I.4. Wireless Video Delivery

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Outline

- Introduction
- Trade-off between Source & Channel Coding
- Combating Transmission Errors
- Error Resilient Techniques for Low Bit-Rate Video
- Discussions

Introduction

- Wireless communication grows rapidly, but transmitting vide over wireless networks faces several challenges
- Wireless network: high error rate, bandwidth variation and limitation, power limitation, etc.
 - Bandwidth limited: need higher compression rate
 - Error-phone: need error resilient abilities

Introduction

- The convergence of mobile and multimedia is now under way
- The goals of current second-generation cellular and cordless communications:
 Supporting integrated voice and data
- In third-generation wireless networks
 - To provide truly ubiquitous access and integrated multimedia services

Introduction

- Today's "second-generation" cellular telephony networks, such as GSM
 - Provide 10 15 kbit/s
 - Suitable for compressed speech, but too little for motion video
- Beyond the limited available bit rate, wireless multimedia transmission, offers some technique challenges:
 - The inability to provide QoS
 - How to combat transmission errors?
 - FEC
 - ARQ

Recent Mobile Networks

- The mobile network progressed from 1G(analog) to 2G (digital) and 3G, drives the mobile multimedia
- Different radio access networks (RANs) are going to be integrated into one IP network
- Today's mobile networks try to preserve a low error rate by controlling available bandwidth

 Out of control results in burst errors
- In 3GPP W-CDMA standards, link error quality in toll-quality services is generally very high
- The trend of mobile network will be heterogeneous rather than homogeneous



Radio Access	PHY	Channel Bandwidth	Modulation	Channel Coding	Error Recovery
W-CDMA [12]	64 - 384 Kbps	5 MHz	OPSK (downlink)	Convolutional	SR-ARO
			BPSK (uplink)	or Turbo Code	
ISDPA [13], [14]	64 Kbps - 14 Mbps	5 MHz	QPSK	Rate 1/3-1	Type-I HARQ with chas
	(downlink)		16QAM (downlink)	Turbo Code based	combining, Type-II HAR
DMA-2000 [12]	1.2 - 307.2 Kbps(1X)	1.25 MHz(1X)	QPSK(downlink)	Convolutional	SR-ARQ
		5 MH2(3X)	BPSK (uplink)	or Turbo Code	
CDMA-2000	Downlink peak rate:	1.25 MHz	QPSK, 8PSK,	Convolutional	Type-II HARQ
IX EV-DO [15]	1.25 - 2 Mops		DISP (useful)	or Tureo Code	
CDDF 1161	Uplink peak rate: 144 Kops	200 V II	BESK (uplink)	Completional Code	ER-ARO
OFKS [10]	171.2 Khos	200 KHz	UMSK	for CS1-4 mode	FEC
	TTTLE Rops			none for CS4 mode	100
EDGE [17]	8.8 to	200 KHz	GMSK	Convolutional Code	HARO II
	473.6 Kbps		8PSK	(CS1-4, MCS1-9)	
801.11	6-54 Mbps (11a)	20-22 MHz	OFDM (11a)	Convolutional	SW-ARQ
[18], [19], [20]	1-11 Mbps (11b)		CCK (11b)		-
	1.54 Mbrs (11e)		OFDM+PBCC(11e)		





Trade-off between Source and Channel Coding

- The classical goal of source coding :
 - To achieve the lowest possible distortion for a given target bit rate
- The classical goal of channel coding :
 - To deliver reliable information at a rate that is closed as possible to the channel capacity
- Joint source/channel coding optimization:
 - Keep the source and channel coder separate
 - But optimize their parameters jointly
 - A key problem of this is the bit allocation between the source and channel coder



Distortion Measures

- $D_{e^{\cdot}}$ the video signal distortion introduced by the source encoder
- D_{d} : the distortion at the output of the video decoder
- In practice, the most common distortion measure is mean-square error (MSE)
- The distortion at the encoder is defined as follows:

$$D_{e} = \frac{1}{XYT} \sum_{x=1}^{X} \sum_{y=1}^{Y} \sum_{t=1}^{T} (i[x, y, t] - e[x, y, t])^{2}$$

Distortion Measures (Cont.) • The MSE at the decoder is : $D_{d} = \frac{1}{XTTL} \sum_{x=1}^{X} \sum_{y=1}^{Y} \sum_{t=1}^{T} \sum_{l=1}^{L} (i[x, y, t] - d_{l}[x, y, t])^{2}$ • The distortion due to source coding is described by D_{e} • The distortion caused by transmission errors is described by ΔD $\Delta D = D_{d} - D_{e}$

Distortion Measures (Cont.)

- MSE is commonly converted to *peak signal-to-noise* ratio (PSNR)
- PSNR is defined as 10 log₁₀(255²/MSE)
- It is expressed in decibels (dB) and increases with increasing picture quality

$$PSNR_{e} = 10 \log_{10} \frac{255^{2}}{D_{e}} PSNR_{d} = 10 \log_{10} \frac{255^{2}}{D_{d}}$$
$$\Delta PSNR = PSNR_{e} - PSNR_{d} = 10 \log_{10} \frac{D_{e}}{D_{d}} = 10 \log_{10} \frac{D_{e}}{D_{e} + \Delta D}$$







Modulation

- Since we cannot feed bits to the antenna directly, an appropriate digital modulation scheme is needed
- Three basic modulation techniques
 - ASK: amplitude shift keying
 - FSK: frequency shift keying
 - PSK: phase shift keying
- The choice of a modulation scheme is a key issue in the design of a mobile communication system
 - Because each scheme possesses different performance characteristics

Burst Errors Encountered for Rayleigh Fading Channel

bit

Channel Coding & Error Control

- Two main categories of channel coding and error control:
 - FEC: Forward Error Correction
 - ARQ: Automatic Repeat Request
 - ARQ requires a feedback channel to transmit retransmission requests
 - FEC has no such requirement
 - We address interleaving as a way to enhance FEC in the presence of burst errors

Forward Error Correction

- Add redundant parity bits,
 used to detect and recover lost information
- Good for random errors with low BER, not suited for long burst errors
- Reduce effective channel bandwidth
- · Increase delay and complexity
- Adding interleaving to against for burst error
 The idea behind interleaving is to spread the error burst in time
 - Frequently used for bursty channels if the additional delay is acceptable



Automatic Repeat Request

- In contrast to FEC, ARQ requires a feedback channel
- The feedback channel conveys the status of received packets (ACK or NACK)
- Effective against burst errors and packet losses
- Cannot be used in systems that do not have a feedback channel (e.g., broadcasting)
- Generally not suited for real-time video communication over error-prone networks

IP over Wireless

- One issue when operating IP over a wireless radio link :
 - Fragmentation and reassembly of IP packet
- One way to avoid fragmentation is :
 Use the minimum packet size along the path from
- Furthermore, the overhead of IP packet headers may become prohibitive
 - neaders may become prom
 - Typically 48 bytes

Error Resilient Techniques for Low Bit-Rate Video

- Input format and rate control:
 - QCIF (176 x 144 pixels)
 - Frame rate -> 12.5 frames/s
- Error detection and resynchronization:
 - With FEC, errors can often be detected
 - A more difficult problem than error detection is resynchronization after a detected error
 - The common solution is to insert unique
 - synchronization codewords into the bitstream
 - ex. H.263 : a 17-bit sync word is "00000000000000001"

Error Detection & Resync

- Data dependencies across slice boundary should be removed to contain the errors within a slice
- may be inserted at various locations to contain the errors to a small spatial region











SP/SI synchronization/switching pictures
 Allow switch video to low bitrate data seamlessly

Error Concealment

- · Previous frame concealment
 - The corrupted image is replaced by corresponding pixels from the previous frame
 - Good results with little motion
- Encoded motion concealment
 - The MV used at the encoder are also used at the decoder for motion-compensated concealment
 - Can't be implemented in practice, because errorfree transmission of MV assumption





Leaky Prediction

- A well-known technique for increasing the robustness of DPCM systems
- · Attenuating the energy of the prediction signal
- Reduce coding efficiency
- Increase error resilience
- More advantageous than for intra coding – Increased flexibility in the design



H.261 Payload Specific Header

- A reliable method to stop interframe error propagation is the regular insertion of I-frames
- there is a trade-off to be considered:
- An increased percentage helps to reduce interframe error propagation.
- coding efficiency is reduced at the same time.



Error Confinement Interframe error propagation not only results in a temporal but also spatial error spreading Confine the error to a well-defined subregion of the frame The *independent segment decoding mode* (ISD mode): Decribed in Annex R of H.263 Each GOB is encoded as an individual subvideo Reduces the efficiency of motion compensation

 To reduce the loss of coding efficiency, combined with the unrestricted motion vector mode (UMV mode)













Reference Picture Selection (Cont.)

- The advantage of the RPS mode over simply switching to intra mode is the increased coding efficiency
- · Increasing round-trip time
- Error propagation is avoid entirely

 Since only error-free pictures are used for prediction
- Increased storage requirements
 - Additional frame buffers



Future Directions

- · Layered or scalable coding:
 - Can be considered as one of data partition methods, can be used as resilient tools with unequal protection
- · Packet scheduling
 - Use source coding information, schedule the packet by its dependency and importance
 - One of the cross layer methods
- Transmission power optimization
 - Power consumption is import in mobile device
 - Manage power optimization with R-D optimization



Estimation of Packet-Level Loss-Impact

- An offline process at the pre-encoding stage
- The estimation is performed for I/P-frames in a GOP since the error propagation is constrained within a GOP
- Evaluate the loss impact value of each slice (packet)
 Mark slice *m* of frame *n* as a lost packet and decode it with zero MV concealment
 - Compute and sum up all the degraded PSNR values in the GOP containing the simulated lost slice

$$Impact_{n}^{m} = \sum_{j=n}^{N_{\text{GOP}}} \left(PSNR_{j} - PSNR_{j}^{\text{EC}} \right)$$





Performance Evaluation

Foreman

PLR	Error-Free	Without ARQ	Proposed Method	EDF
5%	34.11 dB	29.14 dB	30.71 dB	29.28 dB
10%	34.11 dB	27.34 dB	29.00 dB	27.96 dB

· Coastguard

PLR	Error-Free	Without ARQ	Proposed Method	EDF
5%	32.65 dB	29.09 dB	30.77 dB	28.76 dB
10%	32.65 dB	26.68 dB	27.91 dB	27.39 dB



Performance Evaluation - Coastguard (PLR = 5%) -EDF_Drop - Clean Loss -Priorty_Drop = 36 31 NNSA 26 21 16 31 61 91 121 151 181 211 241 271 Frame Num

Conclusions

- Introduce the DDF as a tools for comparing wireless video systems
- Two major problem with wireless video
 - Only low bit rates are available
 - Pass loss and multipath fading cause time-variant error rates
- In designing the digital transmission system
 - Trade-off among throughput, reliability, and delay
- The error resilient techniques fall into two major categories
 - Techniques that reduce the amount of introduced errors (e.g., resync, error concealment)
 - Techniques that limit interframe error propagation (e.g., leaky prediction, intra update)