P) < R)$$

$$\leq \sum_{k=1}^{2} \Pr(C_{\text{opt}}(\{k\}, P) < R) + \Pr(C_{\text{opt}}(\{1, 2\}, P) < R).$$
(50)

Recalling (23) and [26, eq. (5)], we have

$$C_{\text{opt}}(\{1,2\}) = \log \det \left(\mathbf{I} + P \cdot \mathbf{H}_{\text{comb}} \mathbf{H}_{\text{comb}}^{H} \right)$$

$$\geq \log \left(1 + P \| \mathbf{H}_{\text{comb}} \|_{F}^{2} \right)$$
 (51)

and hence

$$\Pr\left(C_{\text{opt}}(\{1,2\},P) < R\right) \le \Pr\left(\log\left(1+P\|\mathbf{H}_{\text{comb}}\|_F^2\right) < R\right)$$

$$= \Pr\left(\|\mathbf{H}_{\text{comb}}\|_F^2 < \frac{2^R - 1}{P}\right)$$

$$= c(8)\left(\frac{2^R - 1}{P}\right)^8 + o\left(\frac{1}{P^8}\right).$$
(52)

It follows that

$$\Pr\left(C_{\text{Ala}}(\{k\}, P) < R\right) = c(4) \left(\frac{2^{R} - 1}{P/2}\right)^{4} + o\left(\frac{1}{P^{4}}\right) \qquad \Pr\left(C_{\text{sym,opt}}(P) < R\right) \\ \leq 2c(4) \left(\frac{2^{R} - 1}{P}\right)^{4} + o\left(\frac{1}{P^{4}}\right) + c(8) \left(\frac{2^{R} - 1}{P}\right)^{8} + o\left(\frac{1}{P^{8}}\right)$$
(53)

Combining (49) and (53), we obtain

$$\frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)}$$

$$\geq \frac{\sum_{k=1}^{2} \Pr\left(C_{\text{Ala}}(\{k\}, P) < R\right)}{\sum_{k=1}^{2} \Pr\left(C_{\text{opt}}(\{k\}, P) < R\right) + \Pr\left(C_{\text{opt}}(\{1, 2\}, P) < R\right)}$$

$$= \frac{2c(4)\left(\frac{2^{R}-1}{P}\right)^{4} + o\left(\frac{1}{P^{4}}\right)}{2c(4)\left(\frac{2^{R}-1}{P}\right)^{4} + o\left(\frac{1}{P^{4}}\right) + c(8)\left(\frac{2^{R}-1}{P}\right)^{8} + o\left(\frac{1}{P^{8}}\right)}$$

$$\xrightarrow{P \to \infty} 16. \tag{54}$$

Hence,

$$\liminf_{P \to \infty} \frac{\Pr\left(C_{\text{sym,Ala}}(P) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} \ge 16.$$
(55)

Combining (48) and (55), we have established (38).

Next, we show that a reduction by a factor of two in transmission power asymptotically as $P \to \infty$ translates to an increase by the same factor of 16 in outage probability, i.e.

$$\lim_{P \to \infty} \frac{\Pr\left(C_{\text{sym,opt}}(P/2) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} = 16.$$
 (56)

To this end, by the union bound, we have

$$\Pr\left(C_{\text{sym,opt}}(P/2) < R\right)$$

$$\leq \sum_{k=1}^{2} \Pr\left(C_{\text{opt}}(\{k\}, P/2) < R\right)$$

$$+\Pr\left(C_{\text{opt}}(\{1, 2\}, P/2) < R\right), \tag{57}$$

which combined with (41), implies that

$$\frac{\Pr\left(C_{\text{sym,opt}}(P/2) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)}$$

$$\leq \frac{2c(4)\left(\frac{2^{R}-1}{P/2}\right)^{4} + o\left(\frac{1}{P^{4}}\right) + c(8)\left(\frac{2^{R}-1}{P/2}\right)^{8} + o\left(\frac{1}{P^{8}}\right)}{2c(4)\left(\frac{2^{R}-1}{P}\right)^{4} + o\left(\frac{1}{P^{4}}\right)}$$

$$\frac{P \to \infty}{16}, \qquad (58)$$

where we have used the asymptotics (53) for the denominator and Hence.

$$\limsup_{P \to \infty} \frac{\Pr\left(C_{\text{sym,opt}}(P/2) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} \le 16.$$
 (59)

Next, from (41), we have

$$= c(8) \left(\frac{2^{R} - 1}{P}\right)^{8} + o\left(\frac{1}{P^{8}}\right). \qquad \Pr\left(C_{\text{sym,opt}}(P/2) < R\right) \ge \sum_{k=1}^{2} \Pr\left(C_{\text{opt}}(\{k\}, P/2) < R\right), \tag{60}$$

which combined with (53), implies that

$$\frac{\Pr\left(C_{\text{sym,opt}}(P/2) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)}$$

$$\geq \frac{2c(4)\left(\frac{2^{R}-1}{P/2}\right)^{4} + o\left(\frac{1}{P^{4}}\right)}{2c(4)\left(\frac{2^{R}-1}{P}\right)^{4} + o\left(\frac{1}{P^{4}}\right) + c(8)\left(\frac{2^{R}-1}{P}\right)^{8} + o\left(\frac{1}{P^{8}}\right)}$$

$$\xrightarrow{P \to \infty} 16. \tag{61}$$

Hence,

$$\liminf_{P \to \infty} \frac{\Pr\left(C_{\text{sym,opt}}(P/2) < R\right)}{\Pr\left(C_{\text{sym,opt}}(P) < R\right)} \ge 16.$$
(62)

Combining (59) and (62), we have established (56).

Therefore, optimal detection for transmission at half the power yields the same outage probability as that achieved with diversity combining, at asymptotically high SNR.

We conclude by noting the proof of the claim for a general number of users follows along similar lines. This can be seen by noticing that the dominant terms in the outage probability (for asymptotically high SNR) are the ones corresponding to single-user constraints. The latter can be approximated at high SNR using (43), tracing the same steps as for the two-user case.

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