Adaptive Coding with Pilot Signals.

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Abstract

We consider adaptive modulation and coding for channels with pilot symbol assisted modulation. Unlike most adaptive transmission schemes, we do not have any channel side information at the sender. The sender adapts not to the state of the physical channel traversed by its transmitted signal, but to the quality of the measurement afforded by pilot symbols at the receiver. We consider the case where no pilot symbols are used, where pilot symbols are used with adaptive reception but without adaptive transmission, and the case where both reception and transmission are done adaptively (while maintaining the average per symbol energy constant). Our results are in terms of capacity for binary signaling. We show that, under appropriate optimizing conditions for the spacing between consecutive pilot symbols, pilot symbol assisted modulation is helpful in terms of capacity. Moreover, adaptation at both the sender and the receiver yields clear advantages over adaptation at the receiver only.

1 Introduction

In order to make best use of wireless resources, many different types of adaptive schemes are employed. Adaptive schemes seek to modify the transmission scheme used by the sender according to the state of the channel seen by the receiver. Generally, such schemes involve feedback, concerning the state of the channel, from the receiver to the sender. In an information-theoretic context, adaptive signaling is used for Markov channels with perfect sender and receiver channel side information [Wol78, GV97] or imperfect channel side information [CS99, MS00, Vis99, Kle01]. Power control is a commonly used type of adaptive transmission. Many practical schemes consider modifying the modulation used in order to combat fading. A common means of adapting transmission is to use different types of modulation [TH99, QC99], for instance different levels of QAM constellations [WH00, WS95, SKM94, TH97, GC97, LJ98, WYH00, THK99], according to the state of the channel and possibly other considerations such as multi-user interference [GA99].

In this paper, we consider a different type of adaptive scheme. While the scheme still adapts the transmission to the channel seen by the receiver, the transmitter does not use feedback to determine its policy. Instead, the transmitter takes into account the time-varying quality of the channel measurement available at the receiver in order to modify its transmission policy. Thus, the transmitter adapts to the quality of the channel measurement, rather than to the quality of the channel (in terms of carrier to noise ratio or other metric.)

Our model is the following. We consider a single sender and receiver, connected by a time-varying Rayleigh fading channel. The Rayleigh fading channel is modeled as a Gauss-Markov process. The sender transmits coded and modulated data. The sender has no information
regarding the state of the channel and, therefore, does not adapt its transmission scheme in response to fades. Moreover, at regular intervals, the sender transmits a constant and known pilot symbol, whose purpose is to enable measurement of the channel at the receiver. The sender thus transmits coded data, periodically interrupted by pilot symbols. Such a scheme is often referred to as pilot symbol assisted modulation [TAG99, KKI+97, Cav91, S93, TH95]. Pilot symbols are commonly used to improve detection [HK96, MB84] and decoding [WC99]. The pilot symbols have energy equal to the average energy constraint. The only channel estimate available at the receiver comes from the pilot symbols, hence there is no data-directed estimation of the channel. The channel estimate at a time sample \( k \) depends on the received pilot symbols through its position with respect to the pilot symbols. Given the estimate of the channel obtained through the pilot symbols, the channel is no longer Rayleigh but Ricean with a known specular component. We consider binary signaling, since such signaling performs well at low SNRs and achieves capacity for low SNRs for Rayleigh channels. We also consider a discrete time, sampled system.

Our purpose is to assess the benefits of performing adaptive transmission which takes into account the quality of the estimate obtained at the receiver. We seek to maximize mutual information. The maximization of mutual information subsumes the optimization of both the modulation and the coding. We consider three cases. First, we consider the case where no pilot symbols are transmitted and the channel is not estimated at the receiver. The channel at the receiver is then a Rayleigh channel. Since we do not perform any data-directed estimation, the capacity of the system is the same as if all the channel samples were mutually independent (by averaging over very long times). The capacity of such channels was found in [AFTS97]. For low to moderate SNRs, binary signaling is optimal. The purpose of considering the Rayleigh channel with no pilot symbols is to establish a basis of comparison for the other two cases, which do employ pilot symbols. Indeed, the first question we pose is whether it is preferable to forego pilot symbols and channel estimation altogether and devote to coded data the time allocated to pilot symbols.

Second, we consider the case where we use pilot symbols to aid in the detection and decoding at the receiver but do not modify the distribution of the transmitted signal. We term this scheme the non-adaptive scheme with pilot tones, since the scheme is adaptive at the receiver but not at the sender. The sender uses the distribution that is optimal for transmission when the channel is block-faded. In effect, the sender behaves as though the channel estimate did not vary between pilot symbols. Since we look at mutual information, we may consider this case to be an upper bound to the case where neither the distribution nor the coding is modified according to the distance between the transmitted signal and the pilot symbols. Alternatively, we may view our results as representing capacity when the distribution of the transmitted signal is fixed.

Third, we consider the case where we use pilot symbols and adaptive transmission, but maintain the average per symbol power constant in the coded data. We consider the case where the estimation of the channel is causally performed using the last pilot symbol, and the case where the estimation is non-causally based on the last transmitted pilot along with the pilot transmitted next. At each time sample, the sender changes his transmission policy according to the distance to pilot symbols. We compare the performance of the second and third cases in order to establish the benefit of adaptive transmission.

In Section 2, we present our channel model and the principles of non-causal and causal estimation. In Section 3, we discuss non-adaptive and adaptive schemes used. In Section 4, we present our numerical results. These results allow us to optimize numerically the spacing between pilot symbols for adaptive and non-adaptive schemes, since such an optimization cannot be obtained in closed form. Using the optimized spacing between pilot symbols for
the different schemes, we can evaluate the benefit of adaptive schemes. Finally, conclusions and directions for future work are presented in Section 5.

2 Channel Model

We consider the following discrete-time model for the Rayleigh fading channel

\[ Y = AX + N, \]

where \( X \) is the channel input, \( Y \) the output, and \( A \) and \( N \) are independent complex circular Gaussian random variables with mean zero and variance \( \sigma_A^2 \) and \( \sigma_N^2 \) respectively. Equivalently, the amplitude of the fading coefficient \( A \) is Rayleigh distributed and its phase is uniform. The input \( X \) is average power limited: \( E ||X||^2 \leq P. \)

We assume that the fading process is a first-order Gauss-Markov process

\[ A_k = \alpha A_{k-1} + Z_{k-1}, \]

where \( k \) denotes the time index. Depending on the scale of the time variations of the channel, \( \alpha \) may vary between one (corresponding to no channel variations) and zero (where the fading is fast, changing independently from one symbol time to the next). A similar channel model was introduced in [M00], where the relation between this channel and the coherence time is established.

For a coherence time of \( T_c \) and transmission over a bandwidth of \( W \), \( \alpha \) is determined by \( \alpha^{T_c,W} = \phi \) where \( \phi \) is the level of decorrelation we deem necessary in our definition of coherence time. In the literature, the correlation coefficient is taken to vary from 0.9 ([CL75]) to 0.37 ([BN63]) for a time separation of \( T_c \). For bandwidths in the 10 kHz range, and Doppler spreads of the order of 100 Hz, \( \alpha \) will typically range between 0.9 and 0.99. For instance, for a Doppler spread of 100 Hz, \( W = 10^4 \) Hz, \( \phi = 0.1 \), we have \( \alpha = 0.977 \). For a Doppler spread of 200 Hz, \( \alpha = 0.955 \). For a Doppler spread of 50 Hz, \( \alpha = 0.989 \).

When no pilot signals are used and the correlation between fading coefficients is ignored, the channel behaves as a memoryless Rayleigh fading channel. The capacity of this channel was studied in [AFTS97] and the optimizing input distribution was found to be discrete with a finite number of mass points one of them located at the origin. Furthermore, a binary distribution was found to be optimal at low and moderate values of the SNR motivating the use of binary input strategies in this paper.

For an adaptively coded system, we consider the case where a pilot signal of power \( P \) is transmitted once at the beginning of each interval of length \( T \), enabling the receiver to estimate the fading coefficients governing the channel statistics. The resulting channel outputs are

\[ Y_{lT} = A_{lT} \sqrt{P} + N_{lT}, \quad l \in \mathbb{N}. \]

Based on the observation of \( \{y_{lT}\}_l \), estimates of the fading coefficients \( \{A_k\}_{lT+1}^{lT-1} \) are obtained and at each time step \( k \), using its estimate the receiver experiences a Ricean channel where the fading coefficient has mean \( \hat{A}_k \) and variance \( v_k \), where \( \hat{A}_k \) is the estimate of the value of \( A_k \) and \( v_k \) is the variance of the estimation error:

\[
p_{Y_k|X_k}(y_k|x_k) = \frac{1}{\pi(v_k|x_k|^2 + \sigma_N^2)} \exp\left\{ \frac{-|y_k - \hat{A}_k x_k|^2}{v_k|x_k|^2 + \sigma_N^2} \right\}, \quad lT < k < (l+1)T.
\]

\(^1\)Since the fading is Rayleigh distributed, it is sufficient to consider only real positive values for the parameter \( \alpha \).
Therefore\(^2\), for a given received sequence \(\{y_{IT}\}\) the mutual information \(I_k(X_k; Y_k|\{Y_{IT}\} = \{y_{IT}\})\) at time \(k \in \{lT + 1, \ldots, (l + 1)T - 1\}\) is given by

\[
I_k(X_k; Y_k|\{Y_{IT}\} = \{y_{IT}\}) = p_k(1) \int p_{Y_k|X_k}(y|x_k(1)) \ln \frac{p_{Y_k|X_k}(y|x_k(1))}{p_{Y_k}(y)} dy \\
+ (1 - p_k(1)) \int p_{Y_k|X_k}(y|x_k(2)) \ln \frac{p_{Y_k|X_k}(y|x_k(2))}{p_{Y_k}(y)} dy,
\]

where the probability mass function of the binary input \(\{x_k(1), x_k(2)\}\) is \(\{p_k(1), (1 - p_k(1))\}\).

Furthermore, since we are eliminating the possibility of data-directed estimation, the achievable rates\(^3\) are hence given by

\[
E \left[ \frac{1}{T} \sum_{k=lT+1}^{(l+1)T-1} I_k(X_k; Y_k|\{Y_{IT}\} = \{y_{IT}\}) \right] = \frac{1}{T} \sum_{k=lT+1}^{(l+1)T-1} E \left[ I_k(X_k; Y_k|\{Y_{IT}\} = \{y_{IT}\}) \right], \quad (1)
\]

where the expectation is over the (correlated) random variables \(\{Y_{IT}\}\) which are circular Gaussian distributed with mean zero and variance \(\sigma_A^2 P + \sigma_N^2\).

Overall we consider two estimation procedures:

1. The receiver performs a causal estimation of the channel parameters based on the most recently sent pilot tone. For each interval, based on the observation of \(y_{IT}\), estimates of the fading coefficients \(\{\hat{A}_k\}_{lT+1}^{(l+1)T-1}\) are obtained by a standard Bayesian least square estimation procedure yielding

   \[
   \hat{A}_k = \frac{\sqrt{P\sigma_A^2}}{P\sigma_A^2 + \sigma_N^2}\alpha^{(k-lT)}y_{IT} \quad \text{(2)}
   \]

   \[
   v_k = \sigma_A^2 - \frac{P\sigma_A^4}{P\sigma_A^2 + \sigma_N^2}\alpha^{2(k-lT)}, \quad lT < k < (l+1)T. \quad (3)
   \]

2. The estimation is non-causal and based on the most recently sent pilot tone along with the one transmitted next, i.e. \(\hat{A}_k\) is based on the values of \(y_{IT}\) and \(y_{(l+1)T}\) whenever \(lT < k < (l+1)T\). In this case

   \[
   \hat{A}_k = \frac{\sqrt{P\sigma_A^2}}{(P\sigma_A^2 + \sigma_N^2)^2 - P^2\alpha^{2T}} \left[ \alpha^{(k-lT)}(P\sigma_A^2 + \sigma_N^2) - P\alpha^{2T-(k-lT)} \right] y_{IT} \\
   + \left[ \alpha^{T-(k-lT)}(P\sigma_A^2 + \sigma_N^2) - P\alpha^{T+(k-lT)} \right] y_{(l+1)T} \quad (4)
   \]

   \[
   v_k = \sigma_A^2 - \frac{P\sigma_A^4}{(P\sigma_A^2 + \sigma_N^2)^2 - P^2\alpha^{2T}} \left[ (P\sigma_A^2 + \sigma_N^2)\alpha^{2(k-lT)} - \alpha^{2[(T-(k-lT)]} \right] - 2P\alpha^{2T} \quad (5)
   \]

3 **Non-Adaptive and Adaptive Coding**

First we consider the scheme where the transmitter does not adapt its transmission strategy to the statistics of the channel estimates used at the receiver. More precisely, we compute the achievable rates when the transmitter is using a single fixed input distribution at all times. This of course will allow us to quantify the performance of a system when the transmitter

\(\text{Under the assumption of binary signaling.}\)

\(\text{Note that the mutual information at times } k = lT \text{ is zero.}\)
does not adapt its coding strategy and uses instead one fixed codebook independently of the time index.

We look mainly at the case where the transmitter considers the channel to be block-faded where, the fading coefficient is assumed constant over intervals of length \( T \) and changing independently from one interval to the next. Consequently, we find first the optimal input distribution for the block-faded system, and then compute the average mutual information under this distribution for different values of \( \alpha \). While when \( \alpha = 1 \) the model based on the first estimation procedure would correspond to block-fading, as \( \alpha \) decreases we expect the performance of this non-adaptive system to deteriorate, and we quantifying this performance loss.

Here we consider the scheme where, without any channel state information, the transmitter takes into consideration the statistics of the channel estimates at the receiver. It adapts accordingly to modulation and coding to maximize the rates that can be reliably transmitted over the channel. While no optimal power allocation is performed here (a constant amount of power is used instead), at each time step the transmitter uses a “good” codebook achieving the highest mutual information of the Ricean channel the receiver sees.

Equivalently, one can think of the problem as that of finding the best input strategy that maximizes the expected mutual information \( \mathbb{E} \left[ I_k \left( X_k; Y_k \mid \{ Y_{IT} \} = \{ y_{IT} \} \right) \right] \) for each time step between \( IT + 1 \) and \( (l + 1)T - 1 \). For these computations, we considered the two estimation methods described in section 2, i.e. for the pair \( \{ \hat{A}_k \}, \{ v_k \} \) given by either equations (2) and (3), or equations (4) and (5).

We use standard matlab tools to optimize, for each time period \( k \in \{ 1, \ldots, T - 1 \} \), the expected mutual information over the input probability distribution. The corresponding optimal distribution yields of course the highest achievable rates depending on how far the transmission is occurring with respect to the pilot signals. Sending pilot tones frequently clearly reduced the rates as a significant portion of the time and power is used to estimate the channel and no information is conveyed from the transmitter to the receiver. On the other hand, when the pilots are used very infrequently, the channel estimates at the receiver are of poor quality and the information rates are low. One of the questions that we will try to answer in the following section is: what is the optimal value of \( T \) that would yield the best compromise?

# 4 Numerical Results

In figures 1 and 2 we have drawn the achievable rates for different values of the SNR, for \( \alpha = 0.95 \) and 0.99 respectively. In these plots the value of \( T \) varies between 2 and 30. The dashed lines in these plots represent the non-adaptive scheme, while the solid lines are those of the adaptively coded system, when the causal estimation method is used.

As is apparent from figure 1, in the considered ranges of the SNR, there is no benefit gained from using pilot symbol-assisted modulation for \( \alpha = 0.95 \). Indeed, for these low correlation models, not sending a pilot tone and coding instead to a memoryless Rayleigh fading channel outperforms the adaptive and non-adaptive causal scheme we have described. However, for \( \alpha = 0.99 \), figure 2 shows not only that an improvement is possible, but also that there is a trade-off between infrequent transmission of pilots and the channel estimate quality. Indeed, for every value of the SNR there is an optimal value for \( T \) where the rates are maximized, and these optimal values are apparent in the figure. This holds for both the adaptive and non-adaptive schemes and the relative improvement that can be achieved with the adaptive scheme considered ranges between 4.2% and 7.5% depending on the SNR. Moreover, as expected, the
adaptive scheme outperforms the non-adaptive one by 1.5% to 3.2% when the SNR ranges between 0 and 5dB. This gain becomes more significant for smaller values of $\alpha$ as it can be seen in figure 1. For illustrative purposes, in figures 3, 4 and 5 we have drawn the achievable rates for given values of the SNR, as a function of $\alpha$ when $T$ varies between 2 and 30.

In figures 6 and 7, we have drawn the achievable rates for the same values of the SNR, $\alpha$ and $T$, when the non-causal estimation method is used. Figure 8 shows the achievable rates for 3dB SNR, for different values of $\alpha$. As expected, the non-causal scheme outperforms the causal one. We note that the optimal values of $T$ appear to be smaller. Initially, one would expect the opposite since the quality of the channel estimates for large values of $T$ is significantly better especially in the mid-interval range. This suggests that choosing a large value of $T$ is beneficial, especially that it reduces the fraction of time when the channel is idle. However, this is not the case. Intuitively speaking, the gain one gets from the two-pilots based channel estimation is considerable making the more frequent transmission of pilots here significantly better.

5 Conclusions and Future Work

We have investigated the use of adaptive and non-adaptive sender strategies for time-varying channels with pilot symbol assisted modulation and no feedback from the receiver to the
sender. We have shown that, depending on the rate of change of the channel, adaptive sender strategies, properly optimized for spacing between consecutive pilot tones, can improve channel capacity.

Our adaptive strategy maintains average power constant on a symbol-per-symbol basis. A natural extension is to investigate to what extent adaptive power modulation can further improve capacity. Establishing an optimal power allocation is difficult, however, because we do not have a closed form for capacity. Preliminary results on some heuristics methods, akin to waterfilling, which attempt to maintain the interference constant, do not generally lead to improvements over constant power allocations.

References


Figure 5: Best achievable rates at 5dB function of $T$ when $\alpha = 0.9, 0.95, 0.97$ and 0.99 for adaptive and non-adaptive coding for causal estimation.

Figure 6: Best achievable rates function of $T$ and the SNR for $\alpha = 0.95$ for adaptive and non-adaptive coding for non-causal estimation.


Figure 7: Best achievable rates function of $T$ and the SNR for $\alpha = 0.99$ for adaptive and non-adaptive coding for non-causal estimation

Figure 8: Best achievable rates at 3dB function of $T$ when $\alpha = 0.9, 0.95, 0.97$ and 0.99


