WIRELESS VIDEO TRANSMISSION OVER UWB CHANNEL USING FUZZY BASED RATE CONTROL TECHNIQUE

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ABSTRACT

Communication has always been on the rise especially when it comes to the transmission of video signals. The problems faced by video transmission include consumption of large bandwidth and video quality at the receiving side. This can be overcome by having a proper video coding and proper control over the rate of transmission based on channel conditions. In our paper, we focus on these areas and present transmission control technique for wireless video transmission over UWB channel. In our technique, the major emphasis lies on the H.264 encoder and the use of fuzzy controller. The technique comprises of three modules, namely transmission module, control module and receiver module. Proposed technique is implemented using Matlab and the evaluation metrics used are BER and MSE. Comparative analysis is made by comparing our proposed technique to the SamiaShimu et al.[1] results. From the results, the net average BER came about 0.034 for the base compared to 0.032 for the proposed technique and net average MSE came about 117 for the base compared to 48.8 for the proposed technique. The obtained values show the effectiveness of the proposed technique by having lower MSE and BER values.

Keywords:- Transmission Control, Fuzzy Controller, PSNR, MSE, BER, UWB, H.264 Encoder And Decoder, QPSK.

1. INTRODUCTION

Wireless Internet video services are anticipating to be widely organized with the increasing real-time Internet video applications and the fast development of wireless networks. Different voice and data services, video applications get through comparatively large bandwidth and need heterogeneous quality of service (QoS) necessities in terms of hold-up, bit error rate (BER) and video quality [2]. A constant growth in wireless communication systems has been accomplished in recent years. In the near outlook, Wireless contact to the Internet may outperform all other types of admittance. It is probable that mobile users will anticipate related levels of service quality as wire line users. One of the features behind this development was the supremacy of digital wireless communication systems, which have better bandwidths and can incorporate voice and data communications [7]. Wireless image and video transmission has turn into an important element of wireless systems in wireless communication. It is the most important structure bottleneck as it needs more bandwidth than transmission of the other data sources. Conventional communication systems, which regarded as detach source and channel coding, are not appropriate for broadcasting video in wireless channel since they failed to consider the highly time-varying channel features, the harsh power of channel loss on the encoded source, and the imbalanced significance of transmitted bits [14, 15].

One of the most significant and challenging objective of current and future communication is transmission of high quality images and videos from source to end quickly with slightest fault, where inadequacy of the bandwidth is a major problem. By the arrival of multimedia communications and the data superhighway has specified increase to a vast demand on high-performance communication systems [10]. Multimedia transmission of signals over wireless links is regarded as one of the major applications of future mobile radio communication
systems. Conversely, compared to voice applications, such applications need the apply of comparatively high data rates (in the Mbps range). With such necessity, it is very challenging to offer acceptable quality of services as computed by the Root Mean Square Error (RMSE) owing to the restrictions enforced by the wireless communication channels such as fading and multipath propagation. Besides, the user mobility formulates such a task more hard of the time fluctuating nature of the channel [3].

Using a huge portion of the radio spectrum, Ultra-wideband is a radio technology which may be applied at a very low energy level for short-range, high-bandwidth communications. UWB has long-established applications in non-cooperative radar imaging. Latest applications target sensor data collection, precision locating and tracking applications [15]. Related to spread spectrum, UWB communications convey in which does not obstruct with conventional narrowband and carrier wave employed in the similar frequency band. Ultra-wideband is a technology for broadcasting data spread over a huge bandwidth (>500 MHz). Ultra wideband was previously known as "pulse radio", however presently UWB is defined in terms of a transmission from an antenna for which the discharged signal bandwidth goes beyond the lesser of 500 MHz or 20% of the center frequency. Therefore, pulse-based systems—where every transmitted pulse engages the UWB bandwidth. Pulse-based UWB radars and imaging systems are liable to employ low repetition rates. Alternatively, communications systems support high repetition rates (typically in the range of one to two gigapulses per second), hence enabling short-range gigabit-per-second communications systems. Every pulse in a pulse-based UWB system occupies the complete UWB bandwidth, different carrier-based systems which are mattered to deep fading and intersymbol interference [1].

The chief resources accessible to communications systems designers are power and bandwidth with system complication. Therefore, for suitable utilization of the communication, it is very important to employ methods that are both power and bandwidth competent [3]. To accomplish this, control methods are important one. An analytical paradigm modelling of a rate-controlled MPEG encoding system is one of the control method for MPEG video transmission [8, 11], in which a feedback creates the quantize scale parameter reliant on both the condition of the output counter/buffer in the controller and the movement level of the frame to be determined. To this conclusion, an SBBP queuing system is applied, where the SBBP (switched batch Bernoulli process) arrival process is different according to the quantizer scale parameter selected by the addressed comment law at the frame level. In terms of both the alteration brought in by the encoding system and the resulting output process information, the analytical structure permits us to assess system presentation [9]. Expected, developing control method for different encoding system and channel parameters are the present spotlight of wireless video transmission.

The technique consists of three modules, namely transmission module, control module and receiver module. In the transmission module, the video is H.264 encoded, interleaved and QPSK modulated. It is transmitted through the UWB channel. In the receiver side, the signal is de-modulated, de-interleaved and decoded. In control module, transmission rate and number of frames per transmission are found out using fuzzy logic with the help of PSNR and MSE values obtained from the receiver side.

The rest of the paper is organized as follows: Section 2 gives the literature review. Section 3 describes the proposed technique. Section 4 gives results and discussions. Conclusions are summed up in Section 5.

2. REVIEW OF RELATED WORK

For wireless video transmission, Literature offers numerous methods. Now, we reassess some of the works accessible in the literature. XiaoanLuet al [4] has suggested for minimizing the total power utilization of a mobile transmitter owing to source compression, channel coding and transmission subject to a fixed end-to-end source alteration. They demonstrated both on a theoretical class of sources and channels and on a practical H.263 video transmission system through a wireless channel. Concert under dissimilar channel environments and execution schemes were examined. By a considerable factor and prolong battery life significantly compared with fixed parameter settings, their numerical study shown that optimized settings have been diminished the total power utilization. For video allocation over 802.11 wireless networks based on packet retransmissions, an intellectual, rate-limited multicast video transmission optimization plan has been offered by Victor Miguelet et al [5]. They have engaged the features of the encoded stream collectively with the
behaviour of the wireless channel. They employed standard Stochastic Dynamic Programming techniques, optimal control policies were attained off-line. These policies were most favourable in the sense of minimized the anticipated distortion at the terminal. Besides, the on-line complexity was extremely low as the optimization problem was worked out off-line. The presentation of their plan has been assessed in an actual scenario and compared with that of a restricted rate ARQ algorithm. With a higher precedence, the Experimental results demonstrated a higher packet revival rate and an improved security of information.

Due to time-varying channel states, a Wireless video transmission was prone to potentially low data rates and unpredictable degradations have been offered by Leonardo Badia et al [6]. Such dreadful conditions were hard to conquer applied conventional video coding methods. To fit the existing channel conditions, Scalable video coding presented a stretchy bits stream that can be vigorously modified. For approaching this challenge, advances in scalable video compression methods, such as, implemented scalable extension of H.264/AVC, with latest advances in wireless access technologies presented potentials. They comprised a content-aware scheduling and resource distribution that employed a gradient-based scheduling framework in combination with scalable video coding methods to offered multiple high quality video streams over a variety of operating conditions to multiple users. The Experimental result demonstrated that the presentation has better than conventional content-independent scheduling methods. A. Lombardo et al [10] have suggested a statistical features of MPEG traffic have been examined and an exact and treatable Markov-based model, which has been able to incarcerated not only statistics previously regarded, such as the autocorrelation function and the Gamma-shaped possibility density function, however, moreover over the relationship among different frames belonging to the similar GoP. In order to incarcerate both the inter-GOP and the intra-GOP relationship, they have comprised that structured in two levels. The famous inverse eigenvalue problem has been worked out in the discrete-time domain to attain the first level of the model. At last, the model precision has been showed by comparing presentation of an ATM multiplexer loaded by an MPEG video source and a combined of external traffic with the loss possibility computed by simulating the system applied actual traffic sequences as driven traces.

A unifying structure for competent encoding, transmission, and quality assessment of atherosclerotic plaque ultrasound video has been suggested by A. Panayides et al [12]. The strategy was based on a spatially fluctuating encoding plan, where video-slice quantization parameters were differed as a function of diagnostic importance. Based on fragmentation algorithm, Video slices were routinely set. They were afterwards en-coded applied an adapted version of H.264/AVC flexible macro block ordering (FMO) method that permits changeable quality slice en-coding and redundant slices (RSs) for resilience over error-prone transmission means. They found that several objective quality evaluation measured calculated over the plaque video slices gave incredibly good relationships to mean opinion scores (MOSs). At this point, MOSs was calculated applied two medical experts. Experimental results demonstrated that accomplished improved presentation in noisy environments, whereas at the similar time accomplished significant bandwidth demands reductions, offered transmission over 3G (and beyond) wireless networks. Fangwen Fu and Mihaela van der Schaar [13] have devised the problem of multi-user wireless video transmission as a multi-user Markov decision process (MUMDP) by plainly regarding the users’ heterogeneous video traffic features, time-varying network conditions with, significantly, the active coupling among the users resource allocations a cross time, which has been frequently disregarded in existing multi-user video transmission solutions. Different in conventional multi-user video transmission solutions stemming from the network effectiveness maximization framework, the decomposition facilitated every wireless user to independently work out its own local MDP (i.e. dynamic single-user cross-layer optimization) and the network coordinator to revised the Lagrangian multipliers (i.e. resource prices) based on not only current. The Experimental results showed the competence of the MUMDP framework as compared to conventional multi-user video transmission solutions.

The consequences of these bandwidth deviations were examined in the case of real-time Motion Picture Experts Group (MPEG) video transmission has been suggested by A. Cernuto et al [9]. In order to acclimatize its emission rate to the current bandwidth presented by the wireless link, the MPEG encoder was managed. To this conclusion, the encoded quality was reduced by the source rate controller when the transmission rate has to be diminished due to a raise in the channel BER, while
it has been enhanced when the transmission rate could be raised due to a decreased in the channel BER. To model the scenario being regarded, a Markov-based model, represented as SBBP/SBBP/1/ K, has been launched. The analytical structure assessed of the presentation of the system and could be applied to optimize the plan of a video transmission system for wireless channels, offered the instruments to derive the trade-off among information bribery in the wireless channel and MPEG video encoding excellence.

3. PROPOSED TRANSMISSION CONTROL TECHNIQUE FOR WIRELESS VIDEO TRANSMISSION OVER UWB CHANNEL

Transmission of video through proper video coding and proper control over the rate of transmission based on channel conditions are two major tasks in wireless video transmission. To compact these, we make use of fuzzy controller based rate control and H.264 encoding in our proposed technique. The technique consists of three modules of transmission module, control module and receiver module. In the transmission module, signals are encoded with H.264 encoder, modulated and transmitted over UWB channel. In the control module, fuzzy controller is used for controlling the rate of transmission based on PSNR and MSE parameters. In receiver module, the reception is made and is decoded. Figure 1 gives the block diagram of the proposed technique.

3.1 Transmitter Module

The HD video is processed and transmitted through the channel in this module. For encoding of the video, we make use of H.264 encoder. The encoded video is subsequently done interleaving operation and QPSK modulation before the final transmission through antenna to the channel. The block diagram of the transmitter module is given in figure 2.

Figure 1: Proposed Transmission Control Technique For Wireless Video Transmission Over UWB Channel

Figure 2: Transmitter Module
a. H.264 Encoder

The input HD video is initially encoded with H.264 encoder. H.264 is a video coding standard which plans for high presentation and discovers applications in broadcasting, storage on optical, magnetic tool and DVD, conversational services over ISDN, Ethernet, LAN, DSL, wireless and mobile networks, modems. It is moreover applied in video-on-demand and multimedia messaging services (MMS).

The H.264 encoding procedure is carried out in two layers namely Video Coding Layer (VCL) and Network Abstraction Layer (NAL). In VCL, the video content is proficiently represented and in NCL, VCL representation is formatted and it gives the header information. Here, H.264 encoding is maintained for video coding in either progressive or interlaced frames, which may be mixed together in the same sequence.

Network abstraction layer

NAL arranges the information and provides the header information in a suitable way satisfactory for the transport layers or storage media. The entire information enfolded in NAL modules, where every module contains an integer number of bytes. The NAL module utters a normal format to be used in packet-oriented and bit stream systems.

Video coding layer

H.264 video coding layer comprises of a combination of temporal and spatial prediction, in union with transform coding. Figure 3 gives the block diagram of the video coding layer.

![Figure 3: Video Coding Layer](image)

The picture is separated into permanent sized macro-blocks primarily. The primary picture in concern is generally intracoded which denotes that it does not employ any information apart from what is enfolded inside the picture. In this case, each sample is predicted using spatially neighboring samples of beforehand coded blocks. Inter coding is applied usually for other pictures. Forecast based on formerly decoded pictures is executed in the inter coding. Now, at first the motion data comprising the reference picture is chosen and next, a spatial dislocation is executed to all models of the block. The motion information is broadcasted as side information and presents the inter prediction signal.

The difference between the original and the predicted block is calculated \(d\) and is subsequently transformed \(p\). Here let the number of blocks be equal to \(N\). Transformation is carried out using the Discrete Cosine transform which is given by the formula (equation 1):

\[
D = \begin{pmatrix}
D_{11} & D_{12} & \cdots & D_{1N} \\
D_{21} & D_{22} & \cdots & D_{2N} \\
\vdots & \vdots & \ddots & \vdots \\
D_{N1} & D_{N2} & \cdots & D_{NN}
\end{pmatrix}
\]
Scaling and quantization operations are done for the transformed coefficients. For quantization, scalar quantization is used. Consequently, the coefficients are done entropy coding and are transmitted alongside with the side information which may be either intra-frame or inter-frame prediction. Here, decoder is contained in the encoder to perform prediction for the subsequent blocks or the next picture. As a result, the quantized transform coefficients are inverse scaled and inverse transformed as in the same procedure in the decoder side, giving the decoded prediction difference. This difference obtained is summed up with the prediction and the sum is given to the de-blocking filter which provides the decoded video as its output.

b. Interleaver

Interleaving is a common method frequently used in communication systems and finds a range of applications. It is a hardware device in which symbols form fixed alphabets are taken as the input and it outputs the same symbols in a dissimilar temporal order. It is mainly used along with fault correction block codes to widen errors over a number of blocks so that the extreme number of faults in every block is below the number of correctable faults.

C. QPSK Modulation

Quadrature Phase Shift Keying (QPSK) modulation is a type of bandwidth preserving modulation method. Here, the information carried by the transmitted signal is contained in the phase. Transmitted signal is defined (equation 2) as:

\[
r(t) = \begin{cases} 
    \frac{2\epsilon}{T} \cos\left(2\pi t + (2i - 1)\frac{\pi}{4}\right), & 0 < t < T \\
    0, & \text{else}
\end{cases}
\]  

Two orthonormal basis functions are defined by (equations 3 & 4):

\[
\phi_1(t) = \sqrt{\frac{2}{T}} \cos(2\pi ft), \quad 0 < t < T
\]

\[
\phi_2(t) = \sqrt{\frac{2}{T}} \sin(2\pi ft), \quad 0 < t < T
\]

Figure 4 gives the QPSK transmitter block diagram. Here the incoming sequence is first transformed into polar form with the use of Polar Non-Return to Zero (PNRZ) encoder. This is divided into waves consisting of odd and even numbered bits by means of a Demultiplexer. The waves are then multiplied with the respective orthonormal signal functions of \( \phi_1(t) \) and \( \phi_2(t) \). The resultant waves are then summed up to have the final QPSK modulated signal.

Hence in the transmitter module, the input HD video is H.264 encoded, interleaved and QPSK modulated and transmitted to the channel through antennas.
3.2 Receiver Module

In the receiver module, the received signal is demodulated, de-interleaved and decoded to have the output. The block diagram is given in figure 5:

![Figure 5: Receiver Module](image)

a. UWB channel

UWB stands for Ultra Wide band and has obtained its name as it employs remarkably extensive transmission bandwidths surplus of 3 GHz. UWB has the capacity for exact position location and ranging, absence of substantial fading, great multiple access capability, protected communications, and feasible undisturbed material penetration. These all supply to having covered and more rapidly wireless networks. For wireless positioning and ranging products, it moreover assists in building enhanced plans. The channel formulates use of nanosecond impulses which offers the bandwidth in the range of 3-7 gigahertz. UWB moreover has the benefit of containing low power transmission model which allows the system to labour over a large bandwidth without meddling with prevailing narrowband systems.

b. QPSK Demodulator

![Figure 6: QPSK Receiver](image)
QPSK demodulator consist of correlators with a common input and supplied with reference orthonormal signals of $\phi_1(t)$ and $\phi_2(t)$. The block diagram of the QPSK demodulator/ receiver is shown in figure 6. The received signals are multiplied with the orthonormal functions and are fed to the respective correlators. The correlator output signals are compared with the threshold set and the respective decisions are taken based on the conditions. Finally the two are combined in a multiplexer to produce the estimate of the original signal.

C. De-Interleaver and decoder

In the receiver section, interleaved data is settled back into the original sequence by the use of de-interleaver. Subsequently, the decoding is carried and H.264 decoder which is shown in figure 7.

![Figure 7: H.264 Decoder](image)

After the de-interleaving, the data is fed to the H.264 decoder where initially entropy decoding is carried out. After decoding, it is inverse quantized and inverse scaled which is subtracted from the intra frame prediction or motion compensation (in case of inter frame) to result in the difference image. The prediction is carried out with comparison to the reference signal which is the feedback from the output signal. The output will be HD video and also parameters like PSNR and MSE values.

3.3 Rate Control Module

Usage of wireless internet video services is on a rapid rise and finds applications in major areas. But normal voice and data services differ from video applications in the fact that videos requires larger bandwidth and better quality of service with respect to the parameters like delay, fading, interference, bit error rate and video quality. The video transmission requires high quality videos to be transmitted from source to the destination rapidly with minimum error. The vital problem faced by such a system is the bandwidth limitation and the power consumption. For proper utilization of bandwidth and power, we need to take the transmission rate and the number of frames per transmission into account. That is, by proper controlling the parameters of transmission rate and number of frames, we can give rise to much efficient bandwidth utilization in HD video transmission. In our proposed technique fuzzy logic is employed which takes into account PSNR and MSE of the output video to control the transmission rate and the number of frames per transmission. Transmission of video through proper video coding also forms a major part for which we have used H.264 encoder. The block diagram of the control module is given in figure 8.
Peak Signal to Noise Ratio (PSNR) and Minimum Squared Error (MSE) of the output are found out and are given as the input to the fuzzy logic. The fuzzy logic in return produces the output values for rate of transmission and number of frames per transmission which will be given to the H.264 encoder in the transmitter module.

The visual quality of the received HD video computed the peak signal-to-noise ratio (PSNR). It is defined as the ratio of the maximum possible power of video signal to the power of corrupting noise and is normally given in logarithmic decibel scale. Mean squared error (MSE) can be defined as the difference between estimated values and the original value. It computes the average of the squares of all errors. MSE (equation 5) and PSNR (equation 5) can be defined as:

\[
MSE = \frac{1}{k} \sum_{i=1}^{k} \hat{X}_i - X, \quad (5)
\]

\[
PSNR = 20 \log_{10}(I_{max}) - 10 \log_{10}(MSE), \quad (6)
\]

Where, \( \hat{X} \) is the vector of \( k \) predictions, \( X \) is the vector of true values, \( I_{max} \) is the maximum possible pixel value. The MSE and PSNR are given to the fuzzy module as the input.

Fuzzy logic converts and processes the crisp values to fuzzy categorical values. In our case, we make use of triangular membership function, in which the input data values are three fuzzy values. Let the values be represented by low, medium and high represented by \( i \), \( j \) and \( k \). These form the three vertices of the triangular membership function defined by \( f(x) \). The triangular membership is defined by the formula (equation 7):

\[
f(x) = \begin{cases} 
0, & \text{if } x \leq i \\
\frac{x-i}{j-i}, & \text{if } i \leq x \leq j \\
\frac{k-x}{k-j}, & \text{if } j \leq x \leq k \\
0, & \text{if } x \geq k 
\end{cases} \quad (7)
\]

Figure 9 shows the plot for triangular membership function. The three curves are for low, medium and high.

Hence, initially the PSNR and MSE values are fuzzy converted and subsequently checked with the predefined rules so as to output the required transmission rate and the number of frames per transmission. The three predefined rules are:
• If PSNR is LOW and MSE is HIGH, THEN transmission rate is HIGH and number of frames per transmission is LOW.
• If PSNR is MEDIUM and MSE is MEDIUM, THEN transmission rate is MEDIUM and number of frames per transmission is MEDIUM.
• If PSNR is HIGH and MSE is LOW, THEN transmission rate is LOW and number of frames per transmission is HIGH.

The obtained transmission rate and the number of frames per transmission is given into the H.264 encoder in the transmitter module.

4. RESULTS AND DISCUSSIONS

In this section, we discuss the results obtained for our proposed technique. In section 4.1, details about the HD videos used for experimentation is given and the evaluation metric employed are discussed. In section 4.2, comparative analysis is made where our proposed results are compared with SamiaShimu et al.[1] results.

4.1 Implementation And Evaluation Metric

Proposed transmission control technique for wireless video transmission is implemented using Matlab on a system having i-7 Pentium processor having 3 Giga Hertz and 4 Gigabytes of RAM. The technique is tested on three videos namely: Foreman video, Hall video and Akiyo video. Image sample from all the three videos are given below:

Assessment metric used is BER (Bit Error Rate) and MSE. The Bit Error occurs when the received bits of the data stream over a communication channel varies from the transmitted signals. This occurs owing to modification of the signal which may happen due to the interference of not needed signals, noise effects, distortions or bit synchronization errors, multipath fading, attenuation. The bit error rate or bit error ratio (BER) is the ratio of bit errors to the total transferred bits during the particular time interval. BER is presentation measure applied for calculating the concert or the functionality of different methods and systems.

4.2 Comparative Analysis

We compare our proposed method to the SamiaShimu et al.[1](existing paper) with the help of evaluation metrics. We plot BER vs. graphs from which we can infer the performance of the system. $\frac{E_b}{N_o}$ is the energy per bit ($E_b$) to noise power spectral density ($N_o$) ratio of the received signal. $\frac{E_b}{N_o}$ is basically a
normalized signal-to-noise ratio (SNR) measure of the signal. $BER\ vs. \ \frac{E_b}{N_0}$ plots are made for all three videos and are given in figure 13, 14 and 15.

From figures 13, 14 and 15, we can observe that our proposed technique has got lower curve indicating lower BER and high performance of the system. The average values of BER are computed for each video and are given in table 1 and figure 16. MSE is also taken for each video and is given in table 1 and plotted in figure 17.

Table 1: Average BER And MSE Values Obtained For The Videos

<table>
<thead>
<tr>
<th></th>
<th>BER</th>
<th>MSE</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>SamiaShimu et al.[1]</td>
<td>Proposed</td>
</tr>
<tr>
<td>Video 1</td>
<td>0.0341</td>
<td>0.0340</td>
</tr>
<tr>
<td>Video 2</td>
<td>0.0340</td>
<td>0.0314</td>
</tr>
<tr>
<td>Video 3</td>
<td>0.0343</td>
<td>0.0323</td>
</tr>
</tbody>
</table>
From the table 1 and figures 16-17, we can infer that our proposed technique has performed well by achieving lower BER and MSE compared with SamiaShimu et al.[1]. We can see that the net average BER for all three videos came about 0.034 for SamiaShimu et al. [1] compared to 0.032 for our proposed technique. Similarly, net average MSE for all three videos came 117 for the SamiaShimu et al. [1] compared to 48.8 for our proposed technique. These values indicate the lowering of errors and higher reception of the signal with better clarity and fidelity.

5. CONCLUSION

In this paper, we present transmission control technique for wireless video transmission over UWB channel using fuzzy logic and H.264 encoder. Transmission module, control module and receiver module forms various modules of the technique. H.264 encoding is done in the transmitter module and subsequently modulated and transmitted through the UWB channel. The received signal is de-modulated, de-interleaved and decoded in the receiver module. Transmission rate for the encoder is found out in the control module with the help of PSNR and MSE values computed from the receiver side. The technique is implemented using Matlab. BER and MSE are evaluation metrics employed. The proposed technique is compared with the SamiaShimu et al. [1]results. From the results, the net average BER came about 0.034 for the base compared to 0.032 for the proposed technique and net average MSE came about 117 for the base compared to 48.8 for the proposed technique. The attained result values
show the efficiency of the proposed technique by having lower MSE and BER values.

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