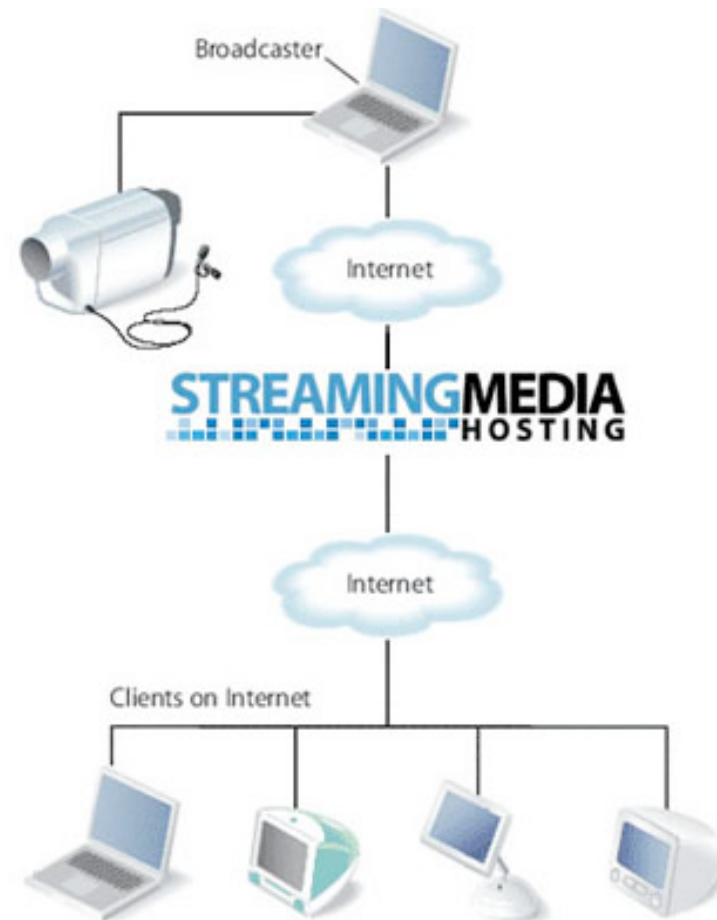


Streaming media

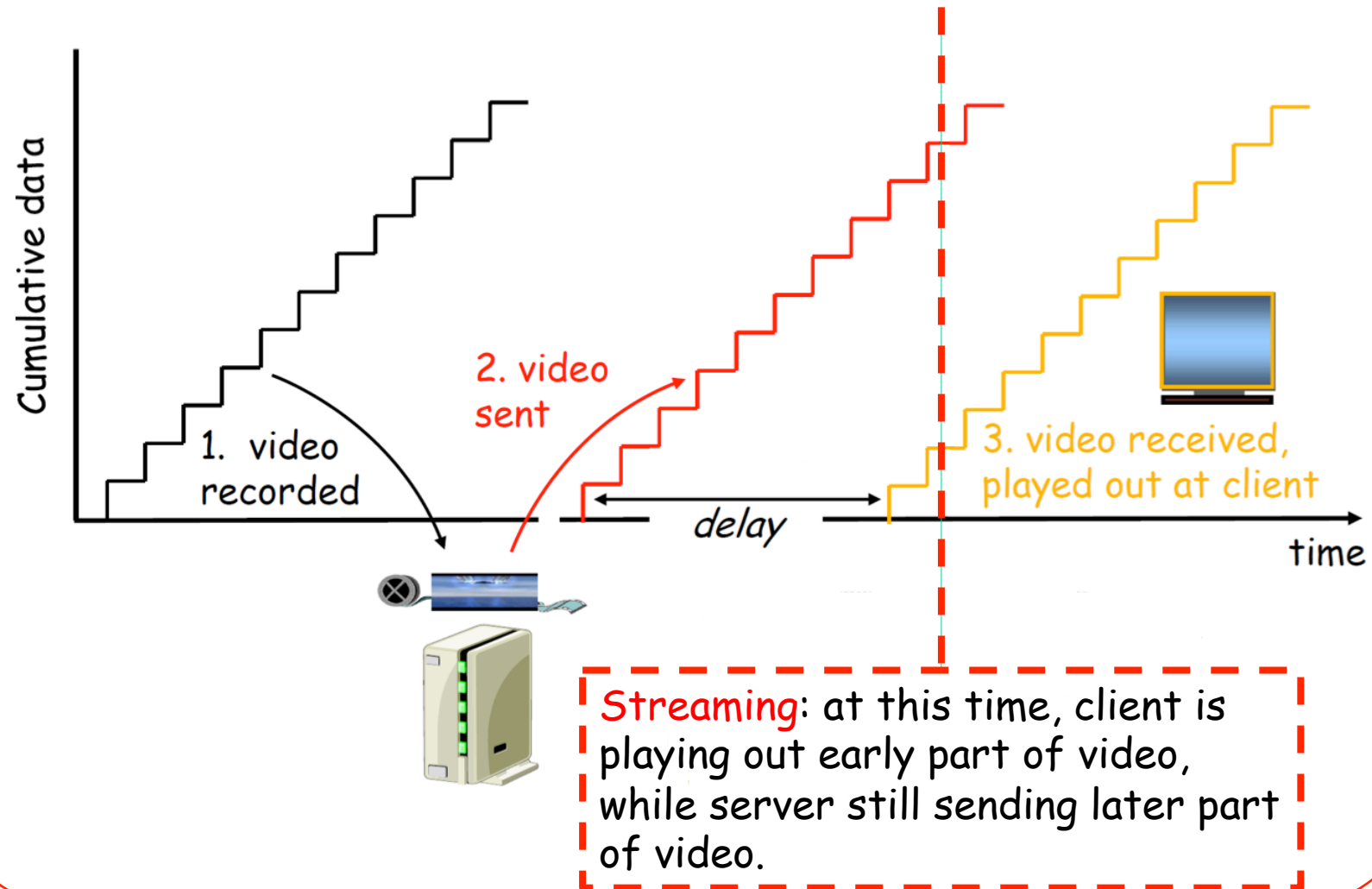


Multimedia applications

- Streaming stored audio and video
 - Jitter
- Streaming of live audio and video
- Interactive audio and video
- Delay sensitive and loss tolerant
- Best effort delivery



Streaming stored multimedia



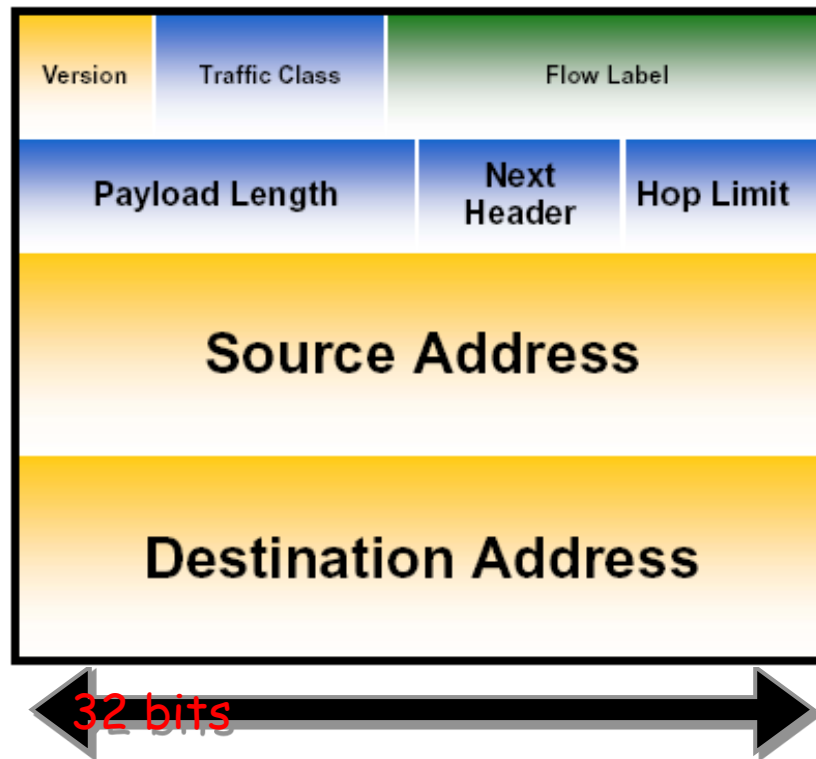
How should the Internet better support multimedia?

- **Laissez-faire**
 - no major changes
 - more bandwidth from ISP when needed
 - content distribution (CDN), application-layer multicast
 - application layer
 - Overlay Networks
- **Differentiated services philosophy:**
 - Fewer changes to Internet infrastructure, yet provide 1st and 2nd class service.
- **Integrated services philosophy**
 - Fundamental changes in Internet so that apps can reserve end-to-end bandwidth
 - Requires new, complex software in hosts & routers

IPv6 packet

Traffic class: allocate different priority to packets based on applications

Flow Label: identify datagrams in same "flow" (concept of "flow" not well defined)



Checksum has been removed entirely to reduce processing at each hop

Quick exercise

- The lack of _____ is the **most common factor** that impairs streaming media quality.
 - A. Network delay
 - B. Network bandwidth
 - C. Network security
 - D. Network administration

Implementing best-effort service

- Some problems to be solved.
 - Limited **bandwidth** compared to requirements.
 - **360p** -> **720p** -> **1080p**
 - Random network delays
 - Packet loss

Compression - reduce bandwidth use

- Digitization -> conversion to bits.
 - = "Encoding"
- Consider problem of encoding an image:
 - 1024x1024 pixels in size.
 - 24 bits per pixel.
 - 3MB per image.
- Consider a video (frame = image):
 - 24 frames per second (at least).
 - 24 frames * 3 Mbytes = 72 MB per second.
 - 72 Mbytes per second = 576 Mbps.
 - 802.11g gives you a maximum of 54 Mbps
- We need compression!!

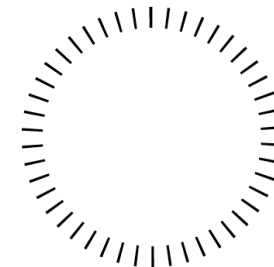
Compression examples

- Audio
 - Encoding techniques
 - Telephone (pulse code modulation) : 64 Kbps
 - CD (pulse code modulation) : 1.411 Mbps for stereo
 - Compression techniques
 - G.729 : 8 Kbps
 - MP3 : 112 or 128 Kbps for CD quality
- Video
 - MPEG2 : 3-6 Mbps for DVD quality
 - MPEG4: high-definition video media like Blu-ray Disc

<http://www.dr-lex.be/info-stuff/videocalc.html>

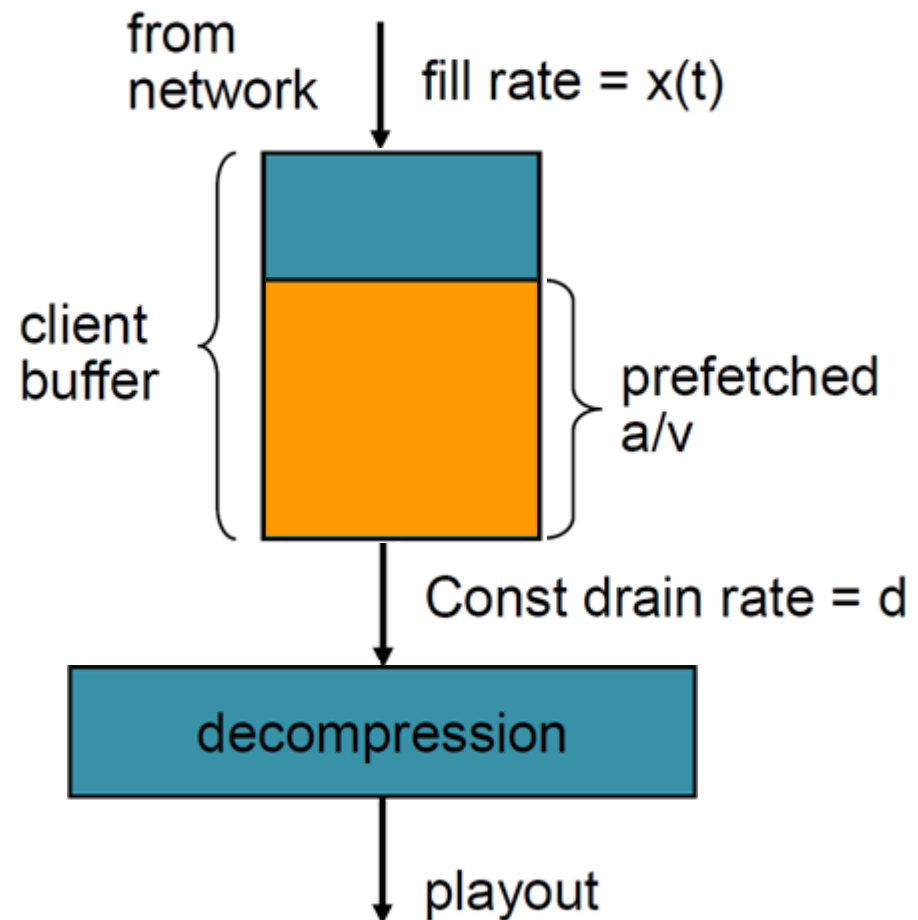
Quick exercise

- Which of the following statements are correct?
 - A. MPEG4 supports lossless video compression
 - B. MPEG4 supports lossy video compression
 - C. MPEG4 supports lossless audio compression
 - D. MPEG4 supports lossy audio compression

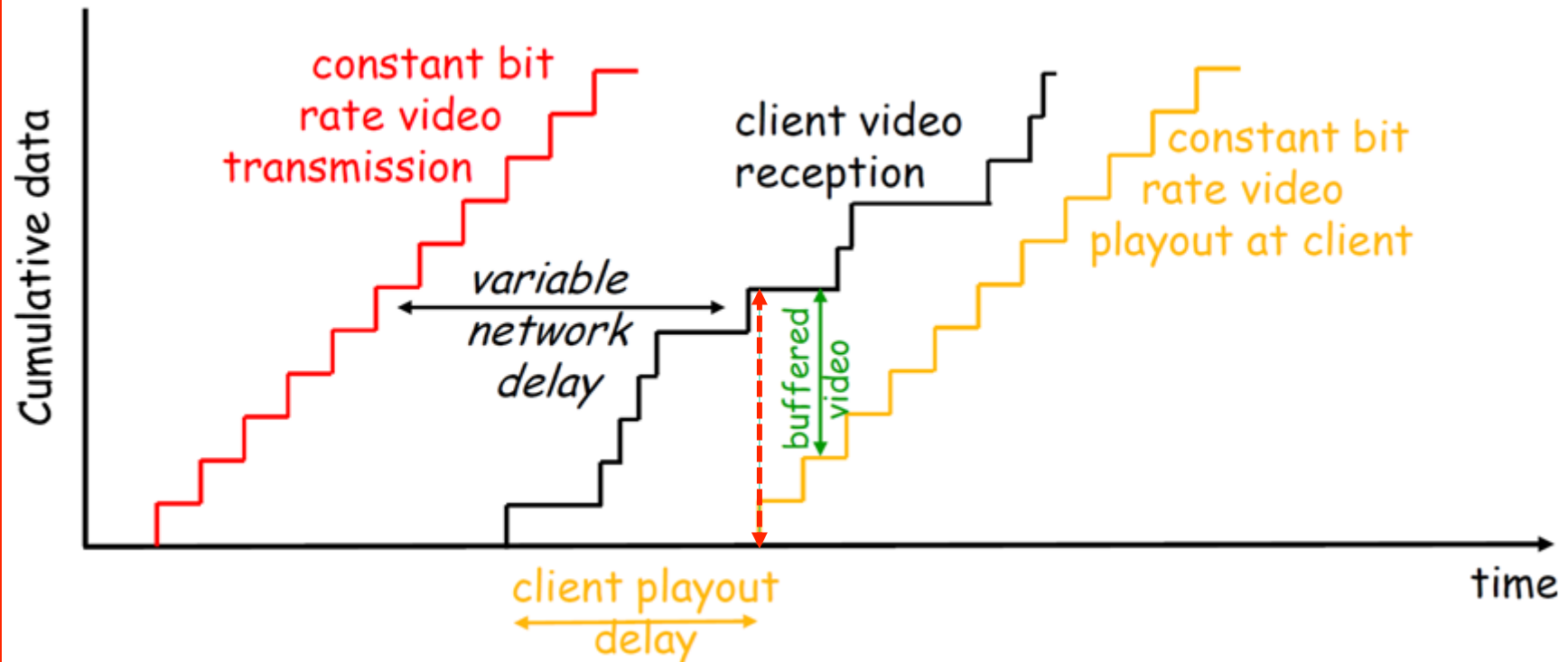


Client Buffering

- Hide jitter by buffering arriving packets
- We talk about **fill rates** and **playout rates**.
- Also supports **decompression**, can process groups of packets.



Effect of Client Buffering



- Client-side buffering, play-out delay compensate for network-added delay, delay jitter

Quick exercise

- Which of the following activities do NOT rely on the playback buffer?
 - A. Reordering incoming chunks of multimedia content
 - B. Decompression
 - C. Cancel the playback process
 - D. Temporarily halt the playback of audio or video clip

Causes of lost packets

- Use of UDP (best-effort).
- TCP would avoid this, but too slow
- Besides humans can tolerate 1-10% loss!
- So aim is not no loss but to cope with some loss.
 - *Conceal packet loss*

Concealing packet loss

- *Forward error correction*
- Principle:
Add redundant information (but not too much)
- Simple XOR FEC scheme (1990):
 - Break transmission into groups of **n packets**
 - Every **n-th packet**, send a special **n+1 packet**
 - **n+1 packet** = previous packet XORed together
 - Should receiver lose any of the packets in the group, receiver can recover the missing packet by XORing the special packet with what has been received

Concealing Packet Loss

What is sent

Packet 1	0	1	1	1	0	1	0	1
Packet 2	0	0	1	1	0	0	0	0
XOR packet	0	1	0	0	0	1	0	1

What is received

Packet 1	0	1	1	1	0	1	0	1
Packet 2	LOST!!!							
XOR packet	0	1	0	0	0	1	0	1

Reconstructing the lost packet

Packet 1	0	1	1	1	0	1	0	1
XOR packet	0	1	0	0	0	1	0	1
Recovered packet 2	0	0	1	1	0	0	0	0

Concealing Packet Loss

What is sent

Packet 1	0	1	1	1	0	1	0	1
Packet 2	0	0	1	1	0	0	0	0
XOR packet	0	1	0	0	0	1	0	1

Simple scheme but problems:

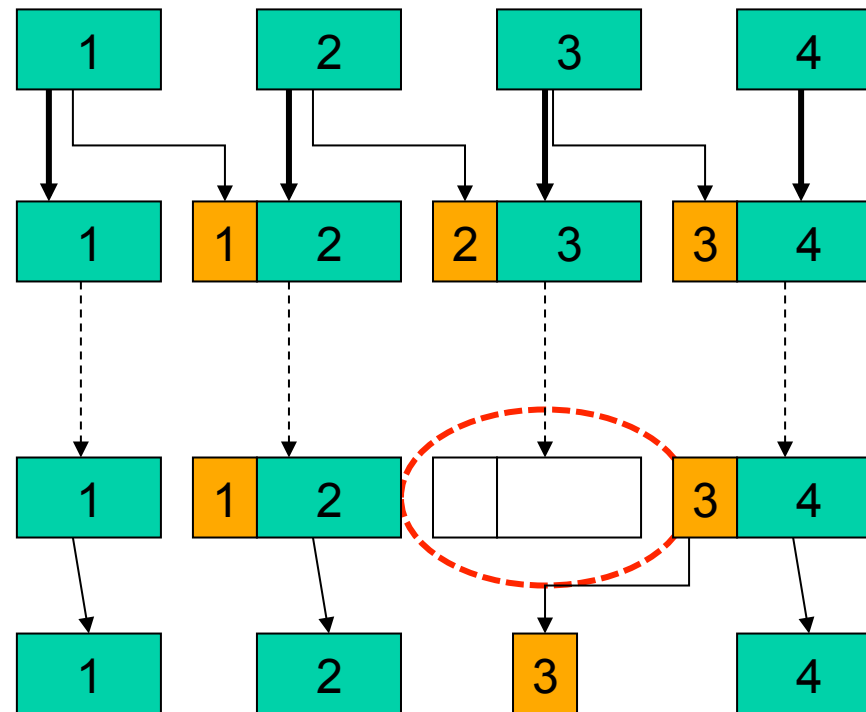
- (1) What if lose more than one packet per group
- (2) What is the effect on playback delay?

Reconstructing the lost packet

Packet 1	0	1	1	1	0	1	0	1
XOR packet	0	1	0	0	0	1	0	1
Recovered packet 2	0	0	1	1	0	0	0	0

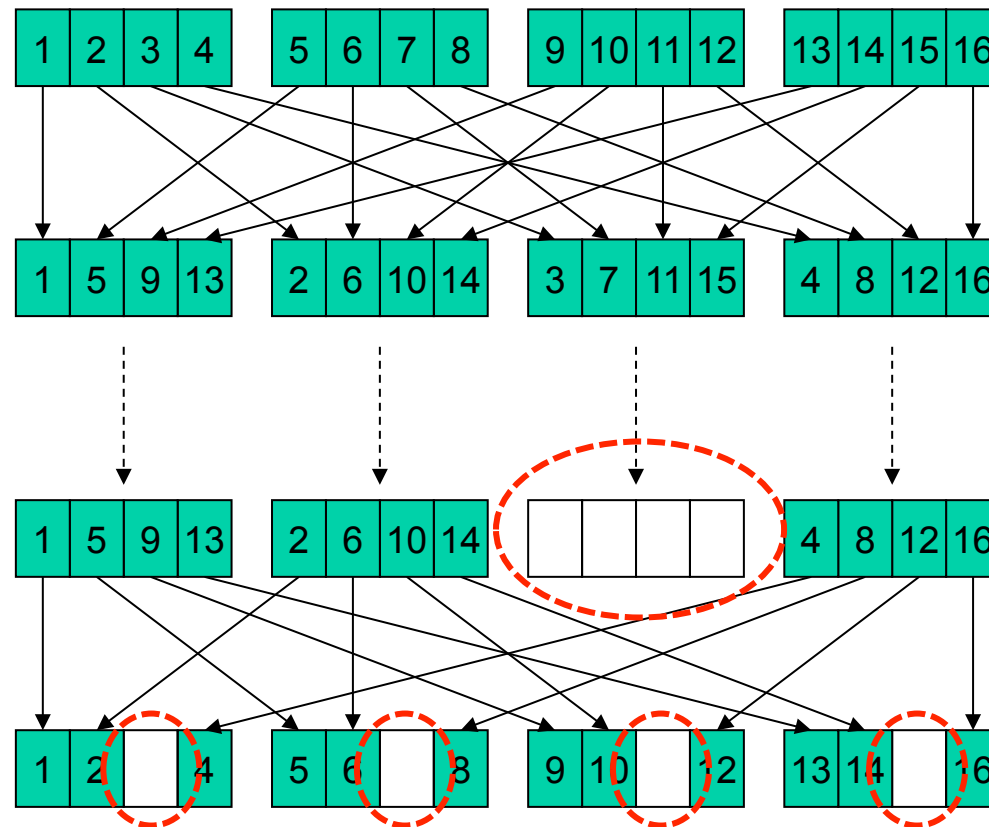
Concealing packet loss

- **Alternative scheme**
 - High resolution stream
 - Low resolution stream
- Carry redundant data with **every** packet.
- Redundant data is lower quality version of previous packet (for example, lower quality audio).



Interleaving

- Interleave so that if do lose packets, do not lose adjacent packets.
- **Good**: no extra data is transmitted.
- **Bad**: playback is delayed.



Summary: bag of tricks for media

- UDP to avoid TCP congestion control (delays) for time-sensitive traffic
- Client-side adaptive playout delay: to compensate for network delay
- Error recovery (on top of UDP)
 - FEC, interleaving
 - retransmissions, time permitting
 - conceal errors: repeat nearby data