

Lecture 11: Media Adaptation

Scalable Coding, Dealing with Errors

Some slides, images were from http://ip.hhi.de/imagecom_G1/savce/index.htm
and John G. Apostolopoulos http://www.mit.edu/~6.344/Spring2004/video_streaming2_2004.pdf

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Recommended Readings

- The Scalable Video Coding Amendment of the H.264/AVC Standard
 - http://ip.hhi.de/imagecom_G1/savce/index.htm
- Video Communication and Video Streaming II: Error Resilient Video Coding, by John G. Apostolopoulos
 - http://www.mit.edu/~6.344/Spring2004/video_streaming2_2004.pdf

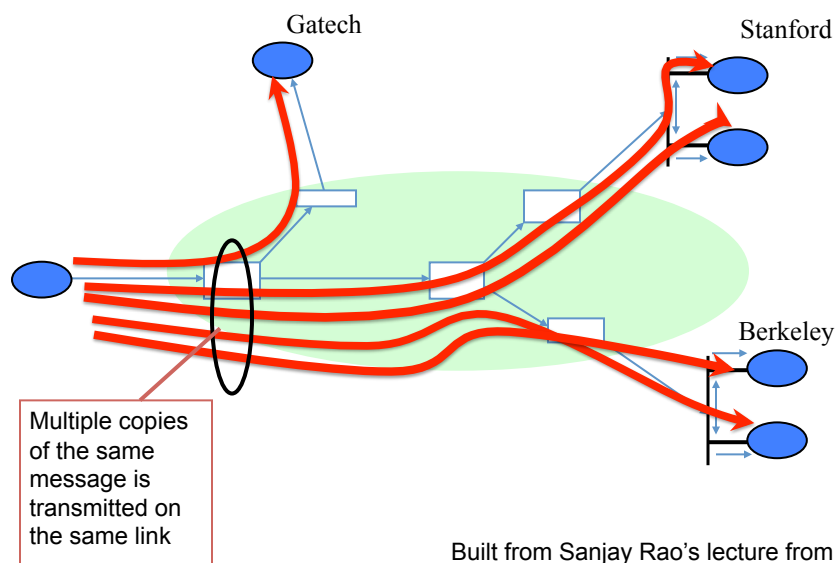
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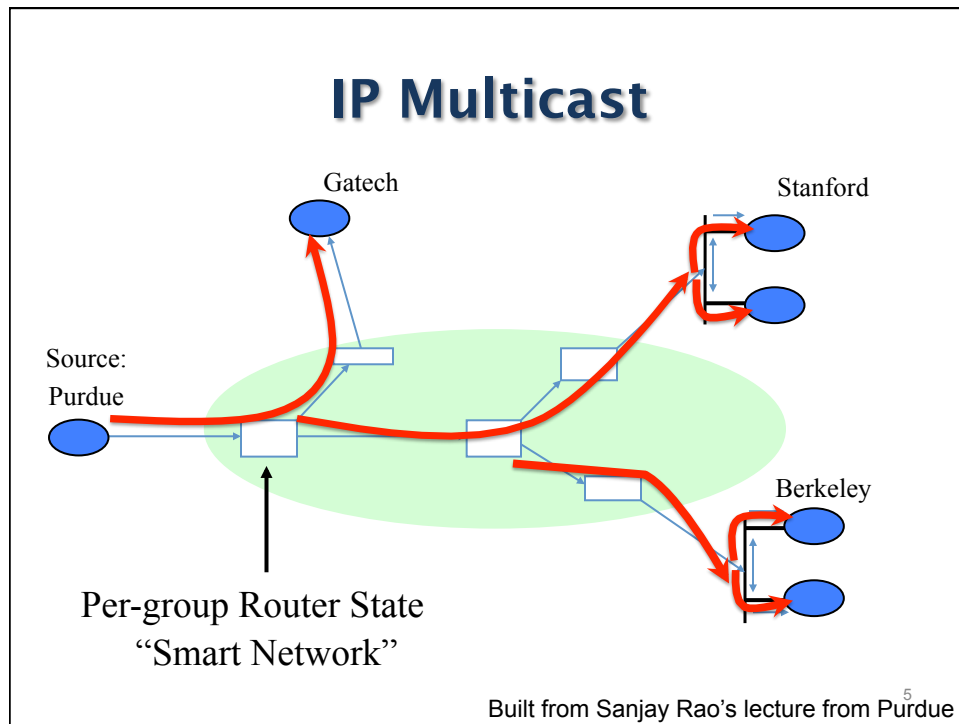
Previously

- Ways to distribute video online
 - Client-server
 - IP Multicast
 - Content delivery networks
 - P2P media streaming design
- YouTube Study

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Client-Server



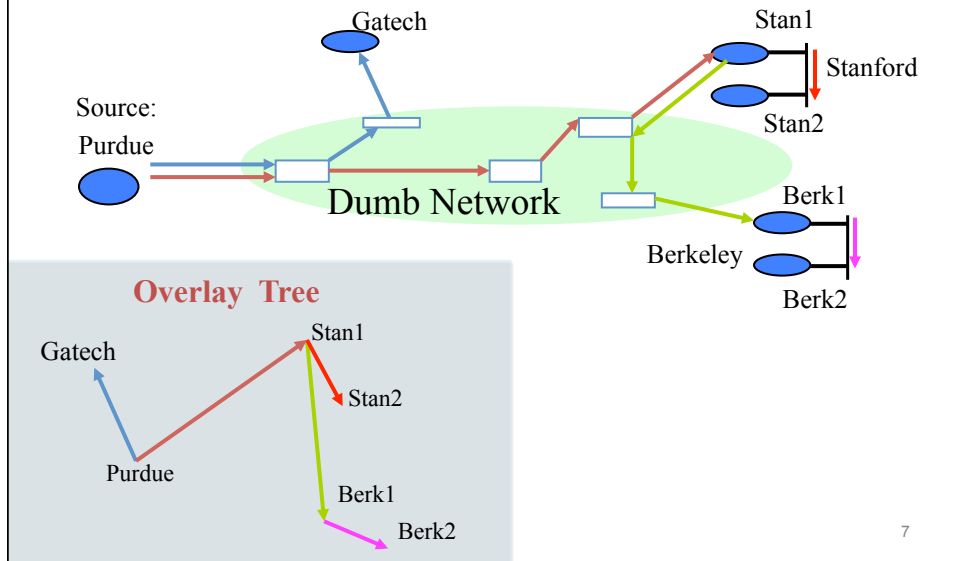


Content Delivery Networks

- Strategically located **replicas** **unicast** content to nearby clients
- Clients can access the content on the replica servers nearest to them or the one with less current load.
 - Reduces burden on primary server
 - Improves perceived performance at client



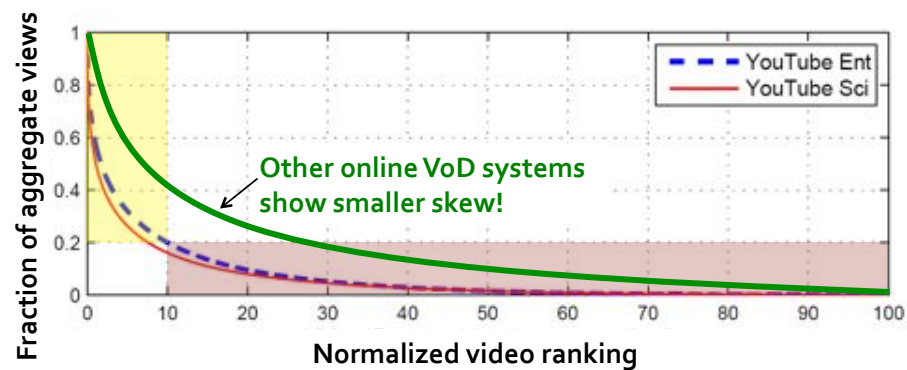
P2P Multicast



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Pareto Principle

■ 10% popular videos account for 80% total views



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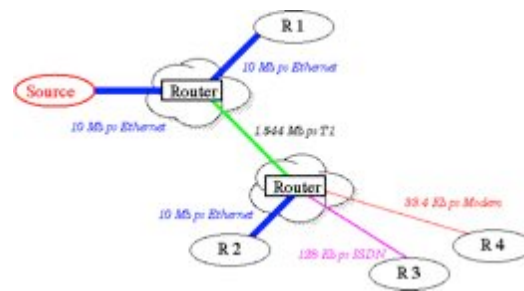
TODAY: WHAT CAN MEDIA DO TO COPE WITH NETWORK ARTIFACTS

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Network Artifacts

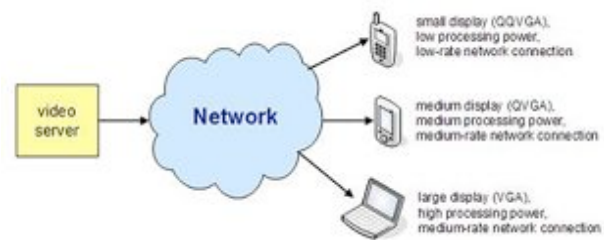
- Network congestion →
 - Time-varying network bandwidth
 - Multiple users (IP multicast) with different bandwidth
 - Packet losses
- Transmission errors →
 - Corrupted bits
 - Common in wireless

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Artifact #1 Time-varying Network Bandwidth

Artifact #2 Heterogeneous Receivers



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The Concept of Media Scaling

- Dynamically adapt the media encoding to accommodate changes in available network bandwidth
 - Not that easy!
- **Need to address the following:**
 - Estimate the bandwidth
 - Figure out what can be adjusted
 - How to adapt encoding, and encoding rate
 - What if both audio and video are present? How to partition bandwidth between the two?
 - What if the media is pre-encoded, i.e. MPEG-1.
 - Decode and then re-encode?

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Estimating Bandwidth

- Requires feedback.
 - How often?
 - In what form?
 - Recall: RTP + RTCP
- TCP-friendliness
 - Current thought: act like TCP.
 - Padhye's equation:

$$B_{TCP} = \frac{M}{t_{RTT} \sqrt{\frac{2Dl}{3}} + t_{out} \min\left(1, 3\sqrt{\frac{3Dl}{8}}\right) l(1 + 32l^2)}$$

M = packet size

l = loss fraction

D = number of packets ACK'ed
by one ACK

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Feedback

- How often?
 - Want to limit this to once per RTT.
 - Otherwise, rate of feedback is a design decision.
 - Tradeoff: responsiveness vs. feedback bandwidth
 - Should be related to when the source can use the info.

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Feedback

- In what form?
 - Media independent information:
 - Loss rate and delay
 - RTCP reports from RTCP layer
 - Media specific information:
 - Quality of reconstructed media.
 - Not necessarily well related to loss and delay.
 - User preferences
 - For example: Regions of interest.
 - Very application dependent.

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Scaling Dimensions

- Different media can be scaled along different dimensions.
- Two major considerations:
 - Encoding support.
 - The media representation often restricts the range and type of scalability.
 - Perceptual quality.
 - Non-linear subjective response to changes.
 - Think of frame size changes.

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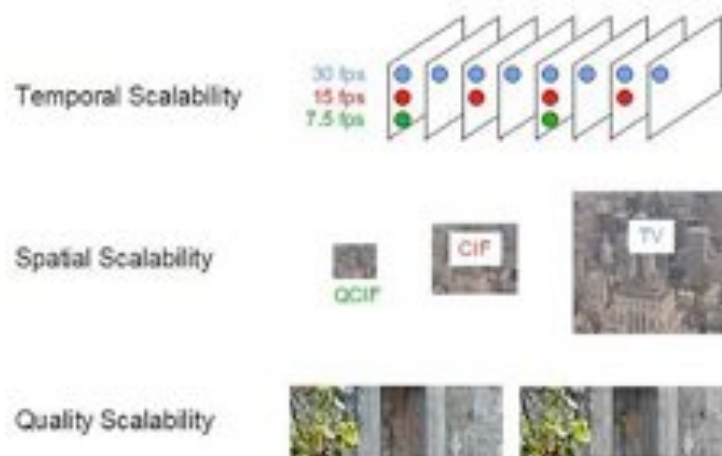
SVC: Scalable Video Coding

- **Temporal scaling**
 - Reduce the resolution of the stream by reducing the frame rate
- **Spatial scaling**
 - Reduce the number of pixels in an image
- **Frequency scaling** Quality Scaling
 - Reduce the number of DCT coefficients used in compression
- **Amplitude scaling**
 - Reduce the color depth of each pixel in the image
- **Color space scaling**
 - Reduce the number of colors available for displaying the image

An excellent read! http://ip.hhi.de/imagecom_G1/savce/index.htm

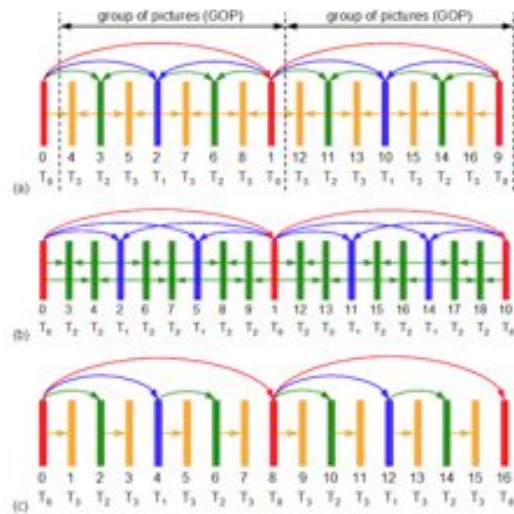
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SVC (2)



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Temporal Scalability

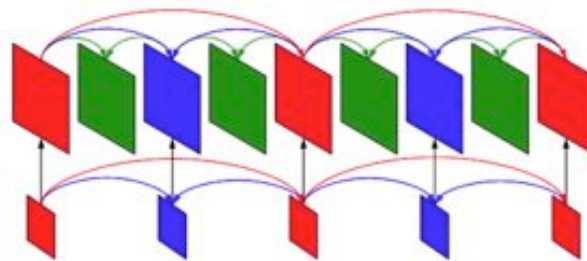


Easy implementation:
choose reference frames
for motion compensation

Minimum coding/decoding delay

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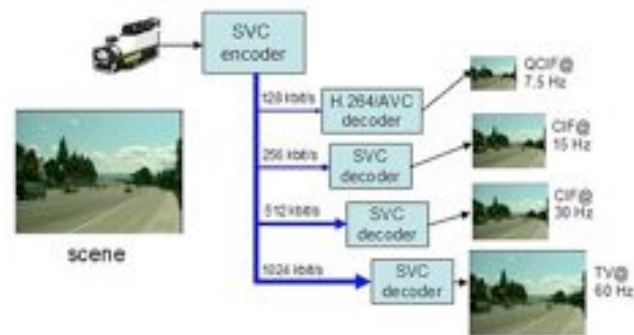
Spatial Scalability



Multi-layer structure with additional inter-layer prediction (black arrows).

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The Result



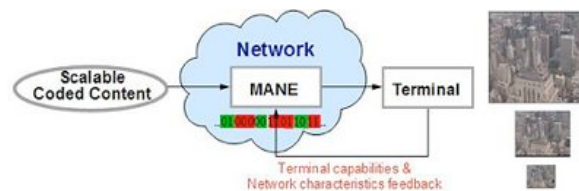
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**Generalizing Scalability →
Layered coding**

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Layered Representations

- An encoding specifically designed to produce multiple representations.
- Characteristics:
 - **Additive**: the more layers you get, the better the media
 - **Efficient**: the sum of the layers is only slightly greater than the best rep. at that quality.



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Layered Solution

- Use a layered representation.
- Receivers decide
 - Layers added and dropped to adjust to appropriate target rate.



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Layering Mechanisms

- Strictly Additive Layering
- Independent Layering
 - Multi-description Coding

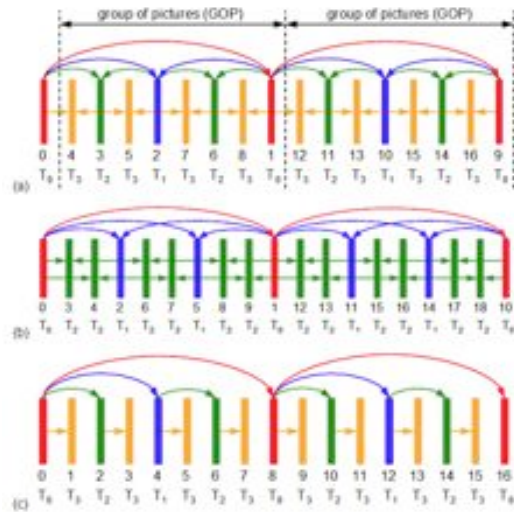
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Strictly Additive Layering

- Layering split into:
 - Base Layer
 - Enhancement Layers
- Each layer depends on all data in lower layers.
- Advantages: increased compression
- Disadvantages: packet loss in lower layers makes packets in higher layers useless.

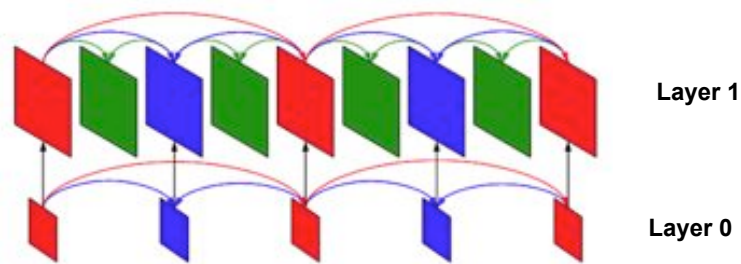
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Examples: Temporal Scalability



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Examples: Spatial Scalability



Multi-layer structure with additional inter-layer prediction (black arrows).

Need T_i to decode T_j if $i < j$

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Independent Layering

- Every packet from every layer improves quality.
- No ordering or dependency between layers.
- Advantages: Good ADU properties
- Disadvantages:
 - Hard to construct
 - Compression suffers
- Most layering schemes fall between these two extremes in some hybrid fashion.

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Examples

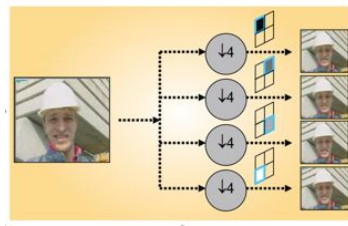
- Temporal layering
 - Video: Odd frames vs. Even frames.
 - Mostly independent layers.
 - Packet loss affects smoothness.
 - Audio: Multiple, offset, lower-rate streams.
 - Inter-sample compression compromised.
 - Not all combinations equally pleasing.
- Spatial layering
 - Each layer improves video size/resolution.
 - Many of the issues as in temporal.

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Examples

- Multiple description coding (MDC)

- Partition a single media stream into n -substreams ($n \geq 2$) referred to as descriptions.
- Packets of each description are routed over multiple, (partially) disjoint paths.
- In order to decode the media stream, any description can be used
- The quality improves with the number of descriptions received in parallel.



http://en.wikipedia.org/wiki/Multiple_description_coding

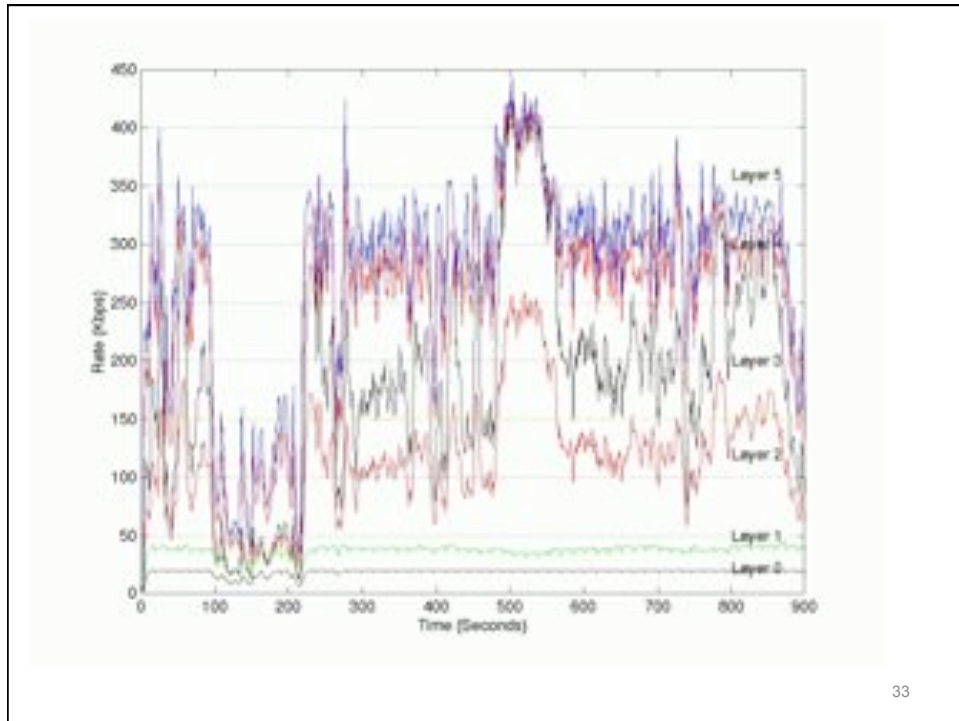
V. K. Goyal, "Multiple Description Coding: Compression Meets the Network," IEEE Signal Processing Magazine, vol. 18, no. 5, pp. 74–94, Sept. 2001.

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Question

- SNR layering
 - Layers contain different DCT coefficient ranges.
 - DC and first few AC
 - Low AC
 - Middle AC
 - High AC
 - **Is this an independent layering?**

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Layered Video Codec Example: LayerStream

- video stream → a base layer + multiple enhancement layers
- The encoder == a lightweight transcoder
 - Takes high quality video as input
 - Transcodes it to a layered stream in real-time.
 - The layers are subsequently packetized and sent to the network
- The decoder : packets are reassembled by the depacketizer and transcoded back into a valid MPEG stream.
- Some enhancement layers may have been dropped by the network due to congestion.
 - This causes the signal-to-noise ratio to decrease.
 - But the client will never experience freezes or artifacts, as real-time delivery of the lower layers is guaranteed by the network at all times.

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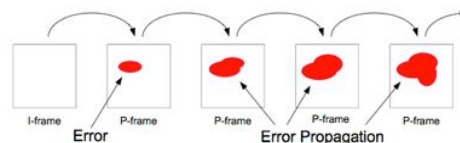
Artifact #3 Packet Losses + Corruptions

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Video is vulnerable to losses

Assuming conventional MPEG-like system: MC-prediction, Block-DCT, run length and Huffman coding

- Losses create two types of problems
 - **Loss of bit stream synchronization:** Decoder does not know what bits correspond to what parameters
 - E.g. error in Huffman codeword
 - **Incorrect state and error propagation:** Decoder's state is different from encoder's, leading to incorrect predictions and error propagation
 - E.g. error in MC-prediction or DC-coefficient prediction



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Measurement-based Study

Effects of Packet Loss on MPEG

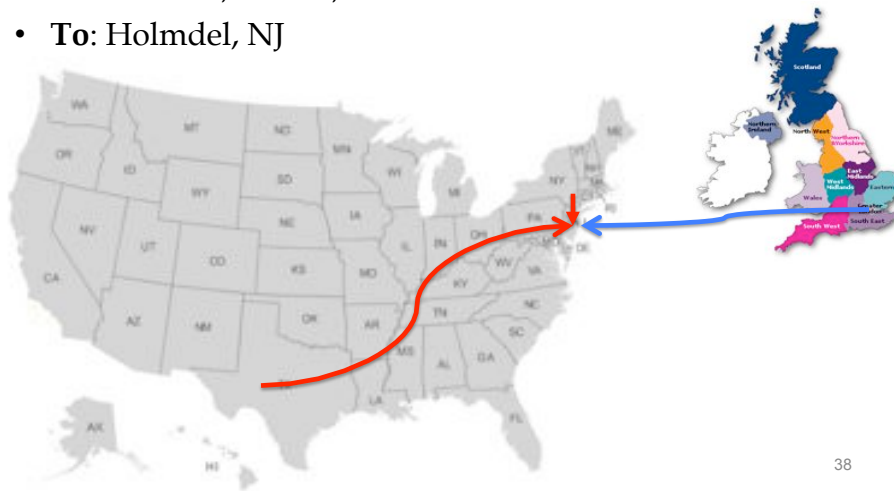
Jill Boyce and Robert Gaglianella
ACM Multimedia 1998

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Measurements

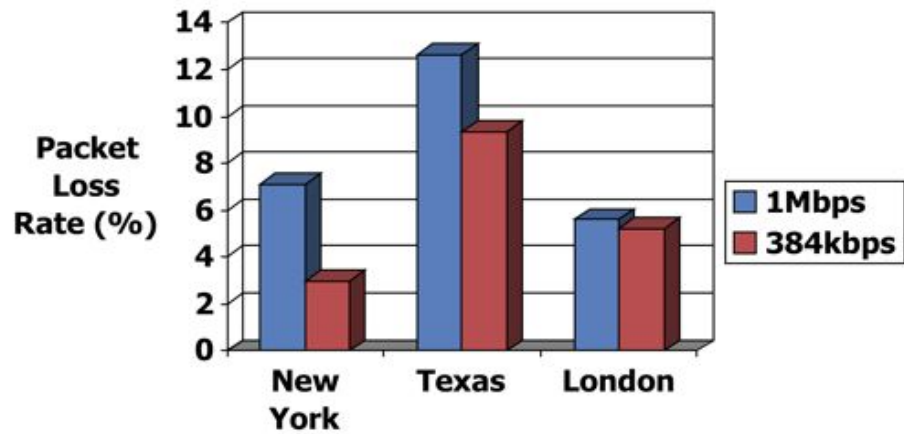
Two 5-mins MPEG videos
30 fps
384 kbps and 1 Mbps
IBBPBBPBBPBBPBB
QSIF 176x112 and SIF 352x240
One row per slice

- **From:** NYC , Austin, London
- **To:** Holmdel, NJ



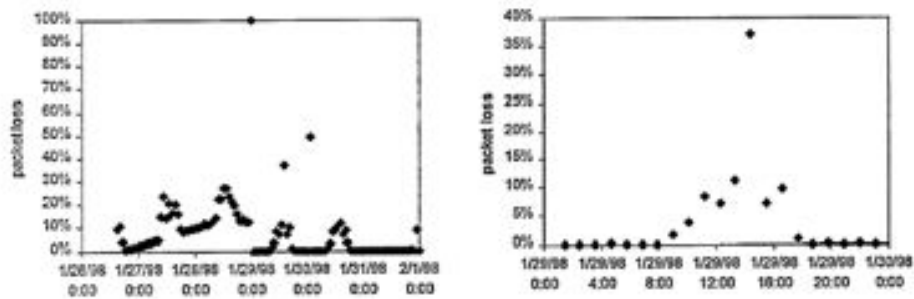
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Average Packet Loss



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Packet Loss vs Time



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How to Fill Packets

- **B**: 1 frame → 1 packet
- **P**: 1 slice → 1 packet
- **I**: 1 slice → 1 packet

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Packet Size Distributions

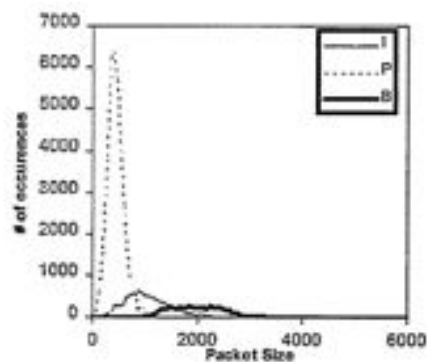


Figure 5. 1Mbps Packet size distribution

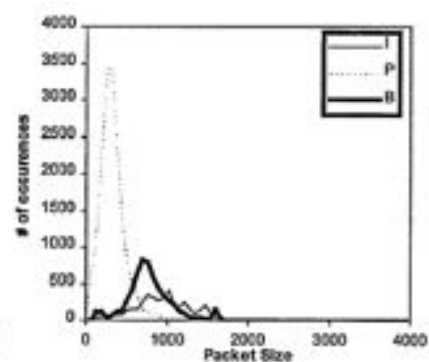


Figure 6. 384 kbps Packet size distribution

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Packet Size vs Loss Rate

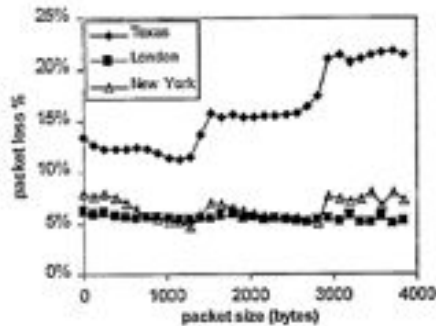


Figure 7. 1 Mbps Packet size vs. loss

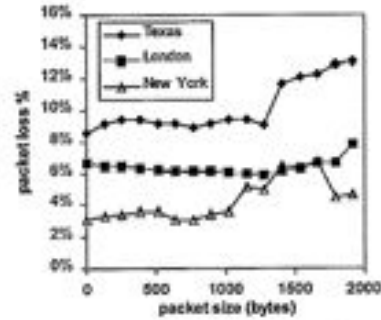


Figure 8. 384 kbps Packet size vs. loss

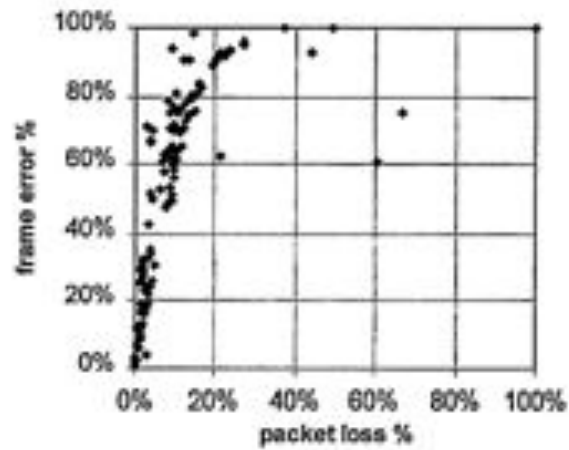
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Lesson

- 1 slice per packet even if larger than MTU
- If smaller packets means higher loss rate:
fill packet until MTU
 - Else one slice per MTU
- Alt: Change size of slice

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Frame Effectuated by Errors



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Correlation of Packet Loss

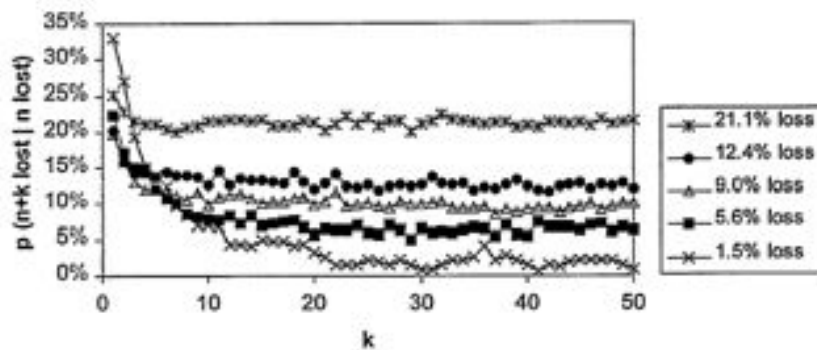


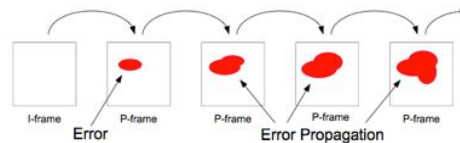
Figure 12. Conditional Probabilities of Packet Loss

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Video is vulnerable to losses

Assuming conventional MPEG-like system: MC-prediction, Block-DCT, run length and Huffman coding

- Losses create two types of problems
 - Loss of bit stream synchronization: Decoder does not know what bits correspond to what parameters
 - E.g. error in Huffman codeword
 - Incorrect state and error propagation: Decoder's state is different from encoder's, leading to incorrect predictions and error propagation
 - E.g. error in MC-prediction or DC-coefficient prediction



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Some Ways to Deal with Packet Losses

- Network-driven (done by the sender)
 - Retransmission
 - Happen infrequently
 - FEC
 - Unequal error protection + layered coding
 - Multiple-description coding
- **Error Concealment (done by the receiver)**
- **Limiting Error Propagation (done by both)**
 - **Reference Frame Selection**

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Yes We Can Do This

http://www.vidyo.com/resources/videos.php#vidyo_vs_legacy

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Decoding: Reference Frame Selection

- If a I-frame gets lost, how to decode subsequent P, B-frames?
 - Select the next I-frame and use it as the reference I-frame

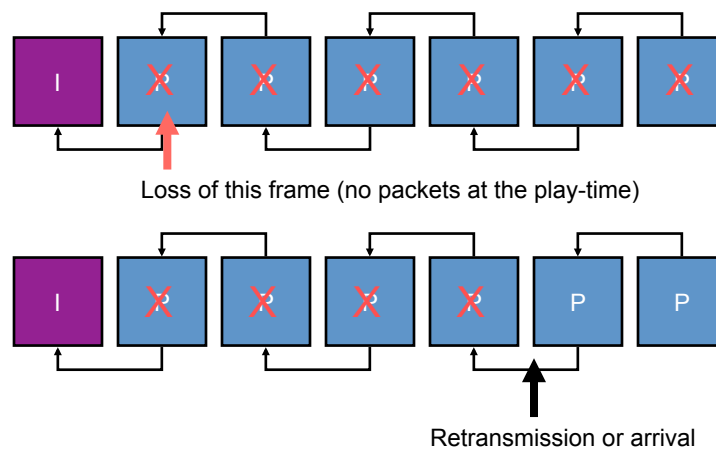
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Decoding: Use Late Packets

- Packets can be dropped by routers due to congestion
- They can also be dropped by receivers because they arrive late

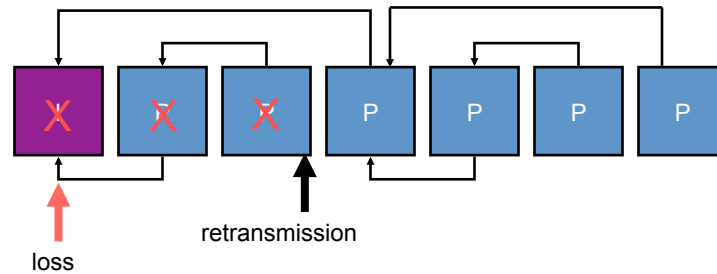
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Illustrative Examples



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Encoding: Changing Temporal Structure



Need to protect I-frame strongly, adding more FEC or allow retransmissions

Another solution: use all I-frames? No error propagation but BAD compression

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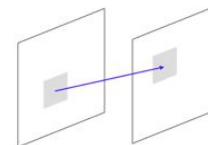
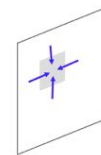
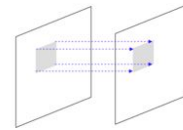
Decoding: Error Concealment for Videos

- Estimate the lost information in order to conceal the fact that an error has occurred
- Performed at the decoder
- Possible because:
 - Video exhibits a significant amount of correlation along the spatial and temporal dimensions
- Basic approach:
 - Perform some form of spatial/temporal interpolation to estimate the lost information from correctly received data

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Error Concealment for Video

- **Repeat** pixels from previous frame
 - Effective when there is no motion, potential problems when there is motion
- **Interpolate** pixels from neighboring region
 - Correctly recovering missing pixels is extremely difficult, however even correctly estimating the DC (average) value is very helpful
- **Interpolate** motion vectors from previous frame
 - Can use coded motion vector, neighboring motion vector, or compute new motion vector



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Impact of Error Concealment



Figure 1
On the left, errors are not concealed. On the right, state-of-the-art concealment has been applied

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An excellent tutorial

- Video Communication and Video Streaming II: Error Resilient Video Coding by John G. Apostolopoulos

http://www.mit.edu/~6.344/Spring2004/video_streaming2_2004.pdf

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Summary of Today's Learning

- Media coding meets networking
 - How can media address various networking artifacts
- Scalable, error-resilient coding
 - Layered coding (dependent vs. independent layering)
 - Reference frame selection
- Error recovery/concealment

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