

CN-Advanced L36

MultiMedia Networking

Dr. Ram P Rustagi
rprustagi@ksit.edu.in
<http://www.rprustagi.com>
<https://www.youtube.com/rprustagi>

Acknowledgements

Chapter 7 Multimedia Networking

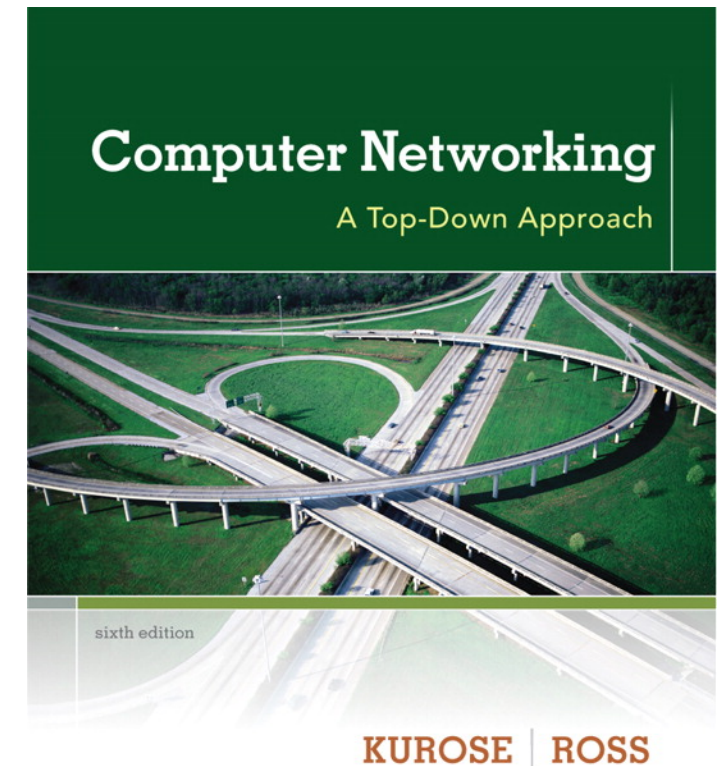
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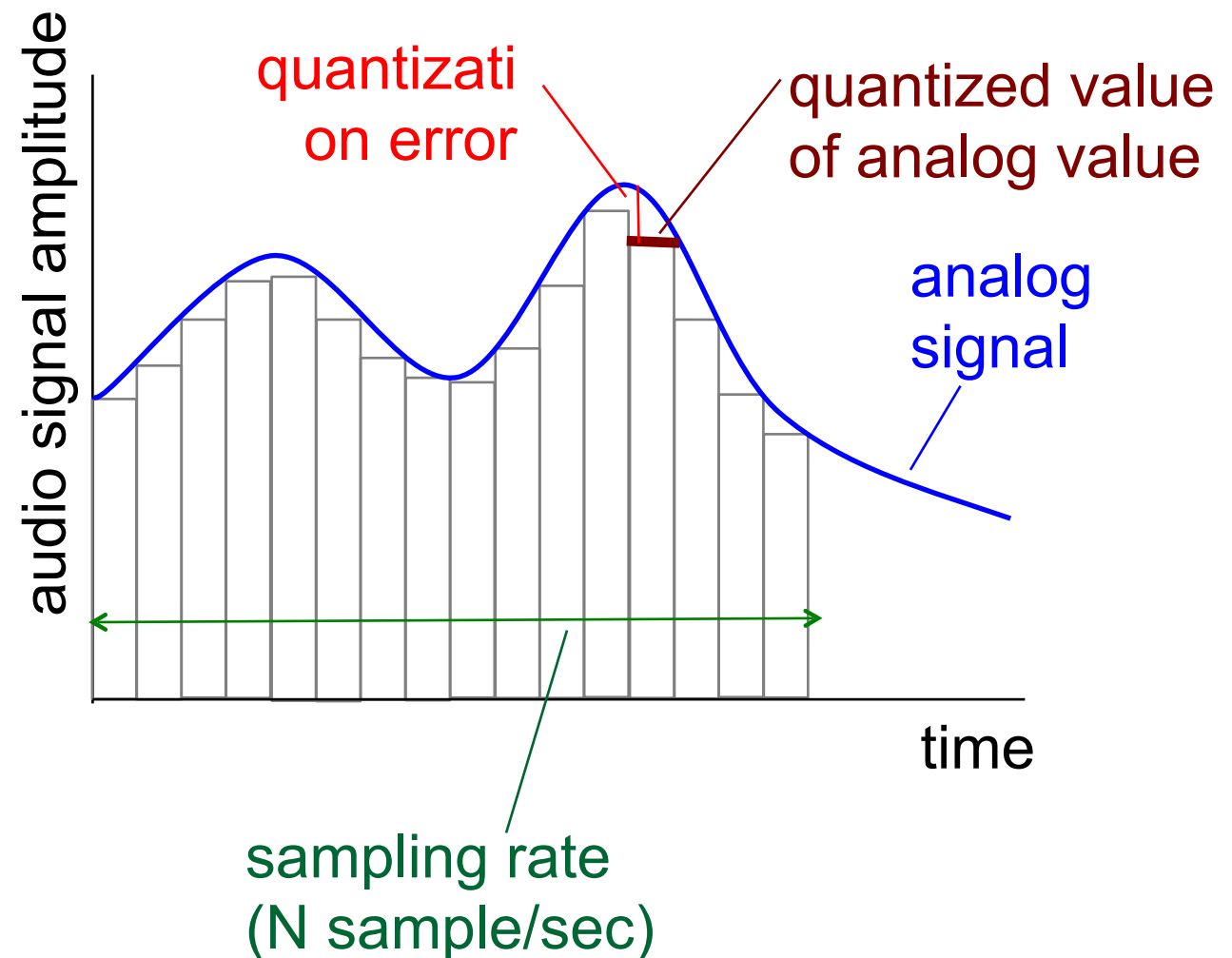
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Computer
Networking: A Top
Down Approach
6th edition
Jim Kurose, Keith Ross
Addison-Wesley
March 2012

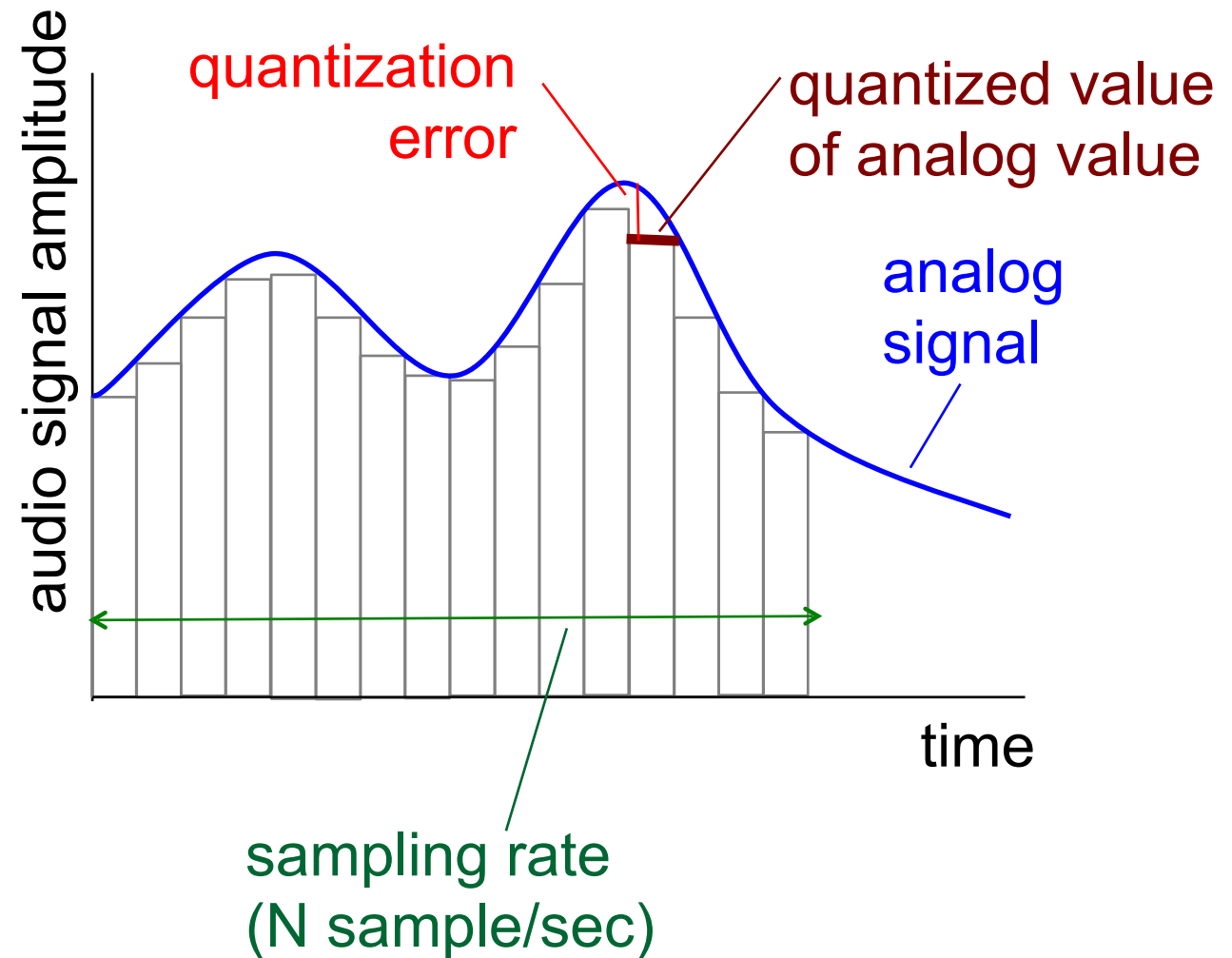
Multimedia: audio

- analog audio signal sampled at constant rate
 - telephone: 8,000 samples/sec
 - CD music: 44,100 samples/sec
- each sample quantized, i.e., rounded
 - e.g., $2^8=256$ possible quantized values
 - each quantized value represented by bits, e.g., 8 bits for 256 values



Multimedia: audio

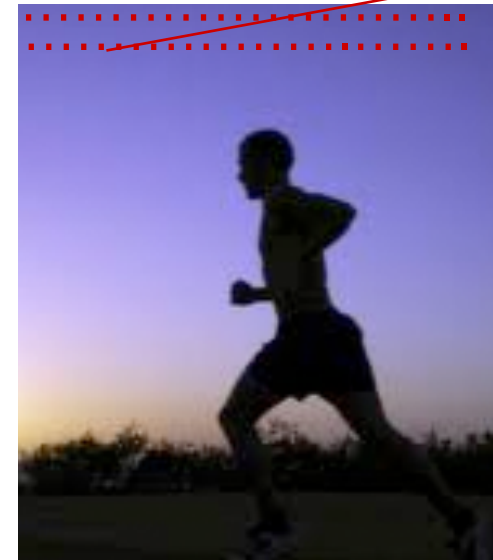
- example: 8,000 samples/sec,
256 quantized values:
64,000 bps
- receiver converts bits back
to analog signal:
 - some quality reduction
- example rates
- CD: 1.411 Mbps
- MP3: 96, 128, 160 kbps
- Internet telephony: 5.3
kbps and up



Multimedia: video

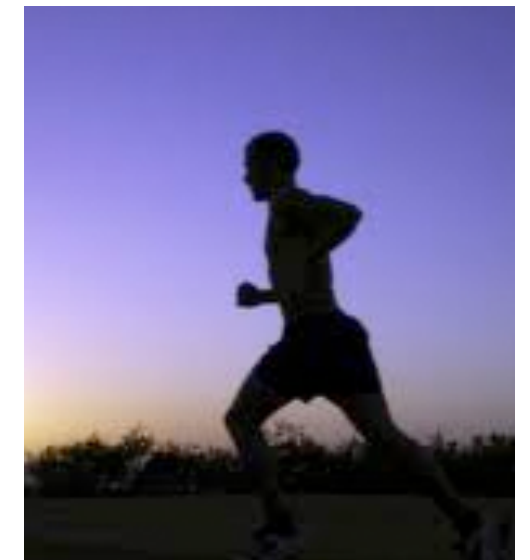
- video: sequence of images displayed at constant rate
 - e.g., 24 images/sec
- digital image: array of pixels
 - each pixel represented by bits
- coding: use redundancy *within* and *between* images to decrease # bits used to encode image
 - spatial (within image)
 - temporal (from one image to next)

spatial coding example: instead of sending N values of same color (all purple), send only two values: color value (purple) and number of repeated values (N)



frame i

temporal coding example: instead of sending complete frame at $i+1$, send only differences from frame i



frame i+1

Multimedia: video

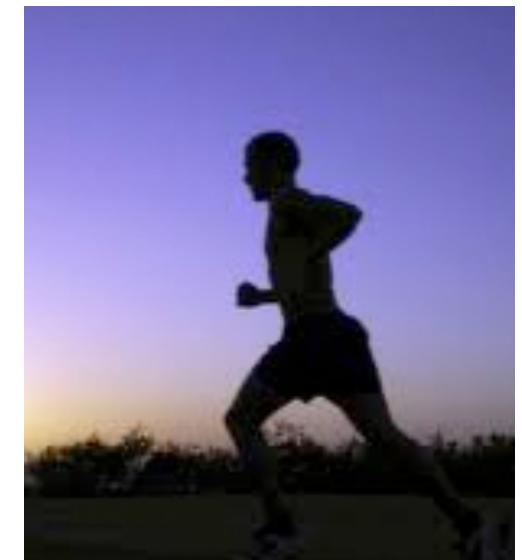
- **CBR: (constant bit rate):**
video encoding rate fixed
- **VBR: (variable bit rate):**
video encoding rate changes as amount of spatial, temporal coding changes
- **examples:**
 - MPEG I (CD-ROM) 1.5 Mbps
 - MPEG2 (DVD) 3-6 Mbps
 - MPEG4 (often used in Internet, < 1 Mbps)

spatial coding example: instead of sending N values of same color (all purple), send only two values: color value (purple) and number of repeated values (N)



frame i

temporal coding example:
instead of sending complete frame at i+1, send only differences from frame i

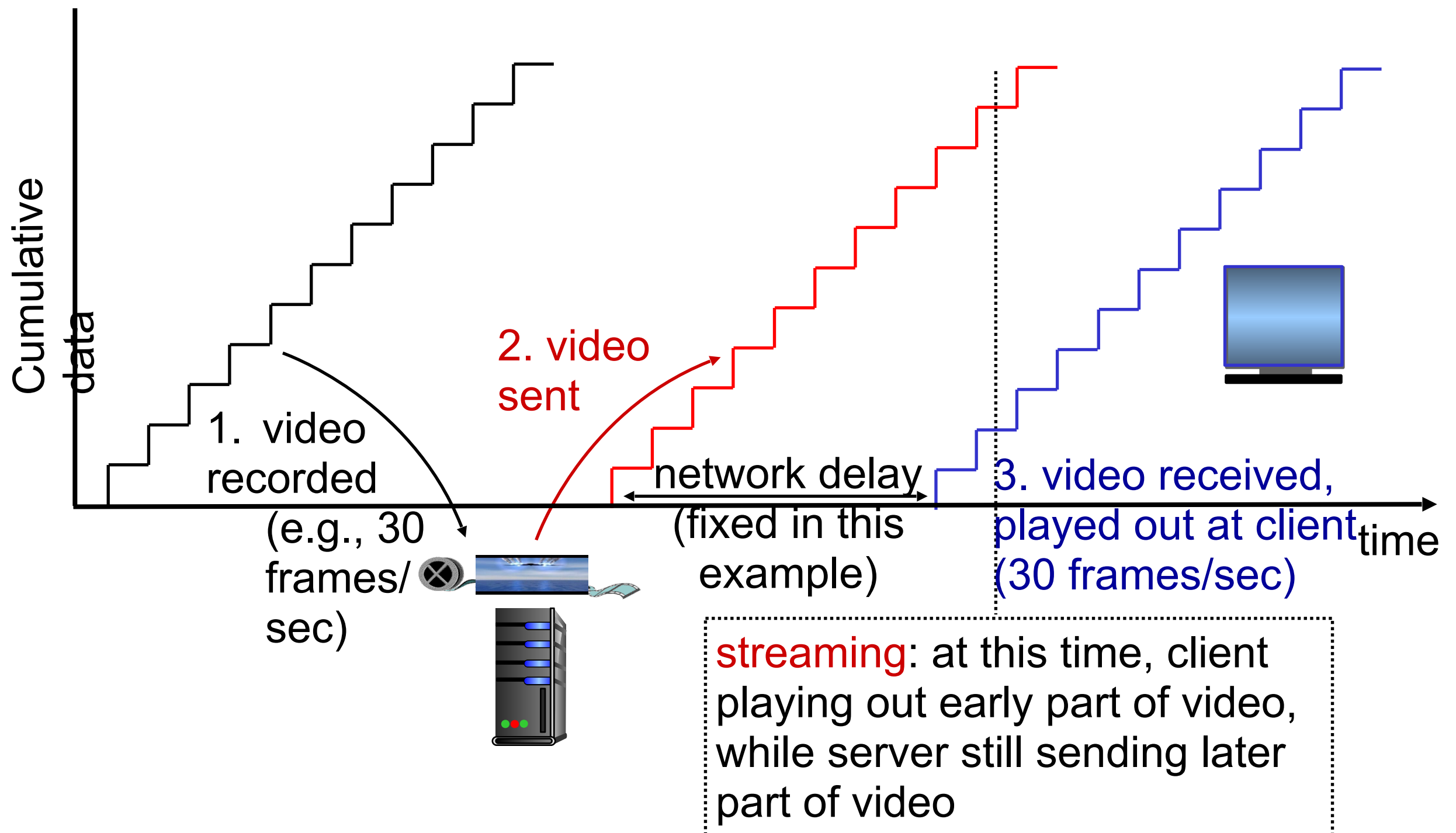


frame i+1

Multimedia networking: 3 application types

- *streaming, stored* audio, video
 - *streaming*: can begin playout before downloading entire file
 - *stored (at server)*: can transmit faster than audio/video will be rendered (implies storing/buffering at client)
 - e.g., YouTube, Netflix, Hulu
- *conversational* voice/video over IP
 - interactive nature of human-to-human conversation limits delay tolerance
 - e.g., Skype
- *streaming live* audio, video
 - e.g., live sporting event (futbol)

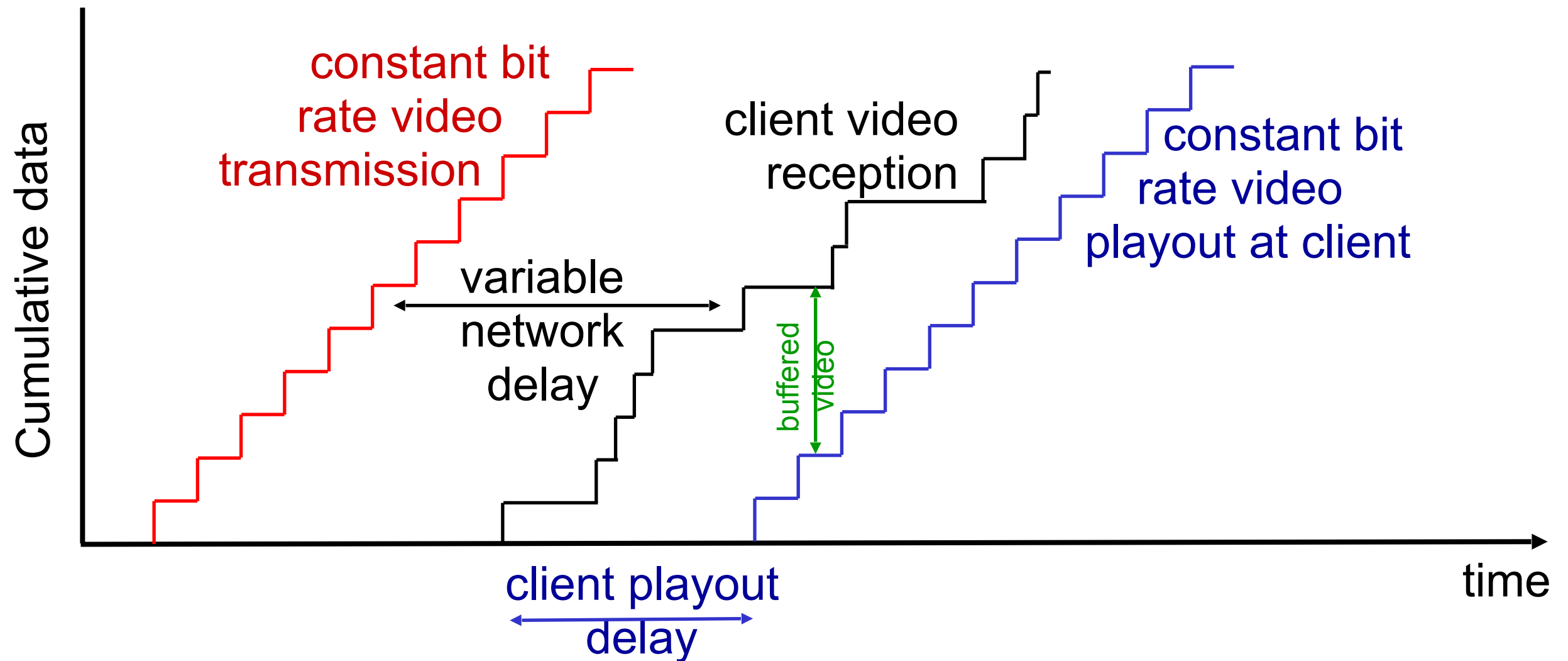
Streaming stored video:



Streaming stored video: challenges

- **continuous playout constraint**: once client playout begins, playback must match original timing
 - ... but **network delays are variable** (jitter), so will need **client-side buffer** to match playout requirements
- other challenges:
 - client interactivity: pause, fast-forward, rewind, jump through video
 - video packets may be lost, retransmitted

Streaming stored video: revisited

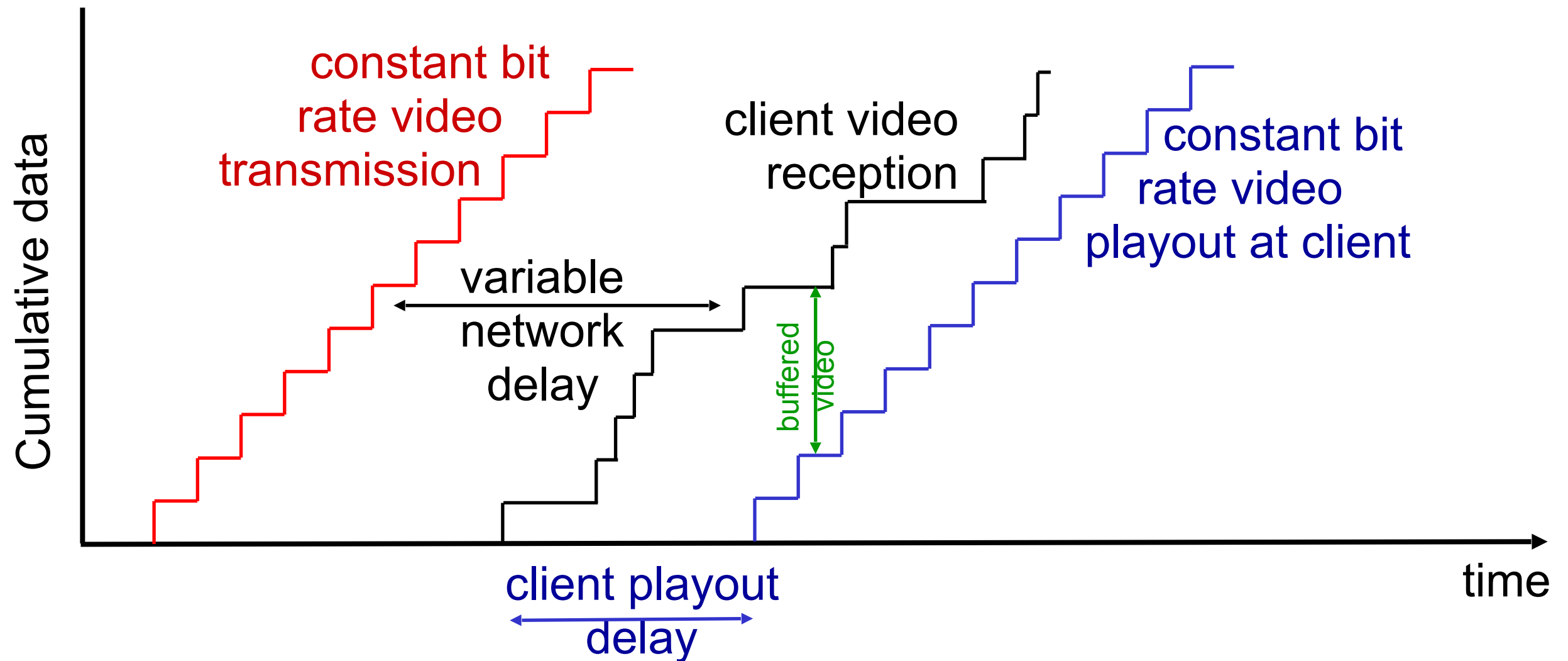


- ***client-side buffering and playout delay:*** compensate for network-added delay, delay jitter

Streaming stored video: challenges

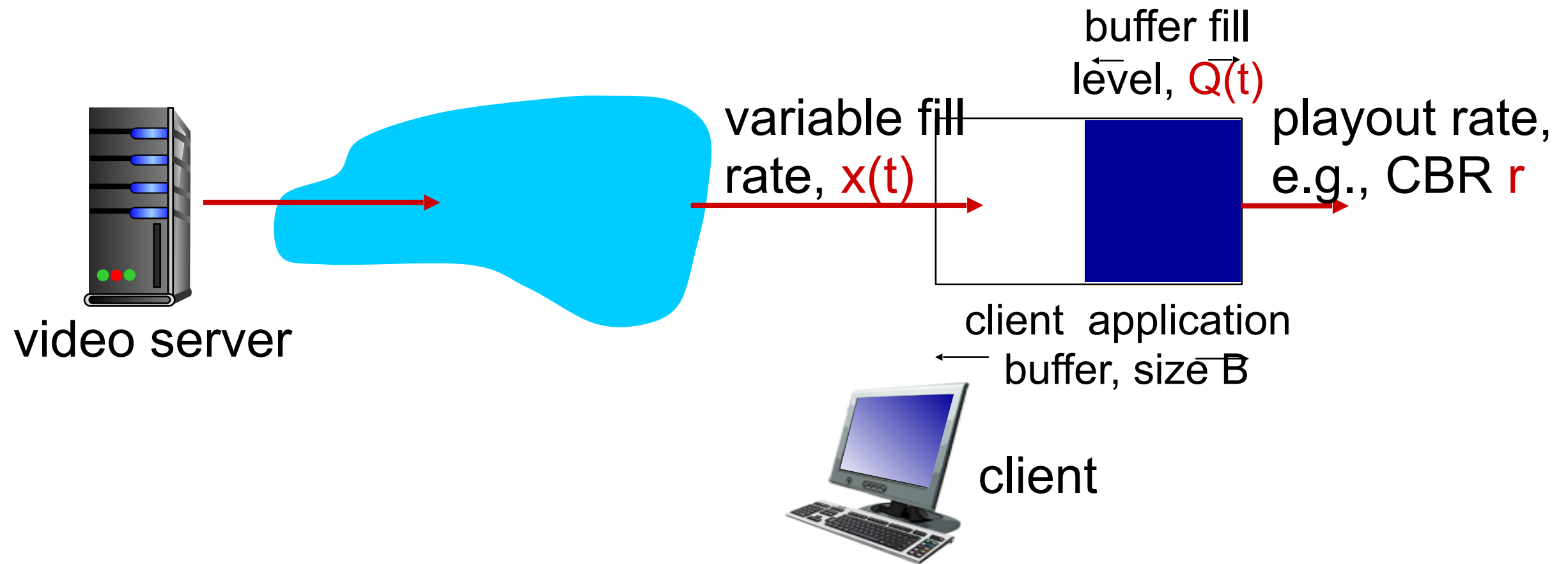
- **continuous playout constraint**: once client playout begins, playback must match original timing
 - ... but **network delays are variable** (jitter), so will need **client-side buffer** to match playout requirements
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Streaming stored video: revisited

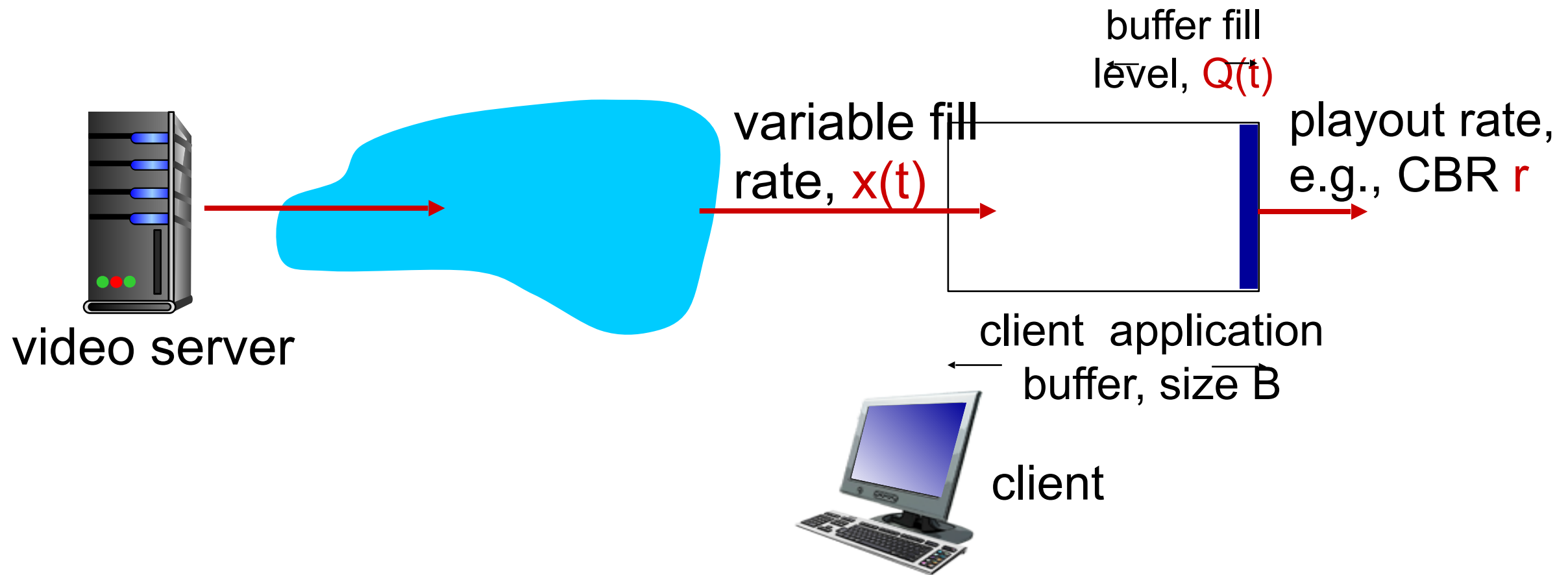


- *client-side buffering and playout delay*: compensate for network-added delay, delay jitter

Client-side buffering, playout

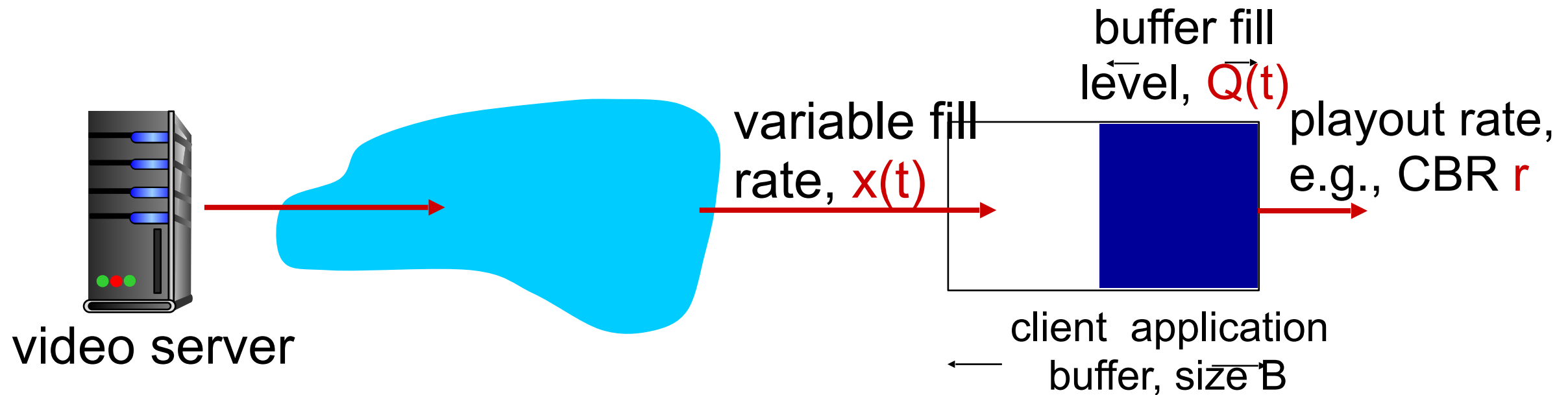


Client-side buffering, playout



1. Initial fill of buffer until playout begins at t_p
2. playout begins at t_p ,
3. buffer fill level varies over time as fill rate $x(t)$ varies and playout rate r is constant

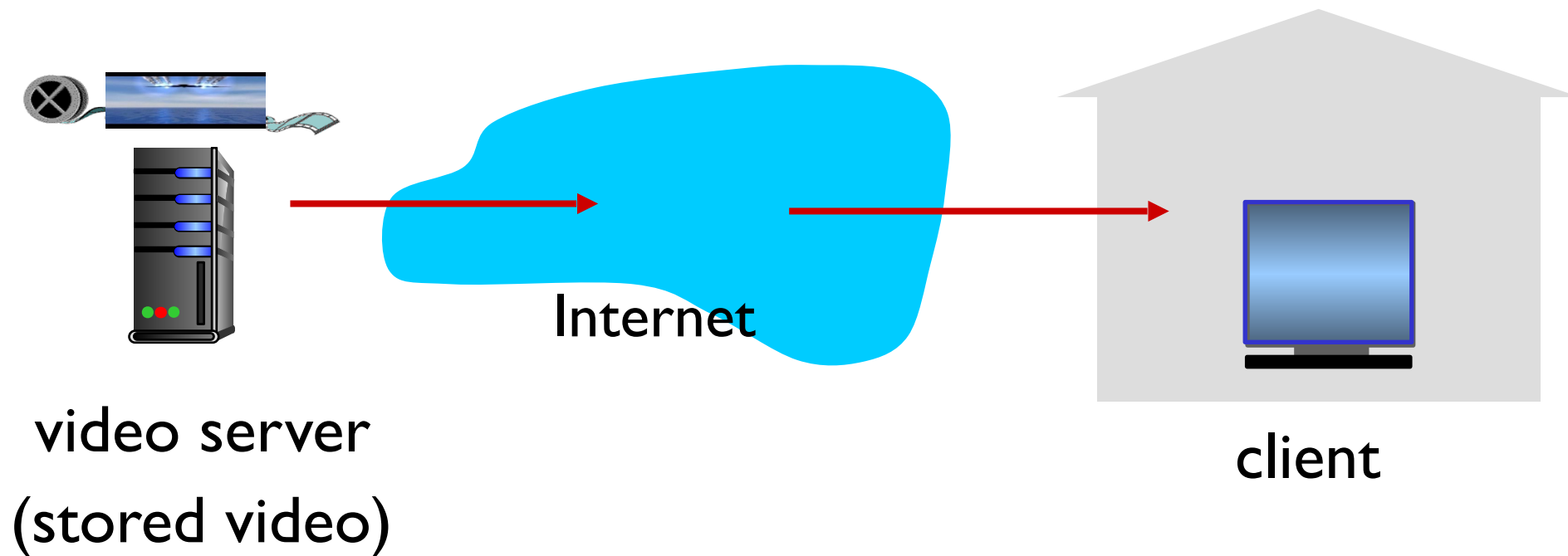
Client-side buffering, playout



- *playout buffering: average fill rate (x), playout rate (r):*
- $x < r$: buffer eventually empties (causing freezing of video playout until buffer again fills)
- $x > r$: buffer will not empty, provided initial playout delay is large enough to absorb variability in $x(t)$
 - *initial playout delay tradeoff*: buffer starvation less likely with larger delay, but larger delay until user begins watching

Streaming stored video:

simple scenario:



Summary

- Multimedia audio
- Multimedia video
- 3 Application types
 - Streaming stored audio/video
 - Conversational audio/video
 - Streaming live audio/video