Improvised LDPC Coding for Reduced Error DVBT Communication under Fading Based Error

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Abstract — In this new technology world, we must have much more expectations on the basis of electronic system. Our expectation of doing better and in very less time is increasing day by day. We need to enhance much more things to perform better and much better. LDPC coding becomes a hottest topic in coding theory. In this research work, We describe Low density parity check codes (LDPC) and its significance, LDPC codes are very popular on the basis of its error detecting and error correcting methods. We already studied about the LDPC codes and its method. In this research work two ldpc codes are compared with DVBT communication system which is wireless and designing of wireless transreceiver and wireless coder and decoder using OFDM. We design the video data base of colored video under different resolutions. We design ldpc encoder and decoder under based technique and proposed techniques and simulation of video transmissions and reception in order to testing the performance of the system on the basis of PSNR (peak signal to noise ratio) , BER (bit error rate) , power , delay and memory.

Keywords——LDPC, LDPC coding, OFDM, DBVT communication, video transmission, LDPC encoder, LDPC decoder PSNR, BER, power, delay, memory.

I. INTRODUCTION

LDPC codes has been discovered long time ago and rediscovered after invention of turbo codes. These two codes are actors of revolution of error correcting codes theory. Low density parity check decoders has become a hot topic in recent year to meet the have experienced an amazing comeback in the past few years. The design and evolution of digital communication system emphasizing areas of new technology. LDPC (Low density parity – check codes) are the class of linear block codes. LDPC is originate from the feature and quality belonging to of their parity check matrix which contain only ones as compared to the quantity of zero’s. These codes are already equipped with very fast encoding and decoding algorithms. Low density parity check codes actually leads to coding and decoding techniques related to error correction technology which is very close to capacity. LDPC are characterized by sharpness of ones in the parity check matrix. It is a linear error correcting codes that lead to transmission of message over a noisy transmission channel.

II. LITERATURE SURVEY

Hisashi futaki et al. [1] described (LDPC) coded MIMO (Multiple inputs and multiple outputs) Systems with Iterative turbo decoding, in which a new MIMO-LDPC system with iterative TD (turbo decoding) (MIMO-LDPC-TD) using two LDPC encoders and two LDPC decoders to improve the performance of the MIMO-LDPC.

Lu et al. [2] described a class of structured LDPC codes with large girth. Turbo-structured LDPC (TS-LDPC) codes—composed of two sub trees connected by an inter-leaver. TS-LDPC codes with good girth properties are easy to design: careful design of the inter-leaver component prevents short cycles in its Tanner graph.

Byers & Takawira et al. [3] explained non-binary and concatenated LDPC codes for multiple-antenna systems. Coded multiple-antenna systems are a promising solution to the provision of high data rate and bandwidth efficient wireless communications. Low-density parity-check (LDPC) codes achieve excellent coding gains in AWGN channels.

Lu & Maura et al. [4] described PS (partition-and-shift) LDPC (PS-LDPC) code. PS-LDPC codes can be easily designed to have large girth. The code construction is simple to explain. He divided the bit and checked nodes in the Tanner graph into subsets and connected nodes in these subsets according to a set of parameters called shifts.


Suresh Dannana et al. [6] described the LDPC encoder is replaced with data encoding schemes in control to reduce the dynamic power consumption in Low Density Parity Check Techniques (LDPC).

Shubham Singh et al. [7] represented an efficient LDPC decoder for which they simply utilized matrix multiplication techniques. The decoder designed can detect and correct the errors of the incoming, encoded data signal of 12 bits length.
III. PROBLEM FORMULATION

Many attempts have been made to solve the problems of error detecting and reduce the complexity by designing ldpc code. Many decoders have been designed to detect and correct the error. But the complexity of ldpc codes are still too high which is major problem to be solve out, till now we are not able to find out that out of these decoder which one will be the best performer so that in future, we will use to perform better on the basis of particular parameter and achieving better result in very less time.

The existing algorithm has loss of performance by decoding complexity. This reduces the time as well as data transmitted/decoded. The existing decoder lacks in perfect communication between DVBT transmitter and receiver, as issues may rise due to interference caused by external factors.

The existing implementation in wireless system issue has not been resolved, which covers the actual performance of the system of broadcast and decoding unit.

LDPC based decoding in Image Processing based data is not covered in previous approach, this results in non-real-time evaluation of data which is not the case in real-time application.

This research work focuses on designing a LDPC decoder and comparing it with different decoder and communication method to show which one will perform better on the basis of bit error rate, power, delay, memory, peak signal to noise ratio. The motivation behind using this analysis and comparison to find the better decoder to achieve accuracy and reduce complexity in very less time.

IV. INTRODUCTION TO OFDM

Orthogonal frequency division multiplexing scheme used as a digital multi carrier modulation method. It is a form of signal modulation that divides a high data rates modulating stream placing them on to many slowly modulated narrow band.

OFDM is a modulation format that is being used for many of the latest wireless and telecommunication standard. OFDM has also been adopted for a number of broadcast standard and digital video broadcast standard. In wireless communication fading is deviation of the attenuation affecting a signal over certain media, fading may vary with time radio frequency and is often modulated randomly.

V. OBJECTIVE

The research has been focused to achieve these objectives:

- To design a Decoder to detect and correct the error coming from data signal of 24 to 64 bits length.
- To implement ldpc encoder / decoder with low data complexity and discrete packet size.
- To implement wireless (DVBT) communication system for comparing two LDPC codes.
- Simulation of wireless distortion as in real time like fading channels using the matlab rf tool kit with Rayleigh and rician type shadowing with different window size.
- Simulation of video transmissions and receptions under the given environment of diversity in variable traffic and interference from other video channels.
- Testing the performance of the proposed system on the basis of PSNR, BER or power, delay and memory.
VI. DESIGN AND IMPLEMENTATIONS

The presented work has been divided into these sections:

1. Designing of video data base of colored video under different resolutions. Divide the video in to frames by applying primary one dimension coding.
2. Conversion of colored frame into gray layer and sampling of 2D gray layer into 1D and applying burst error to the frame data and then bit spread for randomize the data.
3. Addition of zero to the original data encoder using 64 bit value and Designing of wireless transreceiver and wireless coder and decoder using OFDM.
4. The OFDM is simulated with Gaussian channel with the noise mixing with a value up to 15 DB.
5. Merge the two LDPC systems with transceiver systems.
6. Receive the signal after fading channel techniques under based and proposed techniques. and find the bit error ratio per 6 bit count and Calculate the power required in coding and transmitting the data. After transmitting the signal travel with AWGN noise is received at the receiving decoder where 64 bit description take place and interleaver is removed zero and burst error rate is calculated and original frame data is retrieved.
7. Reshaping the frame from 1D to 2D and Comparison of the performance of existing and proposed system on the basis of parameters like PSNR, BER delay, memory, and power.

VII. RESULT AND DISCUSSION

This shows the evaluated output image in term of BER which achieve value of 91.6667 under code rate 2/3 burst error rate 7 and the next evaluation is the visual evaluation. The visual evaluation shows output constructed image similar in structure to the input images with minor difference in edge values with high contrast ratio. Due to the reduction in edge values the output image shows some lossy data. The result image observed for the same image frame in code rate. The difference occurred in the highest code rate ratio which was 7/8.

Fig. 2 Block diagram for the proposed work

Fig. 3 GUI representation of videobroadcast model with its various user defined variables of set parameters

Result:

BER=% 91.6667  Coderate= 2/3  Burst BER= 7

Fig. 4 Result of of videobroadcast model with its various user defined variables of set parameters.
Fig. 5 Coded image for transmission

Fig. 6 Decompressed receive images

The output shows visible evaluation of result for 1:4 Frames showing the BER ratio and channel error probability for different frames of video under the specified parameter.

VIII. CONCLUSION

The proposed research work is based on the previous issue concerning delay and SNR ratio from DVBT broadcast under 128 frames. The new system device was to improve the efficiency using BER analysis and SNR. The proposed system has also reduce the delay and power consumption ratio with respect to previous approach. The experience done both images and video under different dvbt ratio of code parameter like BER code rate.
REFERENCES


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