

Quantized vs. Analog Feedback for the MIMO Broadcast Channel: A Comparison between Zero-Forcing Based Achievable Rates

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Abstract— We consider a MIMO fading broadcast channel and compare the achievable ergodic rates when the channel state information at the transmitter is provided by “analog” noisy feedback or by quantized (digital) feedback. The superiority of digital feedback is shown, with perfect or imperfect CSIR, whenever the number of feedback channel uses per channel coefficient is larger than 1. Also, we show that by proper design of the digital feedback link, errors in the feedback have a minor effect even by using very simple uncoded modulation. Finally, we show that analog feedback achieves a fraction $1 - 2F$ of the optimal multiplexing gain even in the presence of a feedback delay, when the fading belongs to the class of “Doppler processes” with normalized maximum Doppler frequency shift $0 \leq F < 1/2$.

I. MODEL SETUP AND BACKGROUND

We consider a multi-input multi-output (MIMO) Gaussian broadcast channel modeling the downlink of a system where the base station (transmitter) has M antennas and K user terminals (receivers) have one antenna each. A channel use of such channel is described by

$$y_k = \mathbf{h}_k^H \mathbf{x} + z_k, \quad k = 1, \dots, K \quad (1)$$

where y_k is the channel output at receiver k , $z_k \sim \mathcal{CN}(0, N_0)$ is the corresponding AWGN, $\mathbf{h}_k \in \mathbb{C}^M$ is the vector of channel coefficients from the k -th receiver to the transmitter antenna array and \mathbf{x} is the channel input vector. The channel input is subject to the average power constraint $\mathbb{E}[|\mathbf{x}|^2] \leq P$.

We assume that the channel *state*, given by the collection of all channel vectors $\mathbf{H} = [\mathbf{h}_1, \dots, \mathbf{h}_K] \in \mathbb{C}^{M \times K}$, varies in time according to a block fading model where \mathbf{H} is constant over each frame of length T channel uses, and evolves from frame to frame according to an ergodic stationary jointly Gaussian process; i.i.d. block-fading channel, where the entries of \mathbf{H} are Gaussian i.i.d. with elements $\sim \mathcal{CN}(0, 1)$ is a special case of this.

A. Capacity results

If \mathbf{H} is perfectly and instantaneously known to all terminals (perfect CSIT and CSIR), the capacity region of the channel (1) is obtained by MMSE-DFE beamforming and Gaussian dirty-paper coding (see [1, 2] and references therein). Because of simplicity and robustness to non-perfect CSIT, simpler *linear precoding* schemes with standard Gaussian coding have been extensively considered. A particularly simple scheme consists of zero-forcing (ZF) beamforming, where the transmit

signal is formed as $\mathbf{x} = \mathbf{V}\mathbf{u}$, such that $\mathbf{V} \in \mathbb{C}^{M \times K}$ is a zero-forcing beamforming matrix and $\mathbf{u} \in \mathbb{C}^K$ contains the symbols from K independently generated Gaussian codewords. For $K \leq M$, the k -th column \mathbf{v}_k of \mathbf{V} is chosen to be a unit vector orthogonal to the subspace $\mathcal{S}_k = \text{span}\{\mathbf{h}_j : j \neq k\}$. In this case, the achievable sum rate is given by

$$R^{\text{ZF}} = \max_{\sum_k \mathbb{E}[\mathcal{P}_k(\mathbf{H})] \leq P} \sum_{k=1}^K \mathbb{E} \left[\log \left(1 + \frac{|\mathbf{h}_k^H \mathbf{v}_k|^2 \mathcal{P}_k(\mathbf{H})}{N_0} \right) \right]. \quad (2)$$

We consider the situation where $K = M$, and thus do not consider user selection. Furthermore, we are mainly interested in the high-spectral efficiency regime, where we can characterize the achievable sum rate as $\kappa \log P/N_0 + O(1)$, and κ is the “system multiplexing gain” or “pre-log factor” of the ergodic sum rate. Hence, it is well-known that using uniform power $\mathcal{P}_k = P/M$ for all $k = 1, \dots, M$, rather than performing optimal water-filling, incurs a loss only in the $O(1)$ term, and we shall restrict to this choice in the rest of this paper.

It is well-known that, under perfect CSIT and CSIR, both the optimal “Dirty-Paper” sum-rate C and the zero-forcing sum-rate R^{ZF} are equal to $M \log P/N_0 + O(1)$. On the contrary, under non-perfect CSIT the rate sum may behave in a radically different way; for example, if there is perfect CSIR and no CSIT when \mathbf{H} has i.i.d. Gaussian entries, the sum rate is equal to $\log P/N_0 + O(1)$ [1]

B. Channel state feedback models

We consider some specific CSIT and CSIR models and derive lower-bounds to the corresponding achievable ergodic rates by analyzing a *naive* beamforming scheme that computes a mismatched ZF beamforming matrix $\hat{\mathbf{V}}$ from the CSIT. In particular, we consider an “analog” CSIT feedback scheme where the transmitter observation at frame time t is given by

$$\{\mathbf{G}(\tau) = \sqrt{\beta P} \mathbf{H}(\tau) + \mathbf{W}(\tau) : \tau = -\infty, \dots, t-d\} \quad (3)$$

where $\{\mathbf{W}(\tau)\}$ is a spatially and spectrally white Gaussian process with elements $\sim \mathcal{CN}(0, N_0)$ and d is the feedback delay. This models the case where the channel coefficients are explicitly transmitted on the reverse link (uplink) using unquantized quadrature-amplitude modulation [3–6]. The power scaling β corresponds to the number of channel uses per channel coefficient, assuming that transmission in the feedback channel has fixed peak power P and that the channel state vector is modulated by a $\beta M \times M$ unitary spreading matrix [3].

A simplifying assumption of this work is that we consider no fading and orthogonal access in the CSIT feedback link, and we assume that the SNR on the feedback channel is equivalent to the un-faded downlink SNR (P/N_0).

A different CSIT feedback approach is based on quantizing the channel vector at each receiver and transmitting back to the base station a packet of B bits, representing the corresponding quantization index. If a random ensemble of quantization schemes is used (referred to as *Random Vector Quantization*, or RVQ), in [7, Theorem 1] it is shown that the gap between ZF with ideal CSI and the naive ZF scheme is given by

$$\Delta R_{\text{quant.}} \leq \log \left(1 + \frac{P}{N_0} 2^{-\frac{B}{M-1}} \right). \quad (4)$$

II. RATE GAP BOUND FOR ANALOG CSIT FEEDBACK

In the case of i.i.d. block fading and no feedback delay, the analog CSIT feedback yields the observation of $\mathbf{G} = \sqrt{\beta P} \mathbf{H} + \mathbf{W}$ at the beginning of every frame. The transmitter computes the MMSE estimate of the channel matrix, $\hat{\mathbf{H}} = \frac{\sqrt{\beta P}}{N_0 + \beta P} \mathbf{G}$. The k -th column $\hat{\mathbf{v}}_k$ of $\hat{\mathbf{V}}$ is a unit vector orthogonal to the subspace $\mathcal{S}_k = \text{span}\{\hat{\mathbf{h}}_j : j \neq k\}$. Notice that we can write $\mathbf{H} = \hat{\mathbf{H}} + \mathbf{E}$, where $\hat{\mathbf{H}}$ and \mathbf{E} are mutually independent and have Gaussian i.i.d. components with mean zero and variance $\frac{\beta P}{N_0 + \beta P}$ and $\sigma_e^2 = (1 + \beta P/N_0)^{-1}$, respectively.

The signal at the k -th receiver is given by

$$y_k = (\mathbf{h}_k^H \hat{\mathbf{v}}_k) u_k + \sum_{j \neq k} (\mathbf{e}_k^H \hat{\mathbf{v}}_j) u_j + z_k \quad (5)$$

We assume that the frame duration is long enough such that some training scheme can be used in the downlink channel. Training allows each receiver to estimate: 1) the useful signal coefficient, $a_k = (\mathbf{h}_k^H \hat{\mathbf{v}}_k)$ and 2) the variance of the interference plus noise $\zeta_k = \sum_{j \neq k} (\mathbf{e}_k^H \hat{\mathbf{v}}_j) u_j + z_k$, given by $\Sigma_k = \mathbb{E} \left[|\zeta_k|^2 | \mathbf{e}_k, \hat{\mathbf{H}} \right] = N_0 + \sum_{j \neq k} |\mathbf{e}_k^H \hat{\mathbf{v}}_j|^2 P/M$. This conditioning is due to the fact that Σ_k is estimated on each frame, and the coefficients $(\mathbf{e}_k^H \hat{\mathbf{v}}_j)$ are constant over each frame and change from frame to frame, following the block i.i.d. fading model. The maximum achievable rate of user k subject to the above assumptions is lowerbounded by assuming a Gaussian input $u_k = u_k^G \sim \mathcal{CN}(0, P/M)$, and by considering the worst-case noise plus interference distribution in every frame. Using stationarity and ergodicity, we have ¹

$$\begin{aligned} R_k &\geq \mathbb{E} \left[\inf_{\zeta_k: \mathbb{E}[|\zeta_k|^2] \leq \Sigma_k} I(u_k^G; y_k | a_k, \Sigma_k) \right] \\ &\stackrel{(a)}{=} \mathbb{E} \left[\log \left(1 + \frac{|a_k|^2 P}{\Sigma_k M} \right) \right] \end{aligned} \quad (6)$$

where (a) follows from [8], noticing that $a_k u_k^G$ and ζ_k are uncorrelated (even after conditioning on a_k, Σ_k).

Next, we shall bound the rate gap incurred by the naive ZF beamforming and analog feedback with respect to the ZF

¹With some abuse of notation, the term in the second line of (6) have the following meaning:

$$\begin{aligned} &\mathbb{E} \left[\inf_{\zeta_k: \mathbb{E}[|\zeta_k|^2] \leq \Sigma_k} I(u_k^G; y_k | a_k, \Sigma_k) \right] \\ &\stackrel{=}{=} \int \inf_{\zeta_k: \mathbb{E}[|\zeta_k|^2] \leq \sigma} I(u_k^G; y_k | a_k, \Sigma_k = \sigma) dF(\sigma) \end{aligned}$$

where $F(\sigma)$ denotes the cdf of Σ_k .

beamforming with ideal CSIT. Denoting by R_k^{ZF} the rate of user k with uniform (across users) and constant (in time) power allocation $\mathcal{P}_k(\mathbf{H}) = P/M$ in (2), we have

$$\begin{aligned} \Delta R_{\text{analog}} &\triangleq R_k^{\text{ZF}} - R_k \\ &\leq \mathbb{E} \left[\log \left(1 + \frac{|\mathbf{h}_k^H \mathbf{v}_k|^2 P}{N_0 M} \right) \right] - \mathbb{E} \left[\log \left(1 + \frac{|a_k|^2 P}{\Sigma_k M} \right) \right] \\ &= \mathbb{E} \left[\log \left(1 + \frac{|\mathbf{h}_k^H \mathbf{v}_k|^2 P}{N_0 M} \right) \right] \\ &\quad - \mathbb{E} \left[\log \left(1 + \frac{(\sum_{j \neq k} |\mathbf{e}_k^H \hat{\mathbf{v}}_j|^2 + |a_k|^2) P}{N_0 M} \right) \right] \\ &\quad + \mathbb{E} \left[\log \left(1 + \sum_{j \neq k} \frac{|\mathbf{e}_k^H \hat{\mathbf{v}}_j|^2 P}{N_0 M} \right) \right] \\ &\stackrel{(a)}{\leq} \mathbb{E} \left[\log \left(1 + \sum_{j \neq k} \frac{|\mathbf{e}_k^H \hat{\mathbf{v}}_j|^2 P}{N_0 M} \right) \right] \\ &\stackrel{(b)}{\leq} \log \left(1 + \frac{P}{N_0 M} \sum_{j \neq k} \mathbb{E}[|\mathbf{e}_k^H \hat{\mathbf{v}}_j|^2] \right) \\ &\stackrel{(c)}{=} \log \left(1 + \frac{\sigma_e^2 P}{N_0} \frac{M-1}{M} \right), \end{aligned} \quad (7)$$

where (a) follows from the fact that $\sum_{j \neq k} |\mathbf{e}_k^H \hat{\mathbf{v}}_j|^2 + |a_k|^2$ stochastically dominates $|\mathbf{h}_k^H \mathbf{v}_k|^2$ since $|a_k|^2$ and $|\mathbf{h}_k^H \mathbf{v}_k|^2$ are identically distributed, (b) follows from Jensen's inequality and the final expression (c) follows by noticing that the $\hat{\mathbf{V}}$ is a deterministic function of $\hat{\mathbf{H}}$ and therefore it is independent of \mathbf{E} . Therefore, we can write $\mathbb{E}[|\mathbf{e}_k^H \hat{\mathbf{v}}_j|^2] = \mathbb{E}[\hat{\mathbf{v}}_j^H \mathbb{E}[\mathbf{e}_k \mathbf{e}_k^H] \hat{\mathbf{v}}_j] = \sigma_e^2 \mathbb{E}[|\hat{\mathbf{v}}_j|^2] = \sigma_e^2$, since $\hat{\mathbf{v}}_j$ has unit norm by construction.

III. COMPARISON WITH QUANTIZED CSIT FEEDBACK

In this section we compare analog and digital feedback under the assumptions of perfect CSIR, no feedback errors, and no feedback delay. Replacing the estimation error variance $\sigma_e^2 = (1 + \beta P/N_0)^{-1}$ in (7) and further upper bounding we obtain:

$$\Delta R_{\text{analog}} \leq \log \left(1 + \frac{1}{\beta} \right). \quad (8)$$

Let us now consider digital feedback over the same channel. The rate gap obtained in [7, Theorem 1] and reported in (4) is further upperbounded by $\log(1 + (P/N_0) \cdot 2^{-\frac{B}{M}})$. Let us assume (very unrealistically) that the digital feedback link can operate error-free and at capacity, i.e., it can reliably transmit $\log(1 + P/N_0)$ bits per symbol. For the same number of feedback channel periods, βM , the number of feedback bits per mobile is $B = \beta M \log_2(1 + P/N_0)$. Replacing this into the rate gap bound, we obtain:

$$\Delta R_{\text{quant.}} \leq \log \left(1 + \frac{P/N_0}{(1 + P/N_0)^\beta} \right). \quad (9)$$

If $\beta = 1$ the quantized and analog feedback achieve essentially the same rate gap of at most 1 b/s/Hz. However, if $\beta > 1$, unlike the analog feedback case, the rate gap of the quantized feedback vanishes for $P/N_0 \rightarrow \infty$. and digital is far superior to analog for $\beta > 1$.

This conclusion finds an appealing interpretation in the context of rate-distortion theory. It is well-known (see [9] and references therein) that analog transmission is an optimal strategy to send a Gaussian source over a Gaussian channel

with minimal end-to-end quadratic distortion. In our case, the source is the Gaussian channel vector \mathbf{h}_k and the noisy channel is the feedback AWGN channel with SNR P/N_0 . Hence, the fact that analog feedback cannot be essentially outperformed for $\beta = 1$ is expected. However, it is also well-known that if the channel rate is larger than the source rate (i.e., less than one Gaussian source symbol arrives per channel symbol, which corresponds to $\beta > 1$ in our case), then analog is strictly suboptimal as compared to separate source and channel coding because the distortion with analog transmission scales as $1/\beta$ whereas it decreases exponentially with β (i.e., along the vector quantizer R-D curve) for digital transmission.

IV. EFFECTS OF IMPERFECT CSIR

We now consider the scenario where each receiver has only a noisy estimate of its channel acquired via downlink training. In order to allow for channel estimation, $\beta_1 M$ shared pilots ($\beta_1 \geq 1$ symbols per antenna) are transmitted. Each receiver estimates its channel on the basis of $\mathbf{Y} = \sqrt{\beta_1 P} \mathbf{H} + \mathbf{Z}$, which yields (after MMSE estimation) Gaussian error with variance $(1 + \beta_1 P/N_0)^{-1}$. Terminals feed back channel information immediately after completion of this training phase. After the transmitter has chosen beamforming vectors on the basis of the channel feedback, an additional round of downlink training is performed to enable coherent detection and allow each terminal to estimate its useful signal coefficient $a_k = \mathbf{h}_k^H \hat{\mathbf{v}}_k$. This can be accomplished in $\beta_2 M$ symbols by transmitting along each of the beamforming vectors for β_2 symbols. If MMSE estimation of a_k is performed, we have $a_k = \hat{a}_k + f_k$ where f_k and \hat{a}_k are independent complex Gaussian's with variance $\sigma_f^2 = \frac{N_0}{N_0 + \beta_2 P}$ and $1 - \sigma_f^2$, respectively.²

Under this set of assumptions, a lower bound to $I(u_k; y_k | \hat{a}_k)$ can be derived using techniques similar to those in [10, 11]. Using this lower bound and some steps similar to those leading to (7), the following upper bound to the rate gap can be reached at:

$$\Delta R \leq \log_2 \left(1 + \frac{P}{N_0 M} (\sigma_f^2 + (M-1) \mathbb{E}[|\mathbf{h}_k^H \hat{\mathbf{v}}_j|^2]) \right),$$

where the multi-user interference term $\mathbb{E}[|\mathbf{h}_k^H \hat{\mathbf{v}}_j|^2]$ depends on the CSIT and thus on the channel feedback (β) as well as the accuracy of the initial training (β_1). We again assume that βM symbols are devoted to channel feedback (per mobile). If analog feedback is used, we get an upper bound of:

$$\Delta R_{\text{analog}} \leq \log \left(1 + \frac{1}{\beta_1} + \frac{1}{M\beta_2} + \frac{1}{\beta} \right) \quad (10)$$

In the case of digital feedback, under the assumption that $B = \beta M \log_2(1 + P/N_0)$ feedback bits per mobile are sent in an error-free manner, we get:

$$\Delta R_{\text{quant.}} \leq \log \left(1 + \frac{1}{\beta_1} + \frac{1}{M\beta_2} + \frac{P/N_0}{(1 + P/N_0)^\beta} \right). \quad (11)$$

Comparing (10) and (11) we come to the same general conclusions as in Section III: if $\beta = 1$ then digital and analog are equivalent, but if $\beta > 1$ digital is superior to analog because the effect of feedback noise vanishes at high SNR for digital but does not do so for analog.

²Note that additional training is required because terminals do not know the channels of other terminals, and thus are not aware of the chosen beamforming vectors.

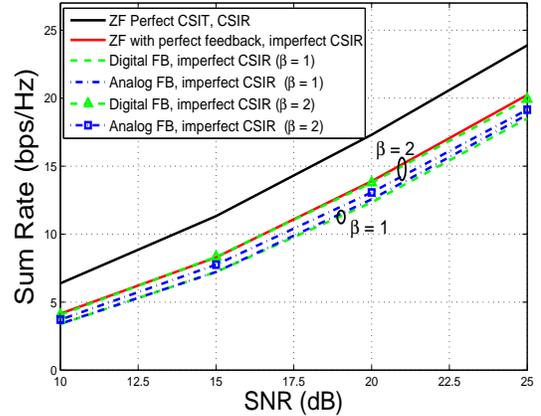


Fig. 1. Quantized vs. Analog Feedback with Imperfect CSIR.

There are, however, some important differences with the perfect CSIR scenario. First note that the imperfect CSIR leads to residual interference that does not vanish with SNR; as a result, the rate gap is not driven to 0 even when $\beta > 1$, assuming β_1 and β_2 are fixed. In addition, when $\beta_1 \approx \beta_2 \approx 1$, imperfect CSIR seems to have a considerably stronger effect than feedback noise, thereby reducing the magnitude of digital feedback's advantage. These effects are both visible in Fig. 1, where analog and digital feedback curves are plotted for $\beta_1 = \beta_2 = 1$ and $\beta = 1$ and $\beta = 2$, along with the throughput of an imperfect CSIR/perfect FB system.

Finally we comment on the tradeoff between downlink training (β_1) and channel feedback (β). Since downlink pilots are shared, training consumes only $\beta_1 M$ channel symbols. Channel feedback, on the other hand, requires βM channel symbols *per mobile*. If the M terminals can simultaneously transmit on the feedback channel, perhaps utilizing the M -antenna receive array at the base as described in [3], then β and β_1 are equivalent in terms of system resources. For the case of analog feedback, from (10) we see that β and β_1 should be chosen equal. On the other hand, if digital feedback is used, it is only necessary to choose $\beta > 1$ (so that the effect of feedback noise vanishes), and the remainder of the resources should be devoted to downlink training, i.e., to β_1 . This is an additional advantage to digital whenever $\beta + \beta_1 > 2$. Note that there is also a tradeoff between β_1 and β_2 , but that the effect of the initial training (β_1) is considerably stronger than the second phase.

V. EFFECTS OF CSIT FEEDBACK ERRORS

We now investigate the impact of removing the optimistic assumption that the quantized feedback channel can operate error-free at capacity. We consider a very simple CSIT feedback scheme that certainly represents a lower bound on the best quantized feedback strategy. The user terminals perform quantization using RVQ and transmit the feedback bits using simple uncoded QAM. No intelligent mapping of the quantization bits onto the QAM symbols is used, and therefore even a single erroneous feedback bit from user k results in CSIT that is completely independent (due to the properties of RVQ) of the actual k -th channel vector. Since uncoded QAM is used, error detection is not possible and the base

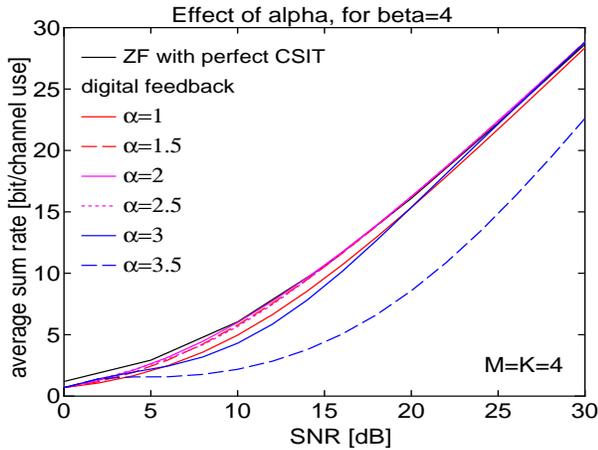


Fig. 2. Quantized feedback with QAM modulation.

station computes beamforming vectors based on the possibly erroneous feedback.

We again use βM symbol periods to transmit the feedback bits. There is a non-trivial tradeoff between quantization and channel errors. In order to maintain a bounded gap, feedback must be scaled at least as $(M-1)\log_2(1+P/N_0) \approx M\log_2 P/N_0$. Therefore, we consider sending $B = \alpha M \log_2 P/N_0$ for $1 \leq \alpha \leq \beta$ bits in βM symbol periods, which corresponds to $\frac{\alpha}{\beta} \log_2(P/N_0)$ bits per QAM symbol.

From [12], using the fact that the QAM constellation size is equal to $L = (P/N_0)^{\frac{\alpha}{\beta}}$, we have the following upper bound to the symbol error probability for QAM modulation:

$$P_s \leq 2 \exp\left(-\frac{3}{2} \left(\frac{P}{N_0}\right)^{1-\alpha/\beta}\right) \quad (12)$$

For $\alpha = \beta$ (which means trying to signal at capacity with uncoded modulation!) P_s does not decrease with SNR and the system performance is very poor. However, for $\alpha/\beta < 1$, which corresponds to transmitting at a constant fraction of capacity, $P_s \rightarrow 0$ as $P/N_0 \rightarrow \infty$. The upper bound on the error probability of the whole quantized vector (transmitted in βM symbols) is given by $P_{e,fb} = 1 - (1 - P_s)^{\beta M}$. A lower bound on the achievable ergodic rate is obtained by assuming that when a feedback error occurs for user k its SINR is zero while if no feedback error occurs its rate is given $R_k^{ZF} - \Delta R_{\text{quant.}}$, that is, the rate of ideal ZF decreased by the (upper bound to) the rate gap. It follows that the ergodic rate of user k is upperbounded by

$$R_k \geq (1 - P_s)^{\beta M} (R_k^{ZF} - \log(1 + (P/N_0)^{1-\alpha})) \quad (13)$$

Choosing $1 < \alpha < \beta$ we achieve both vanishing P_s and vanishing $\Delta R_{\text{quant.}}$ as $P/N_0 \rightarrow \infty$. Thus, even under this very simple CSIT feedback scheme the optimal ZF performance can be eventually approached for sufficiently high SNR.

Fig. 2 shows the ergodic rate achieved by ZF beamforming with quantized CSIT and QAM feedback transmission for $M = K = 4$, independent Rayleigh fading, $\beta = 4$ and different values of α . It is noticed that by proper design of the feedback parameters the performance can be made very close to the ideal CSIT case.

VI. EFFECTS OF CSIT FEEDBACK DELAY

We consider now the case of analog feedback (assuming perfect CSIR) when each entry of \mathbf{H} evolves independently (in the block-fading way described earlier) according to the same complex circularly symmetric Gaussian stationary ergodic random process, denoted by $\{h(t)\}$, with mean zero, variance 1 and power spectral density (Doppler spectrum) denoted by $S_h(\xi)$, $\xi \in [-1/2, 1/2]$.

Because of stationarity, without loss of generality we can focus on $t = 0$. We are interested in the linear MMSE estimation of $h(t)$ from the observation $\{g(\tau) : \tau = -\infty, t-d\}$ where, following the analog feedback model (3), we let $g(\tau) = h(\tau) + w(\tau)$, with $w(\tau)$ i.i.d. $\sim \mathcal{CN}(0, \delta)$ and $\delta = \frac{N_0}{\beta P}$. In particular, we consider the case of 1-step prediction ($d = 1$) and the case of filtering ($d = 0$). From classical Wiener filtering theory [13], we have that the prediction error is given by

$$\epsilon_1(\delta) = \exp\left(\int_{-1/2}^{1/2} \log(\delta + S_h(\xi)) d\xi\right) - \delta \quad (14)$$

and that the filtering MMSE is given by

$$\epsilon_0(\delta) = \frac{\delta \epsilon_1(\delta)}{\delta + \epsilon_1(\delta)}. \quad (15)$$

We shall discuss the rate gap bound (7) letting $\sigma_e^2 = \epsilon_d(N_0/(\beta P))$ for $d = 0, 1$, under different assumptions on the fading process $\{h(t)\}$. We distinguish two cases: Doppler process and regular process. We say that $\{h(t)\}$ is a Doppler process if $S_h(\xi)$ is strictly band-limited in $[-F, F]$, where $F < 1/2$ is the maximum Doppler frequency shift, given by $F = \frac{v f_c}{c} T_f$, where v is the mobile terminal speed (m/s), f_c is the carrier frequency (Hz), c is light speed (m/s) and T_f is the frame duration (s). Furthermore, a Doppler process must satisfy $\int_{-F}^F \log S_h(\xi) d\xi > -\infty$. Following [14], we say that $\{h(t)\}$ is a regular process if $\epsilon_1(0) > 0$. In particular, a process satisfying the Paley-Wiener condition $[\int_{-1/2}^{1/2} \log S_h(\xi) d\xi > -\infty]$ is regular.

A Doppler process satisfying our assumptions has prediction error

$$\epsilon_1(\delta) = \delta^{1-2F} \exp\left(\int_{-F}^F \log(\delta + S_h(\xi)) d\xi\right) - \delta \quad (16)$$

No feedback delay ($d = 0$). In this case

$$P\sigma_e^2 = \frac{N_0}{\beta} \frac{\epsilon_1\left(\frac{N_0}{\beta P}\right)}{\frac{N_0}{\beta P} + \epsilon_1\left(\frac{N_0}{\beta P}\right)} \quad (17)$$

Hence, $\lim_{P \rightarrow \infty} P\sigma_e^2 = \frac{N_0}{\beta}$ for both Doppler and regular processes. For the latter, this is clear from the fact that $\epsilon_1(0) > 0$. For the former, this follows from (16). Applying Jensen's inequality and the fact that $\int S_h(\xi) d\xi = 1$, we arrive at the upper bound

$$\epsilon_1\left(\frac{N_0}{\beta P}\right) \leq \left(\frac{N_0}{\beta P}\right)^{1-2F} \left[\left(\frac{1}{2F} + \left(\frac{N_0}{\beta P}\right)\right)^{2F} - \left(\frac{N_0}{\beta P}\right)^{2F}\right] \quad (18)$$

Using the fact that \log is increasing, we arrive at the lower bound

$$\epsilon_1\left(\frac{N_0}{\beta P}\right) \geq \left(\frac{N_0}{\beta P}\right)^{1-2F} \left[\exp\left(\int_{-F}^F \log S_h(\xi) d\xi\right) - \left(\frac{N_0}{\beta P}\right)^{2F}\right] \quad (19)$$

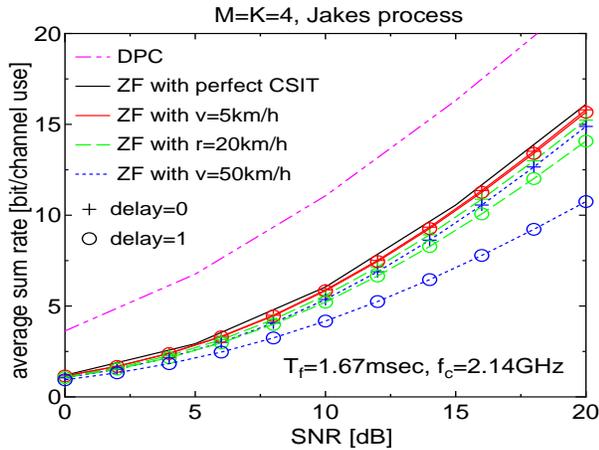


Fig. 3. Rates with feedback delay and Jakes' correlation.

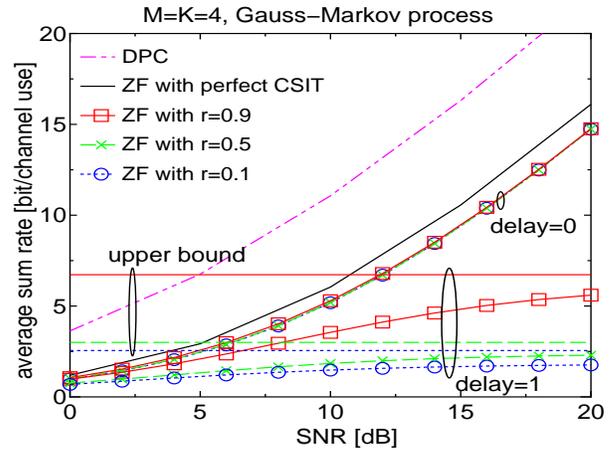


Fig. 4. Rates with Gauss-Markov AR-1 correlation.

These bounds yield that $\epsilon_1(N_0/\beta P) = \kappa P^{-(1-2F)} + O(1/P)$ for some constant κ . Hence, $\epsilon_1 = O(P^{-(1-2F)})$ while $\delta = O(1/P)$, and the limits holds.

We conclude that in the case of no feedback delay the estimation error is essentially dominated by the instantaneous observation and not much improvement can be expected by taking into account the channel memory if analog feedback is used. With quantized feedback the same may not be true because it is possible to exploit memory by feeding back only the innovation process [15]; this is under investigation.

Feedback delay ($d = 1$). In this case, the behavior of Doppler versus regular processes is radically different. For Doppler processes, using (18) and (19), we have that $P\sigma_e^2 = P\epsilon_1(N_0/\beta P) = \kappa P^{2F} + O(1)$. It follows that the achievable rate sum is lowerbounded by

$$\sum_{k=1}^M R_k \geq M(1 - 2F) \log P + O(1) \quad (20)$$

which implies a multiplexing gain of $M(1 - 2F)$.

For regular processes, on the contrary, we have that $P\sigma_e^2 \geq P\epsilon_1(0) = O(P)$. Hence, the rate gap grows like $\log P$ and the achieved multiplexing gain is zero. Furthermore, it can be shown that the following is actually an upper bound to the per-user rate, even when the feedback is noiseless:

$$R_k \leq \log_2 \left(\frac{1}{1-r^2} + (M-1) \right) - \frac{\psi(M)}{\log_e 2} + \frac{1}{\log_e 2} \left(\frac{1}{2M-1} + \frac{1}{2M-2} \right) \quad (21)$$

In conclusions, the most noteworthy result of this analysis is that under common fading models (Doppler processes), the analog feedback scheme achieves a potentially high multiplexing gain even with realistic, noisy and delayed feedback. Notice for example that with mobile speed $v = 50$ km/h, $f_c = 2$ GHz, and frame duration 1 ms, we have $F = 0.0926$. With $M = 4$ antennas we achieve a yet respectable pre-log factor equal to 3.26 instead of 4.³

³It is interesting to notice here the parallel with the results of [14] on the high-SNR capacity of the single-user scalar ergodic stationary fading channel with no CSIR and no CSIT, where it is shown that for a class of *non-regular* processes that includes the Doppler processes defined here, the high-SNR capacity grows like $\mathcal{L} \log P$, where \mathcal{L} is the Lebesgue measure of the set $\{\xi \in [-1/2, 1/2] : S_h(\xi) = 0\}$. In our case, it is clear that $\mathcal{L} = 1 - 2F$.

Figs. 3 and 4 show the achievable ergodic rates for the Jakes' " J_0 " correlation (strictly band-limited) and the Gauss-Markov AR-1 correlation (regular process) for different first-lag correlation values. For the AR-1 process with $d = 1$ the system becomes interference limited. On the contrary, the performance under Jakes' model degrades gracefully as the user mobility (Doppler bandwidth) increases.

REFERENCES

- [1] G. Caire and S. Shamai, "On the achievable throughput of a multiantenna Gaussian broadcast channel," *IEEE Trans. on Inform. Theory*, vol. 49, no. 7, pp. 1691–1706, 2003.
- [2] H. Weingarten, Y. Steinberg, and S. Shamai, "The capacity region of the Gaussian MIMO broadcast channel," *Information Theory, 2004. ISIT 2004. Proceedings. International Symposium on*.
- [3] T.L.Marzetta and B.M.Hochwald, "Fast Transfer of Channel State Information in Wireless Systems," *Submitted to "IEEE Transactions on Signal Processing"*, June 2004.
- [4] M. Kobayashi and G. Caire, "Joint Beamforming and Scheduling for a Multi-Antenna Downlink with Imperfect Transmitter Channel Knowledge," to appear on *IEEE J. Select. Areas Commun.*, 2007.
- [5] T. Thomas, K. Baum, and P. Sartori, "Obtaining channel knowledge for closed-loop multi-stream broadband MIMO-OFDM communications using direct channel feedback," in *IEEE Global Telecommunications Conference, 2005. GLOBECOM '05*, vol. 6, November 2005.
- [6] D.Samardzija and N.Mandayam, "Unquantized and Uncoded Channel State Information Feedback on Wireless Channels," *Proceeding of IEEE WCNC'2005*, pp. New Orleans, LA, USA, March 2005.
- [7] N.Jindal, "MIMO broadcast channels with finite rate feedback," *IEEE Trans. on Inform. Theory*, vol. 52, no. 11, pp. 5045–5059, November 2006.
- [8] B. Hassibi and B. Hochwald, "High-rate codes that are linear in space and time," *IEEE Trans. on Inform. Theory*, vol. 48, no. 7, pp. 1804–1824, 2002.
- [9] M.Gastpar, B.Rimoldi, and M.Vetterli, "To code, or not to code : Lossy source-channel communication revisited," *IEEE Trans. on Inform. Theory*, vol. 49, May 2003.
- [10] M. Médard, "Channel Capacity in Wireless Communications of Perfect and Imperfect Knowledge of the Channel," *IEEE Trans. on Inform. Theory*, vol. 46, no. 3, pp. 933–946, May 2000.
- [11] B. Hassibi and B. Hochwald, "How much training is needed in multiple-antenna wireless links?" *IEEE Trans. on Inform. Theory*, vol. 49, no. 4, pp. 951–963, 2003.
- [12] A. Goldsmith, *Wireless Communications*. Cambridge University Press, 2005.
- [13] H. Poor, *An introduction to signal detection and estimation*. Springer-Verlag New York, Inc. New York, NY, USA, 1994.
- [14] A. Lapidoth, "On the Asymptotic Capacity of Stationary Gaussian Fading Channels," *IEEE Trans. on Inform. Theory*, vol. 51, no. 2, p. 437, 2005.
- [15] J. C. Roh and B. D. Rao, "An efficient feedback method for MIMO systems with slowly time-varying channels," *IEEE WCNC*, March 2004.