

Superposition Turbo TCM for Multi-Rate Broadcast

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Abstract—Bergmans and Cover identified the capacity region of the Gaussian degraded broadcast channel, where different receivers observe the transmitted signal with different signal to noise ratios. This paper presents a superposition turbo coding scheme that performs within 1 dB of the capacity region boundary of the degraded broadcast channel at BER of 10^{-5} . Performance is consistent over the entire useful range of the power allocation parameter α .

Coding for the degraded broadcast channel is equivalent to coding for unequal error protection. Adjusting α changes the transmitter constellation of our encoder, and changes the degree to which error protection is unequal. When α is selected to provide equal error protection, the code is essentially a multilevel code. This multilevel code performs as well as single level turbo trellis-coded-modulation schemes with the advantage of the potential for flexible unequal error protection as α is varied.

Keywords Turbo Coding, Superposition, Time Division, Degraded Broadcast Channel, Trellis Coded Modulation.

I. INTRODUCTION

In many applications, different receivers could experience different signal-to-noise ratios (SNRs). For example, in satellite television broadcasting, receivers in rain-fades have very low SNRs, while other receivers under clear sky have high SNRs.

With progressive source coding, the data from the video source encoder is not uniformly important. It is desirable that the most important data be recovered even by receivers with poor receiving conditions (i.e. low SNRs). An unequal error protection (UEP) code provides more protection to more important data, permitting it to be received on channels with lower SNR.

Coding for UEP is identical to coding for the degraded broadcast channel. Consider the Gaussian broadcast channel with one sender transmitting X and two receivers observing Y_1 and Y_2 respectively with

$$Y_1 = X + Z_1 \quad (1)$$

$$Y_2 = X + Z_2 \quad (2)$$

where $Z_1 \sim \mathcal{N}(0, N_1)$ and $Z_2 \sim \mathcal{N}(0, N_2)$ with $N_2 > N_1$. This broadcast channel can be re-characterized as the statistically equivalent “degraded” broadcast channel.

$$Y_1 = X + Z_1 \quad (3)$$

$$Y_2 = Y_1 + Z_2' \quad (4)$$

where $Z_2' \sim \mathcal{N}(0, N_2 - N_1)$.

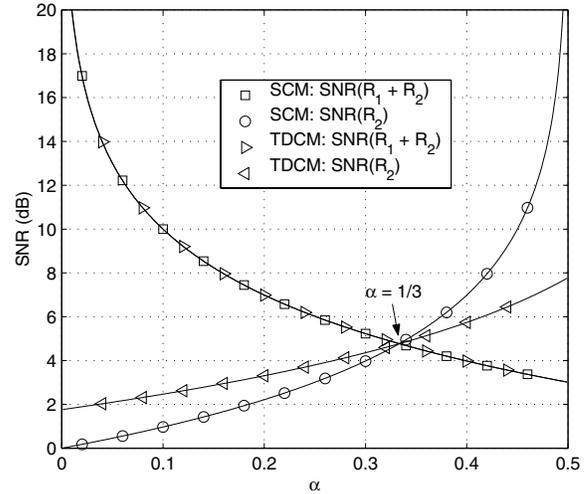


Fig. 1. Comparison of Superposition Coded Modulation (SCM) and Time-Division Coded Modulation (TDCM).

The capacity region for the Gaussian degraded broadcast channel, with signal power constraint P , is given by

$$R_1 < C\left(\frac{\alpha P}{N_1}\right) \quad (5)$$

$$R_2 < C\left(\frac{(1 - \alpha)P}{\alpha P + N_2}\right) \quad \text{for } 0 \leq \alpha \leq 1 \quad (6)$$

where $C(x) = \log(1 + x)$ denotes the capacity in bits per channel use of a two dimensional memoryless Gaussian channel with SNR x . This region is theoretically achieved by the superposition coding scheme given by Cover in [1]. Bergmans [2], [3] proved the converse. See also [4] for an excellent survey of broadcast channel information theory.

Figure 1 illustrates (5-6) for $R_1 = R_2 = 1$ by plotting the required SNRs for R_2 (circles) and $R_1 + R_2$ (squares) as a function of α . Since (5) assumes that the R_2 data has already been decoded, the plot is meaningful only when the $SNR(R_1 + R_2)$ curve lies above the $SNR(R_2)$ curve, i.e. for $\alpha < 1/3$.

Figure 1 shows how α controls the UEP tradeoff. As α decreases, the more important R_2 data can be received at lower and lower SNRs. However, this extra reliability of the R_2 data comes at the cost of decreased reliability for the less important R_1 data which requires a higher and higher SNRs as α decreases.

Figure 1 also shows the performance of an optimal time-division coded modulation (TDCM) scheme designed to match

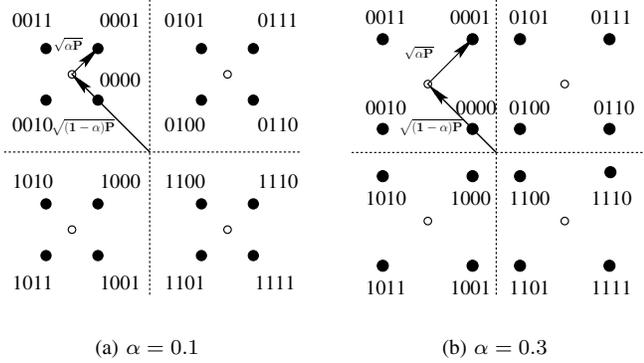


Fig. 2. Example for constellations in simulation for $\alpha = 0.1$, $\alpha = 0.3$

the $R_1 + R_2$ performance of superposition coded modulation (SCM). As shown in the figure, TDCM is theoretically inferior to SCM. However, Wei [5] considered trellis-coded-modulation (TCM) for UEP using both TDCM and SCM approaches, and found TDCM to be superior. Gadkari and Rose [6] explained this behavior by showing that TDCM outperforms SCM for some UEP regions when the channel code performance is sufficiently worse than the capacity-achieving performance assumed in [1], [2], [3]. Recently, Wang and Orchard [7] carefully designed a TCM-based SCM scheme which incorporated shaping. This scheme does outperform TDCM in a UEP region of interest.

In this paper, we use turbo codes to design an SCM. Since turbo codes closely approximate capacity achieving performance, it is not surprising that our SCM closely approach the theoretical SCM curve in Figure 1. In all of our examples, the turbo coded SCM architecture at BER of 10^{-5} performs within 1 dB of the capacity region boundary of the degraded broadcast channel over the entire useful range of the power allocation parameter α .

For our SCM architecture, we use the symbol-interleaved parallel-concatenated turbo TCM (PCTCM) structure of Fragouli and Wesel [8], [9]. The bit-interleaved turbo TCM scheme of Benedetto, Divsalar, and Montorsi [10] is a possible alternative, but bit-interleaving further complicates the initial and subsequent soft information computations. We also implemented SCM using the symbol-interleaved turbo TCM structure of Robertson and Wörz [11], and found its performance to be quite similar to the results presented in this paper, about 0.1 dB worse than SCM based on the Fragouli and Wesel symbol-interleaved architecture.

Section II discusses the power allocation parameter α in detail, and shows that when α takes a certain value, SCM reduces to multi-level coding as discussed in [12]. Section III presents the detail of our SCM turbo TCM architecture. Section IV presents simulation results for two-rate SCMs using 8-point circle constellation and 16-point square constellation as well as a three-rate example using 64-QAM. Section V concludes the paper.

II. RECONFIGURABILITY USING α

The α in (5-6) controls the ratio of power allocation to the two data streams. For practical constellations, it decides the Euclidean distances between the constellation points. For a fixed-rate two-rate structure with $R_1 = R_2 = 1$, α can take a value from 0 to 0.5 according to (5-6). However, as shown in Figure 1, with two-stage decoding, only α between 0 and $1/3$ makes sense.

When α equals $1/3$, the two-rate structure becomes a single rate structure theoretically. It is essentially a multilevel code (MLC), and our two-stage decoding is actually the two-level MLC decoding [12]. By choosing α within $(0, 1/3)$, we configure the system to support two rates with two corresponding operating SNRs.

For standard 16-QAM, the α in (5-6) can be easily determined as 0.2. Solving (5-6) for the maximum N_1 and N_2 with $R_1 = R_2 = 1$ and $\alpha = 0.2$ yields $N_1 = 0.2P$ and $N_2 = 0.6P$ respectively. The resulting error free SNRs are 7.0 dB and 2.2 dB as shown in Figure 1. Additionally, incorporating the constraint to 16-QAM signaling [13] reduces the tolerable noise to $N_1 = 0.1917P$ and $N_2 = 0.5651P$ respectively. As a result, the error free SNRs are 7.2 dB and 2.5 dB for standard 16-QAM ($\alpha = 0.2$).

Figure 2 shows the 16-point constellations resulting from $\alpha = 0.1$ and $\alpha = 0.3$. Figure 2 also illustrates the edge-profile optimal constellation labeling [14] of the 16-point square constellations used in simulation. It can be seen that the 2 MSB bits of four constellation labelings in the same quadrant are identical.

Figure 2 also shows how α allocates power. For the 16-point square constellation in Figure 2, the 16-point square constellation can be considered as one QPSK constellation added to another QPSK constellation. Two bits of rate- R_2 data are modulated as a QPSK signal which can be mapped to one of four “empty dots” with amplitude of $\sqrt{(1-\alpha)P}$. Then, two bits of rate- R_1 data are modulated as another QPSK signal which can be represented by a solid dot with amplitude of $\sqrt{\alpha P}$, using one of four “empty dots” as its origin.

The following theorem shows that the MLC coding scheme is the special case of multi-rate degraded broadcast channel when all receivers have the same error free SNR. Hence, it is likely we can tune a good MLC code into a good degraded broadcast channel code, or vice versa.

Theorem 1: Consider the Gaussian broadcast channel with one sender and k receivers, $k \geq 1$, we have $Y_i = X + Z_i$, where $Z_i \sim \mathcal{N}(0, N_i)$

$$R_1 < C\left(\frac{\alpha_1 P}{N_1}\right), \dots, \quad (7)$$

$$R_i < C\left(\frac{\alpha_i P}{\sum_{j=1}^{i-1} \alpha_j P + N_i}\right), \dots \quad (8)$$

where $1 \leq i \leq k$, and $\sum_{i=1}^k \alpha_i = 1$.

if $N_1 = N_2 = \dots = N$, then, the rate region reduces to

$$\sum_{i=1}^k R_i < C\left(\frac{P}{N}\right). \quad (9)$$

Proof: Sum R_1, \dots, R_N with $N_i = N$. \square

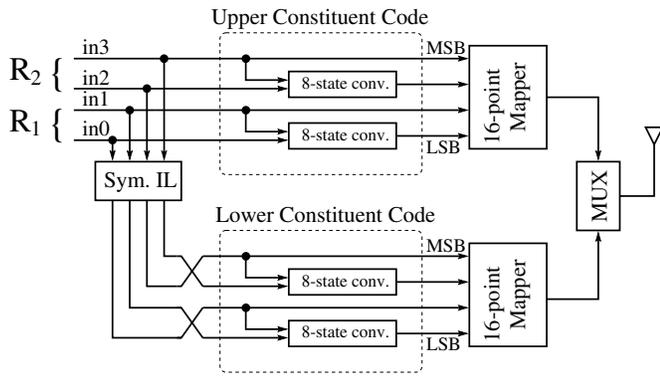


Fig. 3. PCTCM Superposition Turbo TCM Encoder for Two Rates and a 16-point Square Constellation

This theorem explains why our multi-rate superposition turbo TCM schemes, when configured for single rate, can perform as well as the typical single rate turbo TCM in simulation. If $N_i = N$ and the rates have been fixed to satisfy (9), (7-8) can always be solved for $\alpha_1, \dots, \alpha_k$. So a good single rate code may be obtained from a good multi-rate code as a special case.

III. THE SUPERPOSITION TURBO TCM

A. Encoder

Figure 3 shows the structure of a superposition turbo encoder based on the structure of [8], [9] for the family of 16-point square constellations, for which examples are illustrated in Figure 2. It is a parallel concatenated TCM with constituent encoder of rate-4/4. In fact, the constituent code consists of two identical recursive convolutional codes of rate-2/2. The rate-2/2 recursive convolutional encoder may be considered as a rate-2/3 recursive convolutional encoder with one systematic bit punctured.

The upper rate-4/4 constituent encoder has as systematic outputs the two most significant input bits (MSBs) while the lower rate-4/4 encoder has as systematic outputs the two least significant input bits (LSBs). Thus the systematic bits are evenly divided between the constituent encoders, and the overall turbo code cannot be catastrophic.

The rate-2/2, 8-state recursive convolutional encoder within the constituent encoders is obtained through exhaustive computer search, and optimized for normalized d_{s2}^E , where d_{s2}^E represents effective free distance. The superscript E refers to the output squared Euclidean distance, and the subscript $s2$ denotes the symbol-wise input weight is two.

According to Fragouli [9], an upper bound on effective free distance for convolutional encoder with multiple “disconnected” shift registers is constrained by the minimum length shift register. Thus, the number of memory elements of two recursive convolutional encoders in the same constituent code should be the same. We found a good rate-4/4 constituent code with identical rate-2/2 recursive convolutional encoders for the two input MSBs and for the two input LSBs. The 8-state recursive convolutional code we found by optimizing

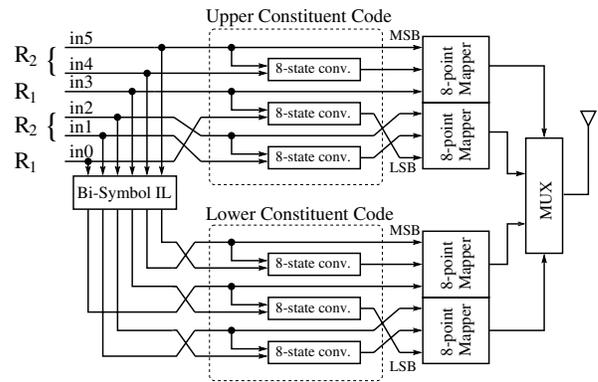


Fig. 4. PCTCM Superposition Turbo TCM Encoder for Two Rates and a 8-point Circular Constellation

normalized d_{s2}^E is

$$A = \begin{bmatrix} 0 & 1 & 0 \\ 0 & 0 & 1 \\ 1 & 1 & 0 \end{bmatrix}, \quad B = \begin{bmatrix} 0 & 0 & 1 \\ 0 & 1 & 1 \end{bmatrix}, \quad (10)$$

$$C = \begin{bmatrix} 1 & 0 \\ 0 & 0 \\ 1 & 0 \end{bmatrix}, \quad D = \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix}. \quad (11)$$

in the encoder state-space description:

$$[s_1 \ s_2 \ s_3]_{j+1} = [s_1 \ s_2 \ s_3]_j \cdot A + [u_1 \ u_2]_j \cdot B \quad (12)$$

$$[y_1 \ y_2]_j = [s_1 \ s_2 \ s_3]_j \cdot C + [u_1 \ u_2]_j \cdot D \quad (13)$$

We applied extended spread symbol interleavers [8] to the above constituent code. The extended spread symbol interleaver is a semi-random interleaver due to the random selection without replacement of N integers from 1 to N under certain constraints.

Figure 4 shows the superposition turbo TCM encoder for 8-point circular constellation. Two levels of superposition are used for this structure. A BPSK data stream (i.e. the LSB) is added on top of a QPSK data stream. Since we use rate-1/2 code for the LSB, we encode two 8-PSK symbols at the same time. The rate-2/2 8-state recursive convolutional code described by (10-13) is also used for each sub-code of the constituent code for the 8-point circular constellation encoder.

Our superposition turbo TCM encoder for 64-point square constellation is the direct extension of the encoder in Figure 3. Three levels of superposition are used for this structure. Three QPSK data streams are summed together to generate the transmitted data stream. Again, the code described by (10-13) is used for each sub-code of the constituent code.

B. Decoder

We use two-stage decoding for the two-rate encoder of Figure 3. Figure 5 shows the detailed data flow of the decoder for implementation. It is similar to the “onion peeling” algorithm in [15].

The first decoding stage decodes the two input MSBs. It uses an 8-state-trellis, and calculates the metric by summing up the squared Euclidean distances between the received symbol and all four constellation points of the same quadrant.

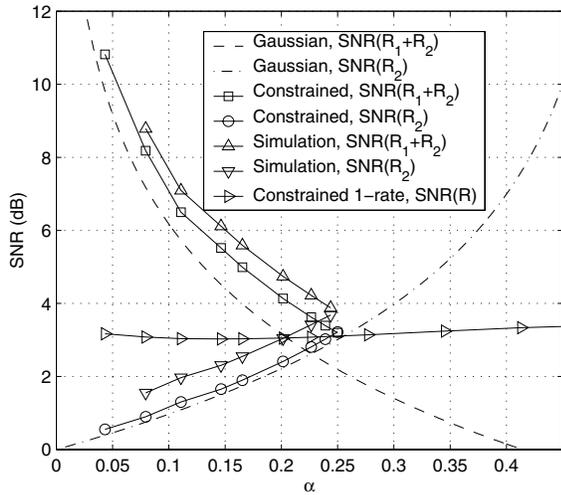


Fig. 7. Error free SNR for high rate and low rate receivers vs. α ; interleaver length = 8,192; number of iter. = 15; $BER = 10^{-5}$; $R_1 = R_2 = 1$ (bit/s/Hz); 8-point circular constellation

due to an 8-point circular signaling constellation, the practical range of α is around 0 to 0.25. For the various α 's from 0.0794 to 0.25 that we simulated, the SNR is within 0.7 dB of constrained capacity at BER of 10^{-5} . The symbol interleaver has design parameters ($S = 54, T = 6, X = 1$) and length 8,192.

C. Three-rate Superposition with 64-QAM

Figure 8 shows BER vs. SNR for three SNR ranges with $R_1 = R_2 = R_3 = 1$, using 64-QAM ($\alpha_1 = 1/21, \alpha_2 = 4/21$), where α_1 and α_2 are the power allocation parameters in (7-8). The constrained capacities are 3.1 dB, 8.7 dB and 13.4 dB for low, middle and high SNR ranges respectively. The interleavers with length of 8,192, 4,096 and 2,048 have constraint parameters ($S = 54, T = 6, X = 1$), ($S = 37, T = 5, X = 1$), and ($S = 26, T = 4, X = 0$) respectively. For 64-QAM at BER of 10^{-5} , the performance is within 0.6, 0.8 and 0.9 dB of constrained capacity at high, middle and low SNR respectively.

In general, for our simulations, the performance at high SNR is closest to the theoretical limit, and the performance at middle SNR is closer than that at low SNR .

V. CONCLUSION

This paper illustrates that, with turbo codes, superposition coding with multi-stage decoding performs quite well, always within 1 dB of the achievable region boundary. In the context of the work of Cover and Bergmans [2], [4], and the well-known power of turbo codes, this is not that surprising.

Multilevel coding is a special case of multi-rate superposition coding with α selected so that all rates are decoded at the same SNR . The proposed superposition schemes offer the same complexity and performance as that of the typical turbo TCM for this power allocation parameter α , but provide the flexibility of controlling unequal error protection through the choice of α .

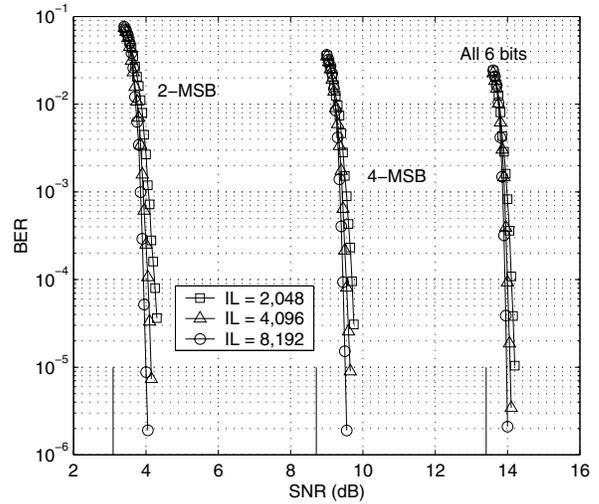


Fig. 8. Three-rate Structure with 64-QAM, Constrained capacities = 3.0913 (dB), 8.7095 (dB) and 13.4094 (dB), 15 iterations

VI. ACKNOWLEDGMENTS

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