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FRAME, BIT AND CHIP ERROR RATE EVALUATION FOR A DSSS COMMUNICATION SYSTEM

EVALUACIÓN DE LA TASA DE ERRORES EN BIT, CHIP Y TRAMAS PARA UN SISTEMA DE COMUNICACIONES DSSS

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Abstract

The relation between chips, bits and frames error rates in the Additive White Gaussian Noise (AWGN) channel for a Direct Sequence Spread Spectrum (DSSS) system, in Multiple Access Interference (MAI) conditions is evaluated. A simple error-correction code (ECC) for the Frame Error Rate (FER) evaluation is used. 64 bits (chips) Pseudo Noise (PN) sequences are employed for the spread spectrum transmission.

An iterative Montecarlo (stochastic) simulation is used to evaluate how many errors on chips are introduced for channel effects and how they are related to the bit errors. It can be observed how the bit errors may eventually cause a frame error, i. e. CODEC or communication error. These results are useful for academics, engineers, or professionals alike.

Keywords: simulation, spread spectrum, multiple access, frame error rate.

Resumen

Se presenta la evaluación de la relación que tienen las tasas de chip, bits y tramas erróneas en un canal con Ruido Blanco Gaussiano Aditivo (RBGA) para un sistema de comunicaciones de espectro disperso de secuencia directa (DSSS) en Condiciones de Acceso Múltiple (CAM). La simulación emplea secuencias de Pseudo Ruido (PN) de 64 bits para la transmisión de espectro disperso.

Se utiliza una simulación de montecarlo (estocástica) para evaluar la cantidad de chips erróneos introducidos por los efectos del canal y cómo éstos se relacionan con los errores en bits. Se puede observar cómo los bits erróneos pueden causar eventualmente una trama errónea, es decir, un error en el CODEC o en la comunicación. Estos resultados pueden ser utilizados por académicos, ingenieros o profesionales del área.

Descriptores: simulación, espectro disperso, acceso múltiple, tasa de tramas erróneas.

Introduction

Simulation of communication systems has changed the way we design and implement communication systems.

Currently, there exists a great interest in modeling and simulation of communication systems. One can either evaluate a small part of the system, or evaluate general characteristics of different design situations in order to reduce the high cost of

system testing and even decrease the human risks. However, the researcher should keep in mind that simulation results are reality approaches in the same way that the model would adequately describe the system in question. The bit error rate can be considered a fundamental part in design and implementation of communication systems.

DSSS systems have been widely accepted in the commercial world of mobile communications; nonetheless, simulating this kind of system requires high signal processes, it is in the best interest of the researcher to utilize the fastest processors at one's disposal, as well as the adequate software (Harada, *et al.*, 2002). In this paper we present a software tool named SisCom, which can be used to evaluate Bit Error Rate (BER), Chip Error Rate (CER) and Frame Error Rate (FER) in DSSS Systems.

System characteristics

In the transmitter, information data frame is fed into a channel encoder, which is a linear block code (10, 6). The 10 bits codeword is then expanded for a spread spectrum code, as can be seen in figure 1. Next, the digital data is fed into a BPSK modulator.

Characteristics for codes in order to be used in CDMA are (Yang, 1998):

1. The cross-correlation should be zero or very small:

$$R_{xy}(0) = x^T y = \sum_{i=1}^I x_i y_i \quad (1)$$

where x and y are two sequences or codes.

2. Each sequence in the set has an equal number of 1s and -1s, or the number of 1s differs from the number of -1s by at most 1.
3. The scaled dot product of each code should be equal to 1:

$$R_{xx}(0) = x^T x = \sum_{i=1}^I x_i x_i \quad (2)$$

$$R_{xx}(0) / N = 1 \quad (2a)$$

The SisCom program has two types of code options for generating spread spectrum signals: Walsh codes and Pseudo Noise (PN) sequences. Walsh codes are generated using a 64 order Hadamard matrix (4), whose rows and columns are mutually orthogonal. The basic Hadamard matrix is (Lee, *et al.*, 1998):

$$H_2 = \begin{bmatrix} 0 & 0 \\ 0 & 1 \end{bmatrix} \quad (3)$$

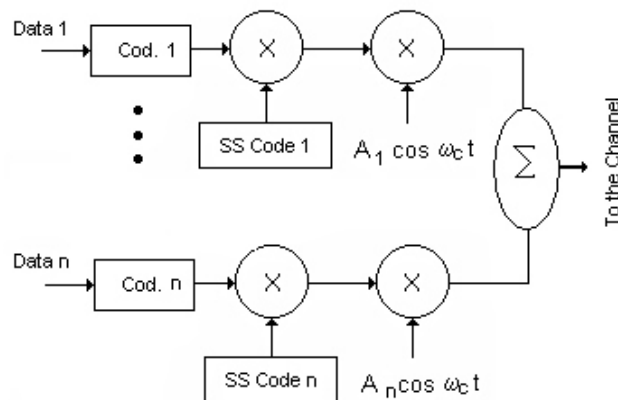


Figure 1. Transmitter

One can find higher order matrices as follows:

$$H_{2N} = \begin{bmatrix} H_N & H_N \\ H_N & H_N \end{bmatrix} \quad (4)$$

PN Sequences are generated with a MSRG (Modular Shift Register Generator) mechanization, known as Galois configuration. The sequence generation starts with the initial state vector: [1 1 1 1 1 1].

Figure 2 shows the MSRG mechanization used to generate PN sequences. Figure 3 shows the correlation function of PN sequences used (Lee et al., 1998; Harada et al., 2002).

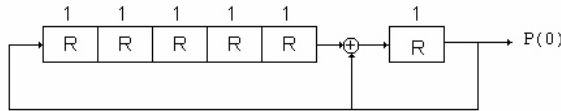


Figure 2. PN sequences generator

The output PN sequence is:

$PN_1 = [0, 1, 0, 1, 0, 1, 1, 0, 0, 1, 1, 0, 1, 1, 0, 1, 1, 0, 1, 0, 1, 0, 0, 1, 0, 0, 1, 1, 1, 0, 0, 0, 1, 0, 1, 1, 1, 1, 0, 0, 1, 0, 1, 0, 0, 0, 1, 1, 0, 0, 0, 0, 1, 0, 0, 0, 0, 0, 1, 1, 1, 1, 1]$. The other 62 vectors are delayed versions of the output sequence. Besides, this sequence has been added by an extra zero.

Continuous time analysis (for simplification) shows that the system wideband is mainly controlled by PN sequences as shown in (5), (6) and (7). The transmitter output signal $s(t)$ is (Stremler, 1998):

$$s(t) = A c(t) p(t) \cos \omega_c t \quad (5)$$

Where

A : Signal amplitude

$c(t)$: Spread signal

$p(t)$: Binary function representing the transmitted data

ω_c : Carrier Frequency

The power spectral density $S_{ch}(\omega)$ of the spread signal $c(t)$ is:

$$S_{ch}(\omega) = T_{ch} \text{sinc}^2(\omega T_{ch} / 2) \quad (6)$$

where

T_{ch} : Chip time

So that the power spectral density $S(\omega)$ of SS-BPSK modulated signal can be described as:

$$S(\omega) = \frac{1}{2} A^2 T_{ch} \text{sinc}^2[(\omega - \omega_c) T_{ch} / 2] + \frac{1}{2} A^2 T_{ch} \text{sinc}^2[(\omega + \omega_c) T_{ch} / 2] \quad (7)$$

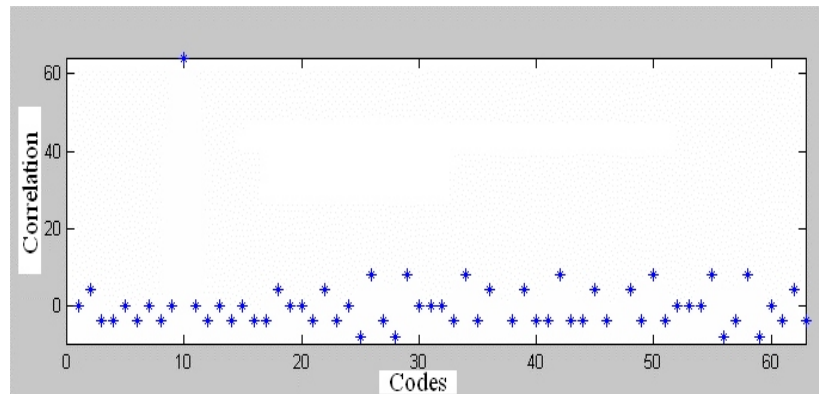


Figure 3. Correlation function for PN sequences with inserted zero

In the channel module of the SisCom program, the AWGN model and some other empirical and mathematical attenuation models are available; however, the most important quantity in this simulation is the Signal to Noise (S/N) ratio.

Figure 4 illustrates the correlation receiver used. In this block, chips, bits and frames are recovered. After one frame is recovered, the program goes back to the channel where noise and attenuation are added to the data signal in an iterative way.

As shown in figure 4, only one frame is recovered at the receiver, the other 19 frames are added only as interferers in the channel for this simulation.

Simulation

A very important characteristic of DSSS systems is the high noise tolerance. This simulation uses S/N ratios from 0.02 to 0.04 into the channel; it means that the noise amplitude is over 30 times greater than the data amplitude. This characteristic is known as “spread spectrum jamming margin”, which is directly proportional to the spread codes longitude. Even though the CER is high, the BER is maintained low.

When the amount of channel noise causes a bit error, the channel CODEC can offer a bigger link

confidence. However, if more than one bit error occurs in one frame, we will have one error in the communication system. Therefore error randomness is a very important characteristic for a good CODEC performance. In this case, “block interleaving” may be included to offer better performance in the case of burst errors (Proakis, 2000).

Instead of using perfect orthogonal codes, we use PN sequences with a low correlation. Simulation shows how this non-perfect characteristic causes extra interference in the channel due to the fact that more users are connected. Synchronization facilities and easier electronic implementation justify the employment of PN sequences; the extra interference is in this case the downside of using them (Figure 5).

Simulation characteristics:

- 1) Frames: 10 bits (640 chips)
- 2) CODEC: linear block (10,6)
- 3) Multiple access: PN20-PN40 sequences
- 4) Modulation: BPSK
- 5) Channel: AWGN
- 6) Demodulator: correlator
- 7) Perfect synchrony Tx-Rx, is assumed

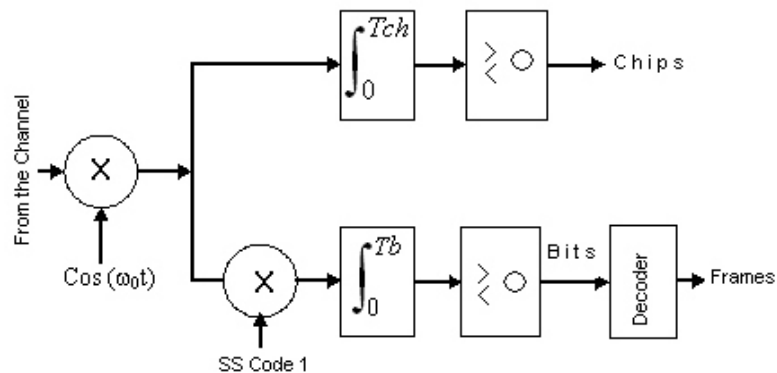


Figure 4. The receiver



Figure 5. View of SisCom program

SisCom program was developed in MATLAB and requires from 30 minutes to 40 hours for one 10^{-4} error rate simulation running in a Pentium 4 CPU, 2.8 GHz personal computer. This occurs because ten or more of the statistical events (errors) are needed in order to guarantee the results significance.

Results

Figure 6 illustrates the CODEC performance in multiple access conditions for 1, 10, and 20 users (blue, red, and black respectively). It is very difficult to compare these results to others. Because this software tool is unique, it is referenced only to underlying mathematics.

Figure 7 illustrates the FER (blue), BER (red) and CER (black) relations. FER is higher than BER for very low rates however, the convenience of using a CODEC is observed for high S/N ratios, where the CODEC takes advantage of random errors on bits; at the same time the CER maintains very high error rates.

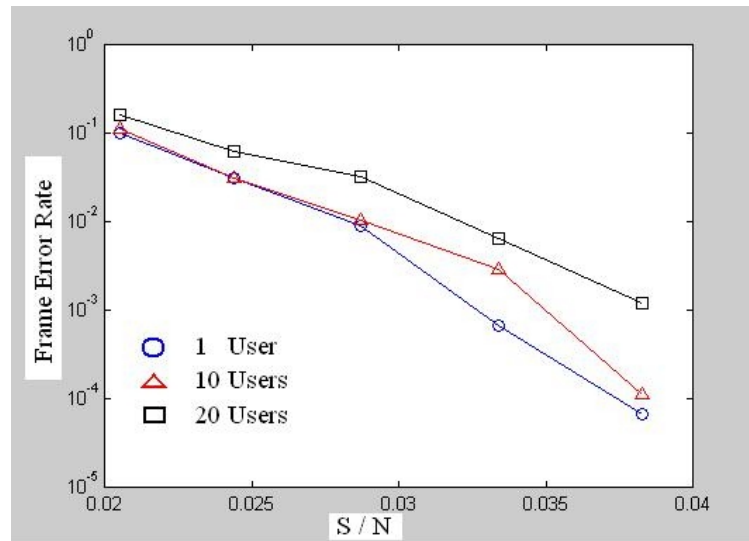


Figure 6. Frames error rate

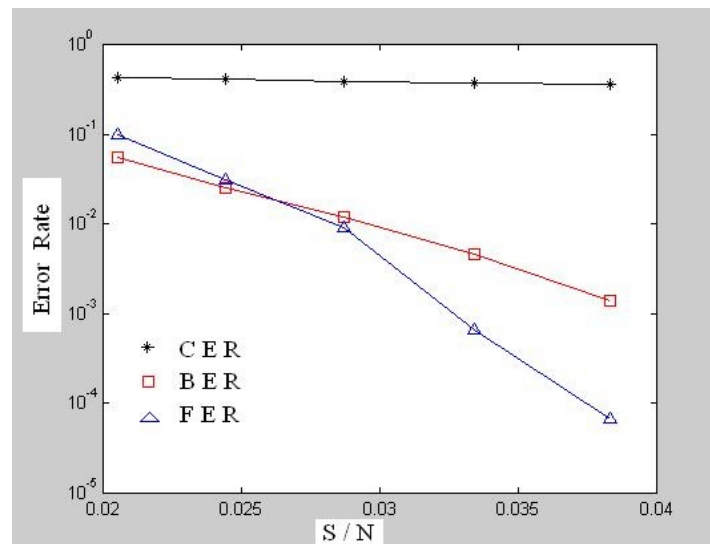


Figure 7. Chips, bits and frames error rates

Conclusions

We studied the behavior and the relation between chips, bits, and frames errors in the AWGN channel for a DSSS system in MAI conditions. It can be observed that low variations in CER, cause important variations in FER. Randomness of errors

in the channel is the key for good CODEC performance.

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