

Research Letter

Combining Coded Signals with Arbitrary Modulations in Orthogonal Relay Channels

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Received 21 March 2008; Accepted 16 July 2008

Recommended by Huseyin Arslan

We consider a relay channel for which the following assumptions are made. (1) The source-destination and relay-destination channels are orthogonal (frequency division relay channel). (2) The relay implements the decode-and-forward protocol. (3) The source and relay implement the same channel encoder, namely, a convolutional encoder. (4) They can use arbitrary and possibly different modulations. In this framework, we derive the best combiner in the sense of the maximum likelihood (ML) at the destination and the branch metrics of the trellis associated with its channel decoder for the ML combiner and also for the maximum ratio combiner (MRC), cooperative-MRC (C-MRC), and the minimum mean-square error (MMSE) combiner.

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1. Motivations and Technical Background

We consider orthogonal relay channels for which orthogonality is implemented in frequency [1]. Since the source-destination channel is assumed to be orthogonal to the relay-destination channel, the destination receives two distinct signals. In order to maintain the receiver complexity at a low level, the destination is imposed to combine the received signals before applying channel decoding. The relay is assumed to implement the decode-and-forward (DF) protocol. We have at least two motivations for this choice. First, in contrast with the well-known amplify-and-forward (AF) protocol, it can be implemented in a digital relay transceiver. More importantly, whereas the AF protocol imposes the source-relay channel to have the same bandwidth as the relay-destination channel, the DF protocol offers some degrees of freedom in this respect. This is a critical point when the cooperative network has to be designed from the association of two existing networks. For instance, if one wants to increase the performance of a digital video broadcasting (DVB) receiver or reach some uncovered indoor areas, a possible solution is to use cell phones, say universal mobile telecommunications system (UMTS) cell phones as relaying nodes. The problem is that DVB signals use a 20 MHz

bandwidth (source-relay channel) while UMTS signals have only a bandwidth of 5 MHz (relay-destination channel). The AF protocol cannot be used here; but the DF protocol can be used, for instance, by adapting the modulation of the cooperative signal to the available bandwidth. In this case, the destination has to combine two signals with different modulations.

In this context, one of the issues that needs to be addressed is the design of the combiner. A conventional MRC cannot be used for combining signals with different modulations (except for special cases of modulations). Even if the modulations at the source and relay are identical, the MRC can severely degrade the receiver performance because it does not compensate for the decoding noise introduced by the relay [2–6]. This is why the authors of [2, 4] proposed a maximum-likelihood detector (MLD) for combining two BPSK-modulated signals coming from the source and relay. The authors of [6] proposed an improved MRC called C-MRC which aims at maximizing receive diversity. The authors of [3] proposed a linear combiner for which the weights are tuned to minimize the raw bit error rate (BER). The main issue is that one is not always able to explicit the raw BER as a function of the combiner weights whereas the likelihood calculation is more systematic. Additionally, when

some a priori knowledge is available, the ML metric can be used to calculate an a posteriori probability (APP). In the context of orthogonal N -relay channels, the authors of [4] derived two new combiners: the best MRC in the sense of the equivalent signal-to-noise ratio and MMSE combiner. They also assessed the BER performance of the latter and MLD in the uncoded case.

Compared to the afore-mentioned works, this paper also aims at designing a good combiner at the destination but it differs from them on two essential points. (1) The interaction between the combiner and channel decoder is exploited in the sense that we want to express the branch metrics of the trellis associated with channel decoding for the MRC, MMSE combiner, C-MRC, and especially for the ML combiner. (2) When the ML combiner is assumed, the source and relay can use arbitrary modulations (not necessarily BPSK modulations as in [2, 3, 5]) and, more importantly, these can be different.

2. Signal Model

At the source, the L -information bit sequence \underline{m} is encoded into a sequence of bits \underline{b} and modulated into the transmitted signal $\underline{x} = (x(1), \dots, x(T))$, where for all $t \in \{1, \dots, T\}$, $x(t) \in \mathcal{X}$, \mathcal{X} is a finite alphabet corresponding to the modulation constellation used by the source and $\mathbb{E}|x(t)|^2 \leq P_0$. At the relay, the message is decoded, re-encoded with the same encoder as the source and modulated into the transmitted signal $\underline{x}_1 = (x_1(1), \dots, x_1(T_1))$ where for all $t_1 \in \{1, \dots, T_1\}$, $x_1(t_1) \in \mathcal{X}_1$, \mathcal{X}_1 is a finite alphabet corresponding to the modulation constellation used by the relay, and $\mathbb{E}|x_1(t_1)|^2 \leq P_1$. We denote by s (resp., r) the number of coded bits conveyed by one source (resp., relay) symbol. By definition: $s = \log_2 |\mathcal{X}|$ and $r = \log_2 |\mathcal{X}_1|$. More specifically, the information bit sequence is assumed to be encoded by a $1/q$ -rate convolutional encoder ($q \in \mathbb{N}^*$). As the sequence \underline{x} comprises T symbols, we have that $q(k + \nu) = sT$ where ν is the channel encoder memory. Assuming time-selective but frequency-nonselective channels, the baseband signals received by the destination from the source and relay, respectively, write $y_0(t) = h_0x(t) + z_0(t)$ and $y_1(t_1) = h_1x_1(t_1) + z_1(t_1)$, where z_0 and z_1 are zero-mean circularly symmetric complex Gaussian noises with variances σ_0^2 and σ_1^2 , respectively. The complex coefficients h_0 and h_1 represent the gains of the source-destination and source-relay fading channels. For insuring coherent decoding, these two gains are assumed to be known to the receiver and relay, respectively. We define $\gamma_0 = \mathbb{E}|h_0|^2(P/\sigma_0^2)$, $\gamma_1 = \mathbb{E}|h_1|^2(P/\sigma_1^2)$, $\gamma'_1 = \mathbb{E}|h'_1|^2(P/\sigma_0^2)$, and $\rho_1 = \mathbb{E}|X_1X^*|/P$, where h'_1 is the gain of the source-relay fading channel. Note that, in order to ensure the conservation of the coded bit rate between the input and output of the relay, s and r have to be linked by the following compatibility relation: $sT = rT_1$. In the sequel, we will use the quantity $k = \text{lcm}(s, r)$, where lcm is the least common multiple function. For simplicity, we assume that the source and relay use the same channel coder. Therefore, the relay has to use a modulation that is compatible with the source's one. We will also assume that the number of times per

second the channel can be used is directly proportional to the available bandwidth. For example, if the source uses a BPSK modulation and the cooperation channel has a bandwidth equal to half the downlink bandwidth, the relay can use a QPSK modulation.

3. A new Trellis Branch Metric

3.1. When the Source and Relay Use Arbitrary and Different Modulations

In this case, the linear combiners derived by [3, 4, 6] cannot be used in general. However, provided that the above compatibility condition is met, the ML combiner can be derived as we show now. Let us denote by \underline{y}_0 and \underline{y}_1 the sequences of noisy symbols received by the destination from the source and relay, respectively. The discrete optimization problem the ML combiner solves is as follows:

$$\hat{\underline{m}} = \arg \max_{\underline{m} \in \mathbb{F}_2^L} p_{\text{ML}} = \arg \max_{\underline{m} \in \mathbb{F}_2^L} p(\underline{y}_0, \underline{y}_1 | \underline{x}). \quad (1)$$

As the reception noises are assumed to be independent, $p_{\text{ML}} = p(\underline{y}_0 | \underline{x})p(\underline{y}_1 | \underline{x})$. The first term easily writes as

$$p(\underline{y}_0 | \underline{x}) = \prod_{t=1}^T \frac{1}{\pi\sigma_0^2} \exp\left(-\frac{|y_0(t) - h_0x(t)|^2}{\sigma_0^2}\right). \quad (2)$$

In order to express the second term, we introduce a sequence of T_1 discrete symbols denoted by \underline{e}_1 which models the residual noise at the relay after the decoding-reencoding process. This noise is, therefore, modeled by a multiplicative error term which is not independent of the symbols transmitted by the relay. Additionally, the statistics of this noise are assumed to be known by the destination. For this, one can establish once and for all a lookup table between the source-relay SNR and the bit error rate after reencoding at the relay. The cooperation signal writes then as $y_1(t_1) = h_1x_1(t_1) + z_1(t_1)$, where $x_1(t_1) = \underline{e}_1(t_1)\tilde{x}_1(t_1)$ and $\tilde{x}_1(t_1)$ is the symbol the relay would generate if there were no decoding error at the relay. For example, when the relay uses a QPSK modulation, $e_1 \in \{1, e^{j(\pi/2)}, e^{j\pi}, e^{j(3\pi/2)}\}$. Therefore, we have that $p(\underline{y}_1 | \underline{x}) = p(\underline{y}_1 | \tilde{\underline{x}}_1) = \sum_{\underline{e}_1} p(\underline{y}_1, \underline{e}_1 | \tilde{\underline{x}}_1) = \sum_{\underline{e}_1} p(\underline{e}_1 | \tilde{\underline{x}}_1)p(\underline{y}_1 | \tilde{\underline{x}}_1, \underline{e}_1)$. At this point, we need to make an additional assumption in order to easily derive the path metric of the ML decoder. From now on, we assume that the discrete symbols of the sequence \underline{e}_1 are conditionally independent. This assumption is very realistic, for example, if the source and relay implement a bit-interleaved coded modulation (BICM) or a trellis-coded modulation (TCM). In the case of the BICM, the channel coder, which generates coded bits, and the modulator are separated by an interleaver. The presence of this interleaver precisely makes the proposed assumption reasonable. Under

the afore-mentioned assumption, one can expand $p(\underline{y}_1 | \underline{x})$ as

$$\begin{aligned} p(\underline{y}_1 | \underline{x}) &= \sum_{\underline{\epsilon}_1} \prod_{t_1=1}^{T_1} p(e_1(t_1) | \tilde{x}_1(t_1)) p(y_1(t_1) | \tilde{x}_1(t_1), e_1(t_1)) \\ &= \sum_{\underline{\epsilon}_1} \prod_{t_1=1}^{T_1} p(e_1(t_1) | \tilde{x}_1(t_1)) \\ &\quad \times \frac{1}{\pi \sigma_1^2} \exp\left(-\frac{|y_1(t_1) - h_1 e_1(t_1) \tilde{x}_1(t_1)|^2}{\sigma_1^2}\right). \end{aligned} \quad (3)$$

The main consequence of this assumption is a significant reduction of the decoder complexity. If the assumption is not valid, the proposed derivation can always be used but the performance gain obtained can be marginal since the errors produced by will not be spread over the data block but rather occurs in a sporadic manner along the block.

In order to express the path metric of a given path in the trellis associated with channel decoding, we need now to link the likelihood expressed above and the likelihood associated with a given bit b_j , where $j \in \{1, \dots, k\}$. The reason why we consider subblocks of k bits is that in order to meet the rate compatibility condition, the ML combiner combines the $k_s = k/s$ symbols received from the source with the $k_r = k/r$ symbols received from the relay. Now for all $(i, j) \in \{0, 1\} \times \{1, \dots, k\}$, let us define the sets $\mathcal{B}_i^{(k)}(j) = \{\underline{b}^k = (b_1, \dots, b_k) \in \{0, 1\}^k, b_j = i\}$, a set of subblocks of rs consecutive bits, $\mathcal{X}_i^{(k_s)}(j) = \{\underline{x}^{k_s} \in \mathcal{X}^{k_s}, b_j = i\}$, and $\mathcal{X}_{1,i}^{(k_r)}(j) = \{\tilde{x}_1^{k_r} \in \mathcal{X}_1^{k_r}, b_j = i\}$, their equivalents in the source (resp., relay) modulation space. With these notations, the bit likelihood can be expressed as follows:

$$\begin{aligned} \lambda(b_j = i) &= \ln \left[\sum_{\underline{b}_1^k \in \mathcal{B}_i^{(k)}(j)} p(\underline{y}_{0,1}^{k_s} | \underline{x}_1^{k_s}) p(\underline{y}_{1,1}^{k_r} | \tilde{x}_1^{k_r}) \right] \\ &= \ln \left[\sum_{\underline{b}_1^k \in \mathcal{B}_i^{(k)}(j)} \prod_{t=1}^{k_s} p(y_0(t) | x(t)) \prod_{t_1=1}^{k_r} p(y_1(t_1) | \tilde{x}_1(t_1)) \right], \end{aligned} \quad (4)$$

where we used the notation $\underline{v}_1^n = (v(1), \dots, v(n))$. When a BICM is used, the obtained log-likelihood sequence is then de-interleaved and given to a Viterbi decoder.

3.2. When the Source and Relay Use Arbitrary and Identical Modulations

The derivation of the coded-bit likelihood in the case where the modulations used by the source and relay are the same is ready since it is special case of derivation conducted previously with $k = s = r$. In this case, both ML and linear combiners can be used since the combination can be performed symbol-by-symbol. The log-likelihood becomes $\lambda(b_j = i) = \ln[\sum_{\underline{b}_1^s \in \mathcal{B}_i^{(s)}(j)} p(y_0(t) | x(t)) p(y_1(t) | x(t))]$, where

TABLE 1: Equivalent channel parameters for the linear combiners.

	h_{eq}	σ_{eq}^2
MRC	$\frac{a_0}{\sigma_0^2} + \frac{a_1}{\sigma_1^2}$	$\frac{a_0}{\sigma_0^2} + \frac{a_1}{\sigma_1^2}$
MMSE	$\frac{a_0}{\sigma_0^2} + \frac{a_1 \rho_1^2}{\sigma_1^2 + a_1 P_0 (\alpha_1^2 - \rho_1^2)}$	$\frac{a_0}{\sigma_0^2} + \frac{a_1 \rho_1^2}{\sigma_1^2 + a_1 P_0 (\alpha_1^2 - \rho_1^2)}$
C-MRC	$a_0 + \frac{\min\{\gamma'_1, \gamma_1\}}{\gamma_1} a_1$	$a_0 \sigma_0^2 + \frac{\min\{\gamma'_1, \gamma_1\}}{\gamma_1} a_1 \sigma_1^2$

$1 \leq j \leq s$. If we further assume that the modulations used are BPSK modulations, the likelihood on the received sequences takes a more explicit form. Indeed, it can be checked that

$$\begin{aligned} \ln [p(\underline{y}_0, \underline{y}_1 | \underline{m})] &= -kq \ln(2\pi) - T \ln(\sigma_0^2 \sigma_1^2) \\ &\quad - \sum_{t=1}^T \left[\frac{|y_0(t) - h_0 x(t)|^2}{\sigma_0^2} - \ln \left[\sum_{e=-1,+1} \Pr[\epsilon(t) = e] \right. \right. \\ &\quad \left. \left. \times \exp\left(-\frac{|y_1(t) - h_1 e x(t)|^2}{\sigma_1^2}\right) \right] \right], \end{aligned} \quad (5)$$

where $\Pr[\epsilon = -1]$ represents the residual bit error rate (after the decoding-reencoding procedure inherent to DF protocol). Denote by π the interleaver function such that $t = \pi(t_0)$ and $t_0 = \pi^{-1}(t)$. Finally, the path metric is merely given by

$$\begin{aligned} \mu^{\underline{x}} &= \sum_{t_0=1}^T \left[\frac{|y_0(t_0) - h_0 x(t_0)|^2}{\sigma_0^2} - \ln \left[\sum_{e=-1,+1} \Pr(\epsilon(t_0) = e) \right. \right. \\ &\quad \left. \left. \times \exp\left(\frac{|y_1(t_0) - h_1 e x(t_0)|^2}{\sigma_1^2}\right) \right] \right]. \end{aligned} \quad (6)$$

So the combining and channel decoding are performed jointly by modifying the branch metrics as indicated above.

When using a linear combiner, one has to compute the APP from the equivalent signal at the combiner output. This computation requires the equivalent channel gain and noise. We provide them for each linear combiner considered here. For a given combiner, denote its optimal vector of weights by $\underline{w} = (w_0, w_1)$ and rewrite the signal at the combiner output as $y = \sum_{i=0}^1 w_i y_i = h_{\text{eq}} x + z_{\text{eq}}$, where h_{eq} and $z_{\text{eq}} \sim \mathcal{N}(0, \sigma_{\text{eq}}^2)$ are the equivalent channel gain and noise, respectively. The bit log-likelihood can then be easily expressed as $\lambda(b_j = i) = \ln[\sum_{\underline{x} \in \mathcal{X}_i^{(s)}(j)} p(\underline{y} | h_{\text{eq}} \underline{x})]$. Table 1 summarizes the values of these quantities with the notations $a_0 = |h_0|^2$ and $a_1 = |h_1|^2$.

4. Simulation Example

For Figures 1 and 2, we assume that the source and the relay implement a 1/2-rate convolutional encoder (4-state encoder

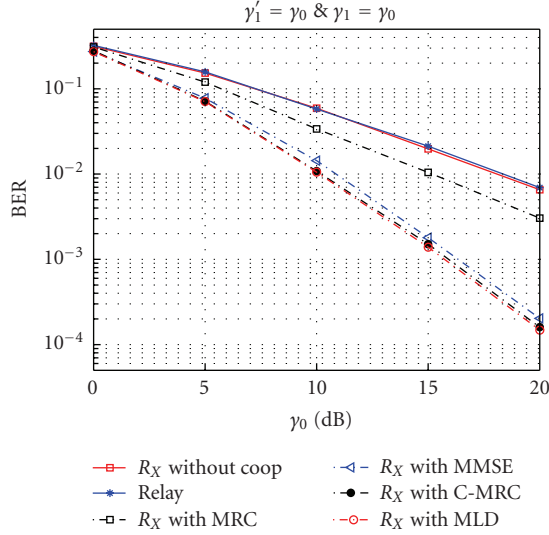


FIGURE 1: BER at the destination with $\gamma'_1 = \gamma_0, \gamma_1 = \gamma_0$.

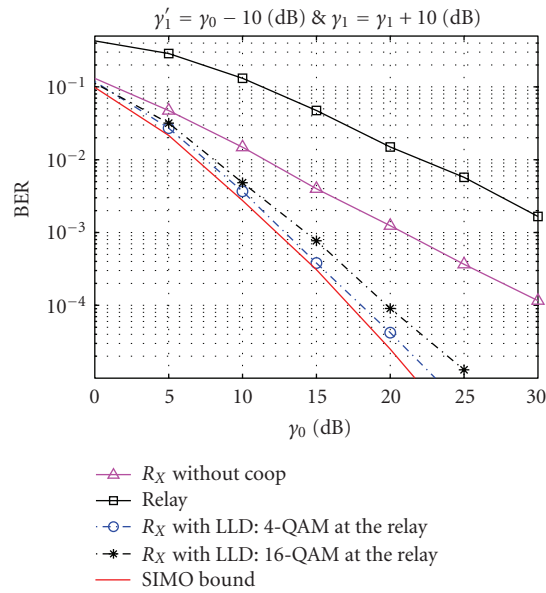


FIGURE 2: BER at the destination with $\gamma'_1 = \gamma_0 - 10, \gamma_1 = \gamma_0 + 10$.

with a free distance equal to 5). Frequency nonselective Rayleigh block fading channels are assumed and the data block length is chosen to be 1024. First, we compare the combiners between themselves when both the relay and source use a 4-QAM modulation. Figure 1 represents the BER at the decoder output as a function of $\gamma_0 = \gamma_1$. There are 6 curves: from the top to the bottom, they, respectively, represent the performance with no relay, with the relay associated with the conventional MRC, MMSE, C-MRC, and ML combiners. When implementing the conventional MRC, the receiver does not significantly improve its performance with respect to the noncooperative case whereas the other combiners can provide more than an 8 dB gain and perform quite similarly. Then (see Figure 2), we evaluate the perfor-

mance gain brought by the MLD when the source and relay have to use different modulations: the source implements a BPSK while the relay implements either a 4-QAM or a 16-QAM. The second scenario would correspond to a case where the source-destination channel bandwidth is 4 times larger than the relay-destination channel bandwidth (e.g., 20 MHz versus 5 MHz). We see that the MLD not only makes cooperation possible but also allows the destination to extract a significant performance gain from it. To have an additional reference, we also represented the performance of the equivalent virtual 1×2 MIMO system, which is obtained for $\gamma_1 = +\infty$.

5. Concluding Remarks

The results provided in this letter and many other simulations performed in the *coded* case led us to the following conclusion: if the source and relay can use the same modulation, the C-MRC generally offers the best performance-complexity tradeoff. On the other hand, if the modulations are *different*, as it would be generally the case when two *existing* communications systems are associated to cooperate, linear combiners and thus the C-MRC cannot be used in general and the ML combiner is the only implementable combiner.

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Special Issue on Femtocell Networks

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Recently, there has been a growing interest in femtocell networks both in academia and industry. They offer significant advantages for next-generation broadband wireless communication systems. For example, they eliminate the dead-spots in a macrocellular network. Moreover, due to short communication distances (on the order of tens of meters), they offer significantly better signal qualities compared to the current cellular networks. This makes high-quality voice communications and high data rate multimedia type of applications possible in indoor environments.

However, this new type of technology also comes with its own challenges, and there are significant technical problems that need to be addressed for successful deployment and operation of these networks. Standardization efforts related to femtocell networks in 3GPP (e.g., under TSG-RAN Working Group 4 and LTE-Advanced) and IEEE (e.g., under IEEE 802.16m) are already underway.

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- Mobility support and handover
- Time synchronization
- Multiple antenna techniques
- Tradeoffs between femtocells, picocells, relay networks, and antenna arrays
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Special Issue on Network Coding for Wireless Networks

Call for Papers

The main idea in network coding was introduced in 2000 by Ahlswede et al. With network coding, an intermediate node in a network cannot only forward its incoming packets but also encode them. It has been shown that the use of network coding can enhance the performance of wired networks significantly. Recent work indicates that network coding can also offer significant benefits for wireless networks.

Communications over wireless channels are error-prone and unpredictable due to fading, mobility, and intermittent connectivity. Moreover, in wireless networks, transmissions are broadcast and can be overheard by neighbors, which is treated in current systems as interference. Finally, security poses new challenges in wireless networks, where both passive and active attacks have quite different premises than in wired networks. Ideas in network coding promise to help toward all these issues, allowing to gracefully add redundancy to combat errors, take advantage of the broadcast nature of the wireless medium and achieve opportunistic diversity, exploit interference rather than be limited by it, and provide secure network communication against adversarial attacks.

In this special issue, we are interested in original research articles which can carry the momentum further and take the wireless network coding research to the next level. The areas of interest include novel network code designs and algorithms, new applications of wireless network coding, network coding capacity, and performance analysis. In addition to original research articles, we are open to review articles. The following list indicates topics of interest which is by no means exhaustive:

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