

Effect of Video Streaming Space–Time Characteristics on Quality of Transmission over Wireless Telecommunication Networks

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Abstract—The spate in popularity of multimedia applications has led to the need for optimization of bandwidth allocation and usage in telecommunication networks. Modern telecommunication networks should by their definition be able to maintain the quality of different applications with different Quality of Service (QoS) levels. QoS requirements are generally dependent on the parameters of network and application layers of the OSI model. At the application layer QoS depends on factors such as resolution, bit rate, frame rate, video type, audio codecs, etc. At the network layer, distortions such as delay, jitter, packet loss, etc. are introduced. This paper presents simulation results of modeling video streaming over wireless communications networks. The differences in spatial and time characteristics of the different subject groups were taken into account. Analysis of the influence of bit error rate (BER) and bit rate for video quality is also presented. Simulation showed that different video subject groups affect the perceived quality differently when transmitted over networks. We show conclusively that in a transmission network with a small error probabilities (BER = 10⁻⁶, BER = 10⁻⁵), the minimum bit rate (128 kbps) guarantees an acceptable video quality, corresponding to MOS > 3 for all types of frames.

Index Terms— video streaming, trace file, BER, PSNR, MPEG Codec

I. INTRODUCTION

The growing popularity of multimedia applications brings with it the attendant necessity for optimizing the distribution of telecommunication network bandwidth. To a certain extent, the quality of video playback is dependent on application type. For example, in the playback of highly dynamic events such as sports and films, it is imperative to maintain a high video quality, while for relatively static events such as newscast and videoconference the accent would be more on the content. Contemporary telecommunication networks are required to support the quality of different applications with varying QoS levels [1]. The requirements on QoS are as a rule dependent on network and application layer parameters [2]. At the application level, QoS depends on such factors as bit rate, frame speed, types of video- and audio-codecs, etc. Some

distortions such as delay, jitter, packet loss, etc are introduced at the network level.

Normally, data transmission over wired networks with limitless bandwidth is characterized by a very low probability of bit error occurrence. However, due to the unpredictability of real-time transmission conditions, communication over a wireless network has certain peculiarities [3], [4], [5]. Wireless channels are characterized by independent and randomly distributed bit errors. It is for this reason that the White Gaussian Noise (WGN) model is used in the modeling and simulation of wireless channels. In this model, a bit in the video sequence is distorted (inverted) with a priori probability. [6] – [8] give detailed explanation of the effect of bit error on video quality during transmission.

The effect of different types of video subjects on the quality of video playback (for example from static e.g. newscast to highly dynamic e.g. sporting events) as related to the parameters of network and application layers is however not usually considered in the literature. The main objective of this paper is to fill this glaring gap. To this end we present two questions that are very important in relation to the network and application layers:

- Q1.** What is the minimum bit rate value for all types of video subjects for transmission over telecommunication networks that will guarantee acceptable QoS (PSNR > 27 dB), which corresponds to a MOS > 3 [9]?
- Q2.** What is the acceptable number of error bits for all types of video subjects, and what consequently is the limit (threshold) vis-à-vis playback quality under which viewer experience remains of acceptable quality?

II. EXPERIMENT

In order to answer these two questions, video clips were grouped based on their inherent spatial and time redundancies [10]. Next, an experiment of video transmission over a wireless network under various transmission conditions was conducted and the threshold for high, medium and low quality were consequently determined.

A. Classification of video into subject groups (SG)

The process of grouping video streams into different subject groups makes it possible to examine members of a particular group in terms of their similar characteristics. This allows for priority control and consequently optimization of the bandwidth of a given video stream. An automatic

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classification of video subject will make it possible to forecast video quality with a priori probability.

Defining such dynamics is of great interest for video coding, since the space-time characteristics of video signal defines the effectiveness of the coding procedure. In addition to quality measurement, it is possible to calculate the spatial and time features of the video (Fig. 1). As such, predicting video quality in relation to the dynamics of the subject based on the change in MOS becomes possible [11].

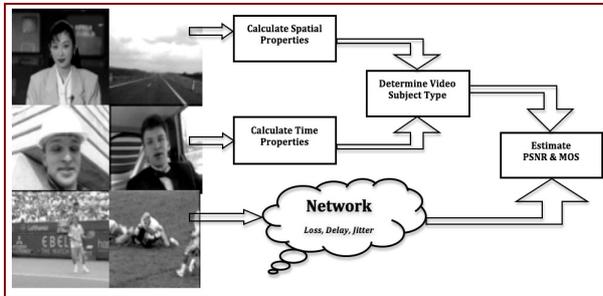


Fig. 1. Method of estimating video quality based on subject type

The subject of each video clip may differ considerably from others depending on its dynamics (i.e. the spatial complexity and time activity of the image). In the space-time plan introduced in [9], each video clip (of limited duration and homogeneous content) may be presented in the Cartesian coordinate system, where the spatial characteristic is on horizontal axis, and the time characteristic is on the vertical axis (Fig. 2).

In line with the approach depicted in Fig. 2, each video clip depending on the dynamics of its subject group may be classified into one of the following four categories, namely:

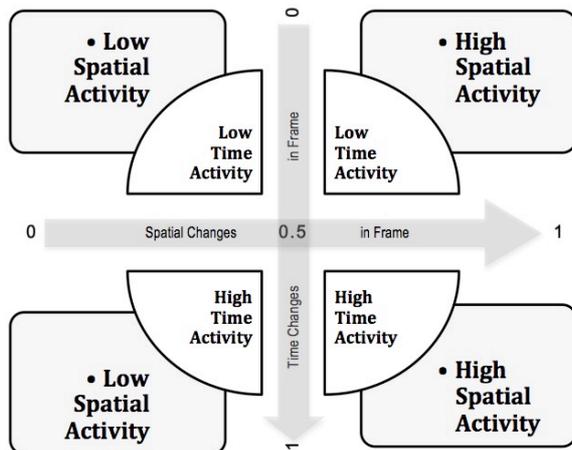


Fig. 2. Space-time diagram for video classification

1. *LSLT*: Low spatial activity – Low time activity (Top left)
2. *HSLT*: High spatial activity – Low time activity (Top Right)
3. *LSHT*: Low spatial activity – High time activity (Bottom left)
4. *HSHT*: High spatial activity – High time activity (Bottom Right)

Suffice to note here that this classification will not be effective for video sequences that are of considerable duration, because of the inherent non-homogeneous nature of their content – a consequence of their length. Video clips

are classified into different groups based on the spatial and time dynamics of change of picture elements (pixels) [11].

B. Calculation of time changes

Motion in a video clip is estimated using the Sum of Absolute Difference (SAD) indicator, which calculates the pixel-wise sum of absolute differences between two frames being compared. The formula for calculating SAD is given in (1):

$$SAD_{n,m} = \sum_{i=1}^N \sum_{j=1}^M |B_n(i, j) - B_m(i, j)| \quad (1)$$

where B_n, B_m – two $N \times M$ sized frames; i, j – pixel coordinates.

C. Calculation of spatial changes

Spatial peculiarities are calculated at the edges of block segments, as well as through the contrast and brightness between current and previous frames. Brightness is calculated as the modal difference between the average brightness value of the previous and present frames using the formula in (2).

$$Br_n = \sum_{i=1}^N \sum_{j=1}^M |Br_{av(n)}(i, j) - Br_{av(n-1)}(i, j)| \quad (2)$$

where $Br_{av(n)}$ – average brightness of n th frame with size $N \times M$; i and j – pixel coordinates.

Video sequences were grouped into three types of subjects on the basis of the calculations above [11]. The groups are as described below:

1. *Static Subject Group (SSG)*: Includes sequences with minimal observation area (e.g. face of a telepresenter) on a static background (Fig. 3);
2. *Pseudo Static Subject Group (PSSG)*: Includes video sequences with continuous and homogenic change in picture e.g. Movies (Fig. 4);
3. *Highly Dynamic Subject Group (HDSG)*: Includes video sequences, where both local and global parts of the picture undergo abrupt and heterogeneous change e.g. sporting events (Fig. 5).



Fig. 3. Examples of typical static video sequences: a) Akiyo, b) Hall



Fig. 4. Examples of typical pseudo static video sequences: a) Foreman, b) Highway.



Fig. 5. Examples of typical highly dynamic video sequences: a) Football, b) Soccer.

The subject classification adopted is based on the parameters described above. Due to the availability of a plethora of objective parameters, a statistical method was adopted for data analysis and subject classification. Each of subject type defined above is characterized by a unique set of statistical motion-related features.

D. Modeling of video streaming over a wireless network

Twelve standard test video clips of YUV format and resolution CIF (352x288) available in [12] and recommended in [13] by the ITU for carrying out test experiments were used as input test video sequences. The block diagram of the experimental setup is shown in Fig. 6. The input video sequence of YUV format is coded by an

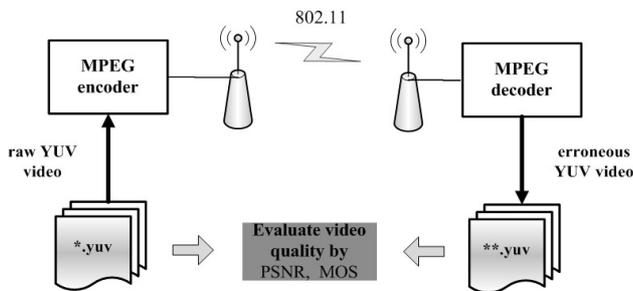


Fig. 6. Block diagram of experimental setup

MPEG-2 codec with *GOP* type *IBBPBBPBB*. Each video was coded using a different bit rate (128, 384, 768, and 1,150 kb/s).

Coding and decoding of the input video and modeling of a wireless network with random bit errors in its channel was achieved using the VCDemo software [14]. Simulation of video streaming over a wireless network was done in accordance with the OSI model on known layers: *Application*,

Transport, *Network*, *Data link* and *Physical*.

Application layer: Coding, decoding and packetization of the video stream are done at the application layer of the OSI model. The video stream is divided into variable length packets of sizes up to 1,500 Bytes, with subsequent addition of a 12-Byte RTP heading. The addition of the RTP heading ensures that the MPEG bit stream is segmented in such a way that MPEG start codes are withheld in the beginning of data packets.

Transport layer: The UDP protocol is modeled at this layer, with the addition of heading and control sum (8 Bytes). A 20-Byte IP heading is subsequently added at the network layer.

Data Link layer: The modeling of IEEE 802.11 protocol is done at this layer. The channel bandwidth is set to 20 Mb/s.

Physical layer: Simulation of random bit error (*WGN*) in the channel with BER probability taken as equal to 10^{-6} , 10^{-5} , 10^{-4} , and 10^{-3} is done at this layer.

Thus, changing of the parameters of transmitted video over wireless network was achieved both for the application layer (by changing the bit rate speed) and the network layer (by changing BER). The Peak Signal-to-Noise Ratio (PSNR) and Mean Opinion Score (MOS) were used as quality indicators [15]. Transmission of each video clip over the network was simulated 16 times with different settings.

III. RESULTS AND DISCUSSION

A large quantity of experimental data was generated in the course of the experiment reported in section 2.0 above. Average values of PSNR from different BER for varying bit rate values are presented in Fig. 7.

From Fig. 7, it is seen that by increasing the BER to 10^{-4} for

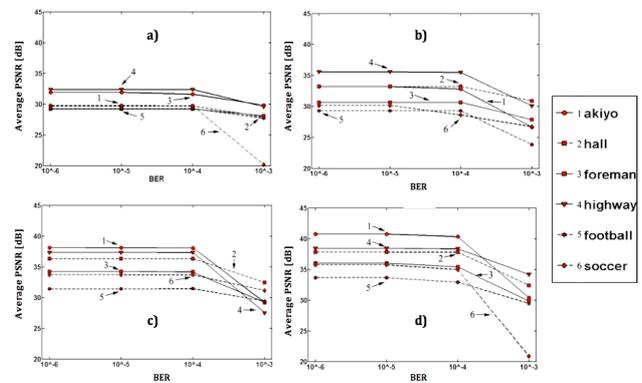


Fig. 7. Average PSNR of different Bit Error Rates for four bit rates: a) 128 kb/s; b) 384 kb/s; c) 768 kb/s and d) 1,150 kb/s

different bit rates, the average PSNR remains practically the same. This indicates that the quality of transmitted video clip remains unchanged a pointer to the fact that the decoder is able to correct errors at this BER level. For $BER=10^{-4}$, a slight change in quality is observed only for *HDSG* video sequences. However, a decrease in the average

TABLE 1. PARAMETERS USED IN SIMULATION

A	B	C	A	B	C	A	B	C	A	B	C
1	128	10^{-6}	5	384	10^{-6}	9	768	10^{-6}	13	1,150	10^{-6}
2	128	10^{-5}	6	384	10^{-5}	10	768	10^{-5}	14	1,150	10^{-5}
3	128	10^{-4}	7	384	10^{-4}	11	768	10^{-4}	15	1,150	10^{-4}
4	128	10^{-3}	8	384	10^{-3}	12	768	10^{-3}	16	1,150	10^{-3}

A – Experiment serial №; B – Bit rate in kb/s; C – BER.

value of PSNR is observed for practically all the video clips for $10^{-4} \geq BER \geq 10^{-3}$. The highest degradation of 10 dB is gotten for the *Soccer* video.

In addition to the average values, it is necessary to estimate the distribution of PSNR value for each experiment, since the average value cannot adequately depict the change in quality that occurs for different modeling conditions. The histograms of PSNR value distribution for the video clips are shown in Fig. 8 through Fig. 10 for all the 16 modeled cases. The numbers 1 to 16 on the right horizontal axis represents the serial number of the experiment with the corresponding parameters given in Table 1.

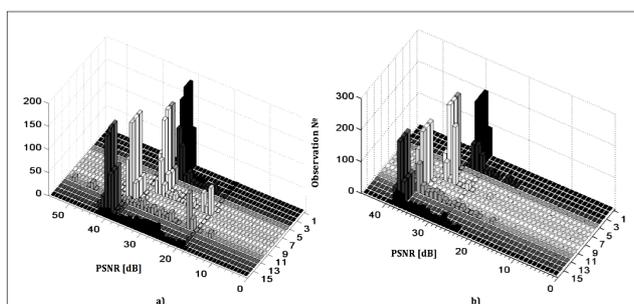


Fig. 8. PSNR distribution histograms for *SSG* video clips under various modeling conditions: a) *Akiyo*; b) *Hall*

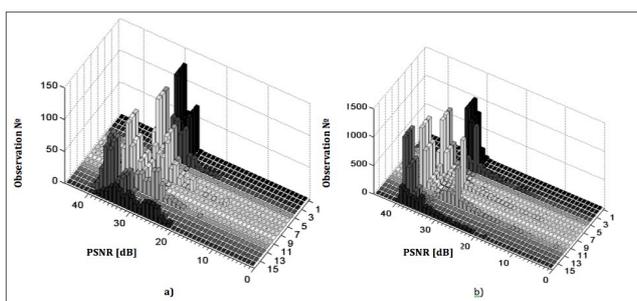


Fig. 9. PSNR distribution histograms for *PSSG* video clips under various modeling conditions: a) – *Foreman*; b) – *Highway*

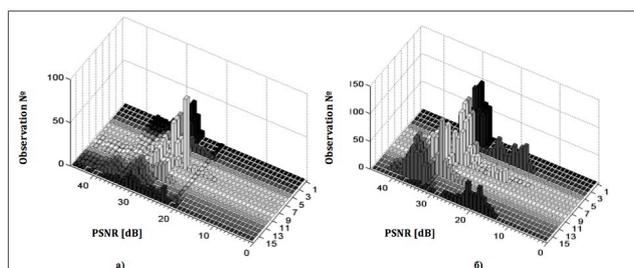


Fig. 10. PSNR distribution histograms for *HDSG* video clips under various modeling conditions: a) *Football*; b) *Soccer*

We see from Fig. 8 – Fig. 10 that for BER values of 10^{-6} and 10^{-5} , the distribution of quality indicator is practically identical and situated in the high PSNR region. For $BER = 10^{-4}$, the distribution of *HDSG* PSNR value is different when compared with those of other subject groups. In the case of $BER=10^{-3}$ for different bit rate values, the distribution is spread out and tends toward the region of low PSNR values. This indicates degradation in the quality of transmitted video sequence. The presence of low PSNR value components in the distribution indicates the existence of frames with minimum acceptable quality. For example, the worst reception quality (PSNR < 20 dB) is lowest for the *SSG* and higher for *PSSG*. The highest quantity is seen in *HDSG*. We also note that an increase in the bit rate brings about a shift in PSNR towards its region of higher value.

This is clearly observable for video sequences of the *SSG* type, where the difference between average PSNR values for 128 kb/s and 1,150 kb/s bit rates oscillates between 8 and 11 dB for $BER=10^{-6}$, $BER=10^{-5}$ and $BER=10^{-4}$. For *PSSG*, this indicator comprises from 4 to 7 dB, and 4 to 6 dB for *HDSG*. We can infer from the foregone that an increase in the bit rate has most effect on *SSG*.

The relationship between average PSNR value of video clips, Bit rate and BER is shown on Fig. 11 – Fig. 13. The general picture of changes in the average PSNR value while increasing bit rate and BER allows for drawing the following conclusions. For *SSG*, average value of PSNR for changes in bit rate from 128 kb/s to 1,150 kb/s has an upward increase from 32 dB to 41 dB (*Akiyo*) and from 30 dB to 38 dB (*Hall*). It may be noted that increasing error level to $BER=10^{-4}$ has no effect on quality. However, for $BER = 10^{-3}$ the quality of *SSG* video clips degrades considerably down to 27 dB.

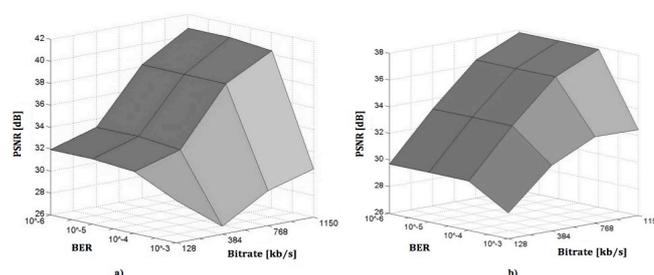


Fig. 11. Average PSNR value from *SSG* video clip Bit rate and BER values for various modeling conditions: a) – *Akiyo*; b) – *Hall*

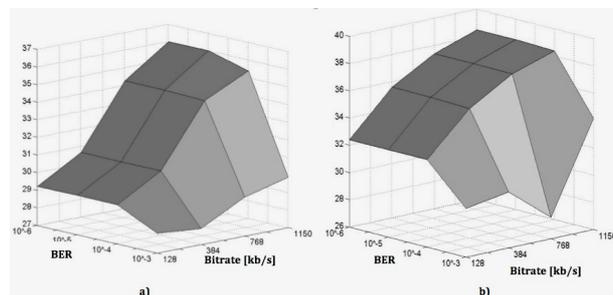


Fig. 12. Average PSNR value from *PSSG* video clip Bit rate and BER values for various modeling conditions: a) – *Foreman*; b) – *Highway*

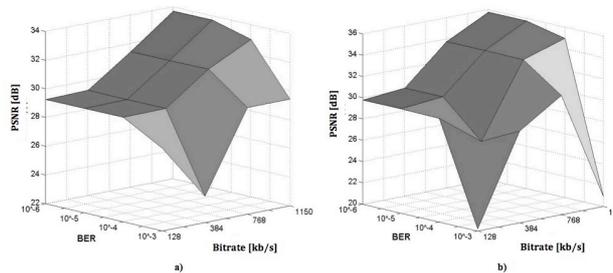


Fig. 13. Average PSNR value from video clip *HDSG* Bit rate and BER values for various modeling conditions: a) *Football*; b) *Soccer*

With an increase in bit rate from 128 kb/s to 1,150 kb/s, the average PSNR value for *HDSG* increases from 29 dB to 36 dB and from 32 dB to 38 dB for *Foreman* and *Highway* respectively. It is obvious that an increase in error to $BER=10^{-4}$ has practically no effect on signal quality for low bit rate values. For a bit rate of 1,150 kb/s, a slight degradation of 0.5 dB in quality is observed. It can be inferred that the decoder is incapable of reproducing the input video signal with moving elements under high bit rates

and BER of 10^{-4} . With an increase in error to an error bit in every 1000 bits received ($BER=10^{-3}$), the quality of *HDSG* degrades to 28 dB for all bit rate values.

For *HDSG* and with an increase in bit rate from 128 kb/s to 1,150 kb/s, the average PSNR value increases from 30 dB to 36 dB and from 29 dB to 34 dB for *Soccer* and *Football* respectively. A comparison with other subject groups reveals that a distinct degradation in quality is observable

and high ($BER=10^{-3}$) probability of error occurrence. Probability distribution for 128 kb/s and $BER=10^{-4}$ for six video clips lies in the region of 22 to 37 dB. The probability has an abrupt vertical rise for all videos, thus videos of different subject groups are practically identical. Probability distribution $BER=10^{-3}$ for six video clips lies in the region of low PSNR with values from 13 to 37 dB. For *HDSG* we observe a high probability of poor quality occurrence

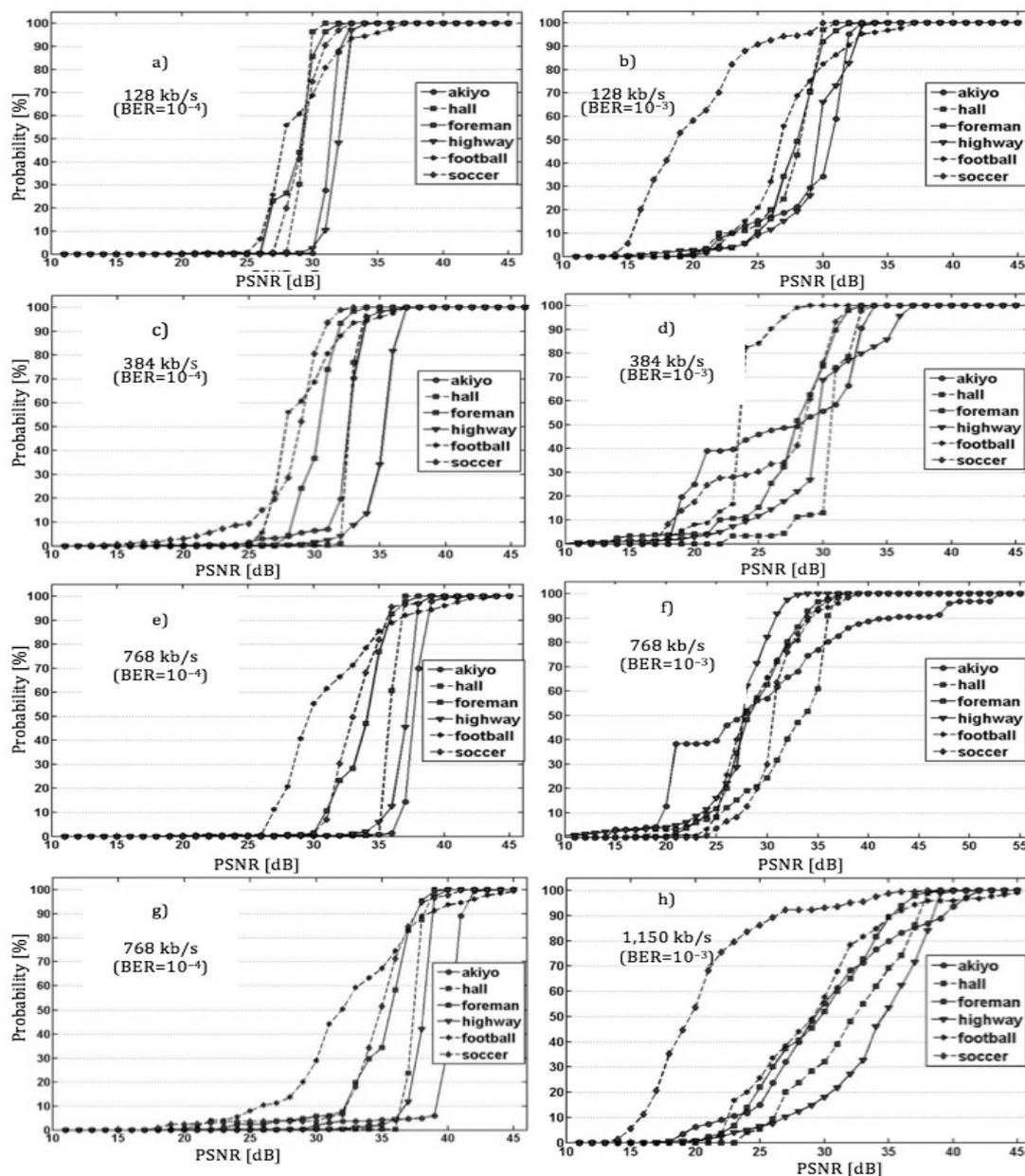


Fig. 14. Probability distribution of experimental data with different bit rates and BER: a) 128 kb/s ($BER=10^{-4}$); b) 128 kb/s ($BER=10^{-3}$); c) 384 kb/s ($BER=10^{-4}$); d) 384 kb/s ($BER=10^{-3}$); e) 768 kb/s ($BER=10^{-4}$); f) 768 kb/s ($BER=10^{-3}$); g) 1,150 kb/s ($BER=10^{-4}$); h) 1,150 kb/s ($BER=10^{-3}$).

for $BER = 10^{-4}$. This corroborates our earlier submission on the decoder's inability to reproduce the input picture with moving elements under high bit rate conditions given that $BER=10^{-4}$. With a further increase in BER to 10^{-3} , average PSNR value decreases to 20 dB for bit rates of 128 and 1,150 kb/s.

The probability distribution of experimental data for different bit rate values is shown on Fig. 14a – 14h. The distribution values for $BER=10^{-6}$, $BER=10^{-5}$, $BER=10^{-4}$ are almost identical and as such shall not be discussed. An interesting fact worth mentioning is the distribution of video clip data for different subject groups under low ($BER=10^{-4}$)

(degradation by 10 dB).

Probability distribution for 384 kb/s, $BER=10^{-4}$ for six videos lies in the region of 22 to 37 dB. And in similarity to previous bit rates, probability distribution for all videos has a vertical spiky character. For *HDSG* the distribution characteristic not unlike for 128 kb/s bit rate is less abrupt, indicating a high probability of poor quality occurrence.

Probability distribution for 768 kb/s, $BER=10^{-4}$ for six videos lies in the region of 26 to 45 dB, which attests to a rise in the quality of all video clips. For almost all the videos, probability distribution has a vertical spiky character, showing constancy in quality.

For $BER=10^{-3}$ the distribution has a slope shape, which indicates a high probability of occurrence of poor quality. The greatest change in quality is demonstrated by (*akiyo*) *SSG* video clip and (*Football*) *HDSG* video.

Probability distribution for 1,150 kb/s bit rate and $BER=10^{-4}$ for six videos lies in the region of 17 to 45 dB, which indicates a wide spread of quality. For *SSG* and *PSSG*, the distribution has a vertical spiky form in the region of high PSNR value. For *PSSG* and *HDSG*, the distribution is less abrupt, which indicates PSNR data heterogeneity tending towards low values. The probability distribution of all videos is of a slanting form, highly dispersed and lies in the region of 13 to 45 dB. The distribution has the most vertical form for *HDSG*, but lies in the low value region, which indicates the presence of a large number of poor quality frames.

IV. CONCLUSIONS

Imitation modeling showed that different video subject groups affect the perceived quality of video transmitted over wireless telecommunication network differently. Hence, a mere consideration of only the quality of transmitted (and subsequently decoded) video sequence at the application (i.e. bit rate) and network (i.e. BER) layers will not suffice, but another criterion of no mean importance must also be considered – i.e. the subject group category to which video sequence belongs.

We have shown conclusively in this paper that in transmission conditions over a network with low probability of error occurrence ($BER=10^{-6}$, $BER=10^{-5}$), *the minimum bit rate value of 128 kb/s provides acceptable video quality corresponding to an MOS > 3 for all types of video subject groups*. This incidentally answers the first objective question (*Q1*) posed in the introduction to this paper. Increasing the bit rate value further affects the *SSG* mainly, while no appreciable improvement in quality is observed for video clips with moving elements (i.e. *PSSG* and *HDSG*). A low error level ($BER=10^{-4}$) does not affect the quality of decoding *SSG* video sequences, but it affects that of *PSSG* and is very pronounced for *HDSG* videos. This observation leads us to safely conclude that the decoder is only capable of effectively correcting small errors in static pictures. Increasing the error rate to $BER=10^{-3}$ affects the decoding of video sequences of all the subject groups. The *SSG* and *PSSG* demonstrate satisfactory quality, while *HDSG* shows poor quality from a subjective assessment point of view.

Hence, the acceptable number of error bits, and consequently, the threshold from quality point of view, for which the viewer picture perception remains of acceptable quality for all types of video subject groups is **$BER=10^{-4}$** . This is an appropriate answer to the second objective question (*Q2*) raised at the onset of the paper. We have thus been able to achieve the objectives of this paper by proffering satisfactory answers to *Q1* and *Q2*, and by so doing fulfilling the aim of filling an identified gap in the literature *vis-à-vis* the effect of different video subject groups on the quality of decoded video transmitted over wireless telecommunication networks.

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