

# Internet Video Delivery Basics

ECE 50863 – Computer Network Systems

# Types of Internet Video

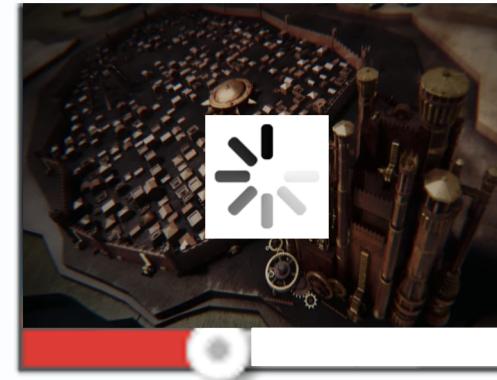
- Video Conferencing
  - Interactive: User must get video in 100-200 milliseconds.
- Video Streaming/ Video on Demand (VoD)
  - E.g., streaming a movie, recorded lecture or newsclip
  - Once video starts, playback should be “smooth”
  - Can prefetch content (tens of seconds or even a few minutes) in the video player buffer.
- Live video broadcast
  - E.g., tuning into a sports game
  - Typically 5-10 seconds of delay may be acceptable: not as interactive as video conferencing
  - Similar to VoD, but keep client buffer small (a few seconds).

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# Key considerations for video streaming

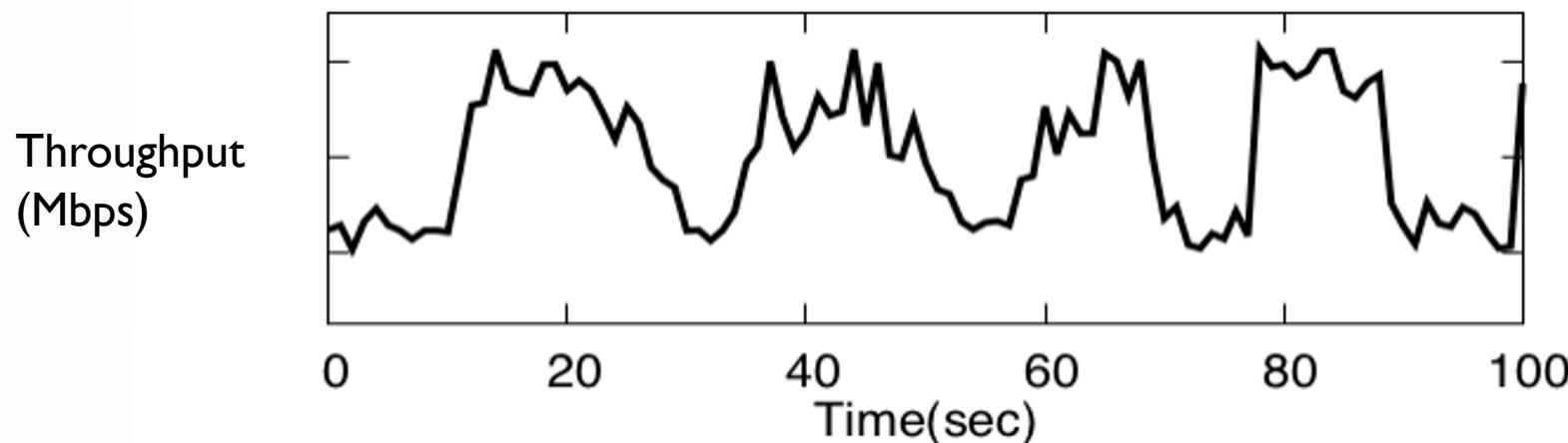
- Join Latency
  - How soon should video start playing back, once a user decides to play the video
- Avoid “rebuffering” events.
  - Once client starts playing video, avoid “stalls”.



Rebuffing

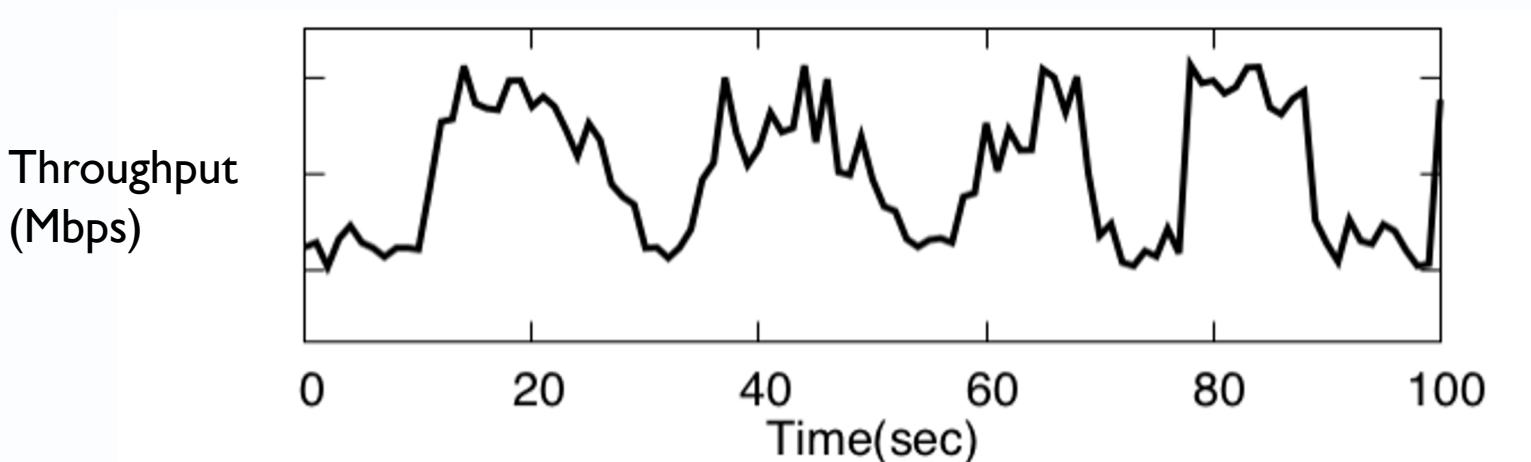
# Challenges in streaming Internet video

- Internet throughput is highly variable owing to congestion
- Internet video needs a certain throughput to avoid stall.
  - E.g., to send video at 1080p, typically need 5 Mbps
- How do we handle dips in throughput?

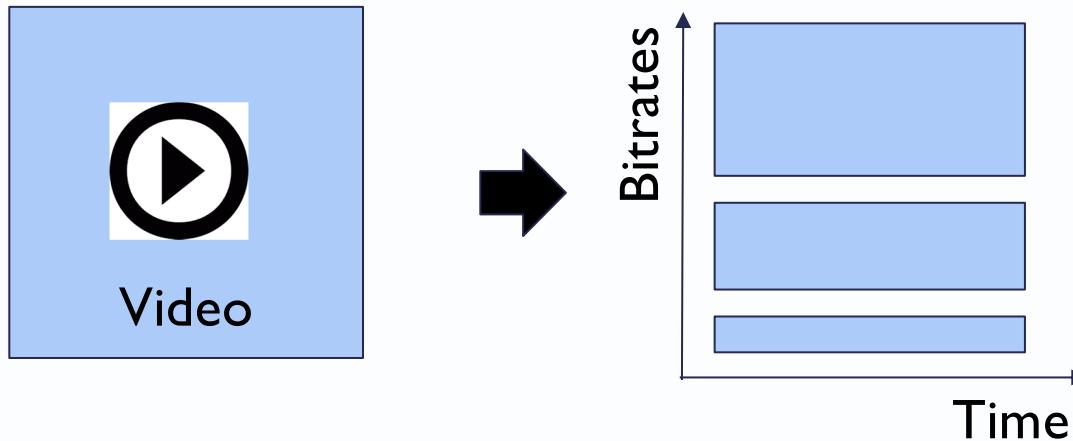


# Approaches for streaming Internet video

- Static approach: Pick video rate conservatively at start of session
  - Low enough that it is highly unlikely the connection will see poorer throughput
  - Problem: May need to be very conservative, and even then cannot handle occasional dips when they arise.
- Adaptive approach: Adapt video rate (and quality) based on network conditions

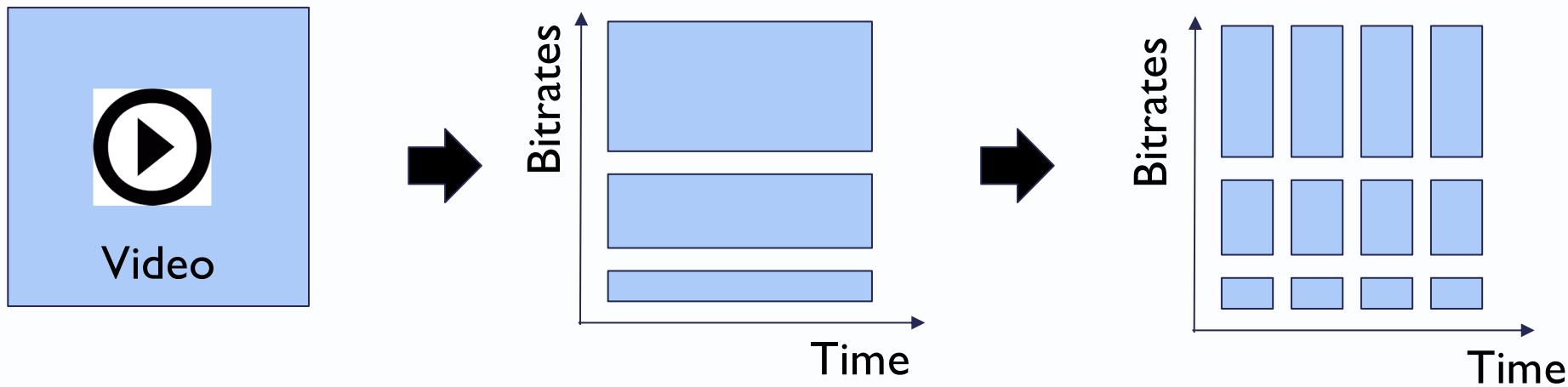


# Adaptive Bitrate Streaming



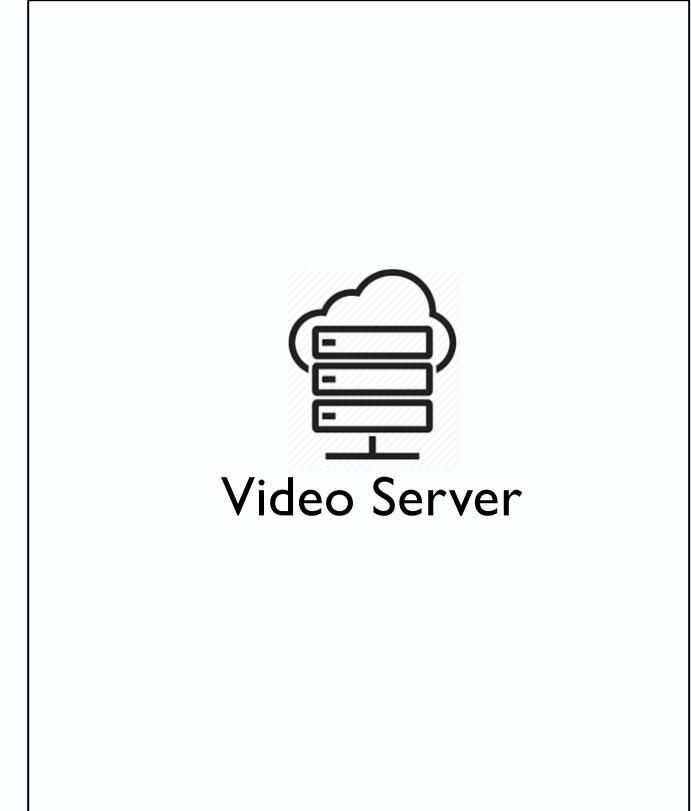
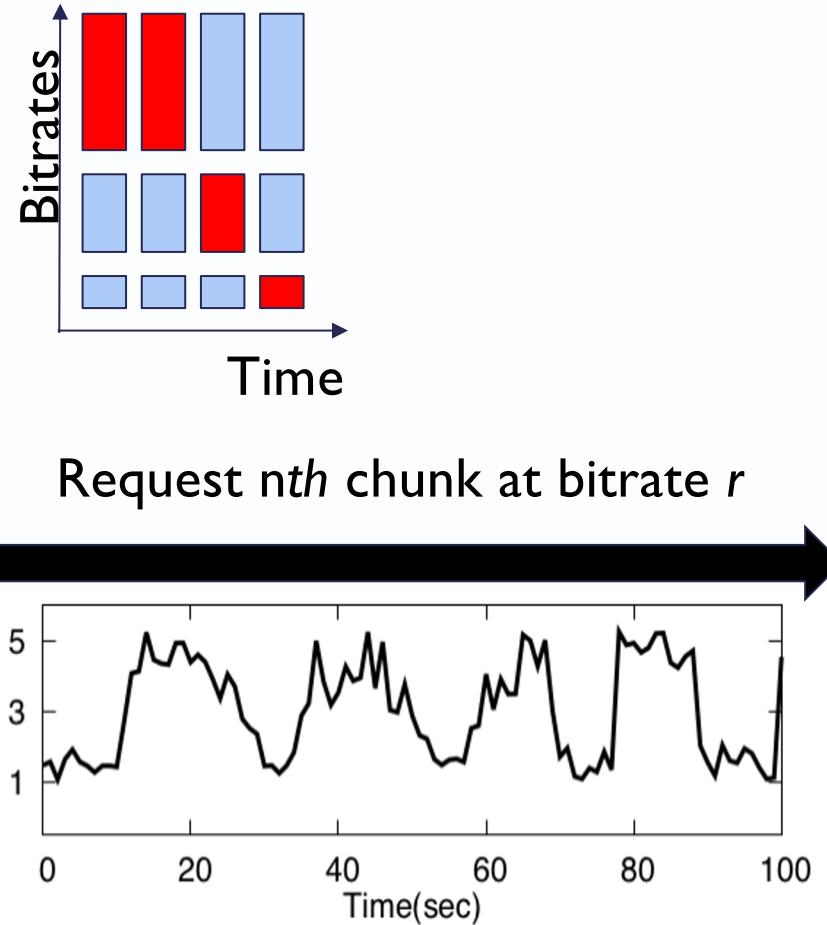
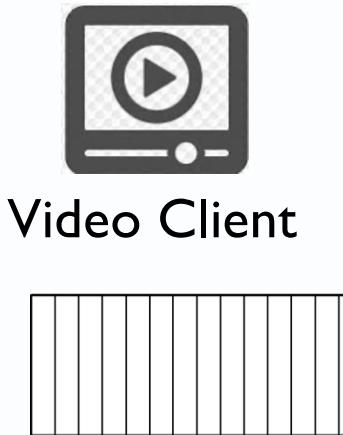
A video clip is encoded  
with multiple qualities (bitrates)

# Adaptive Bitrate Streaming.



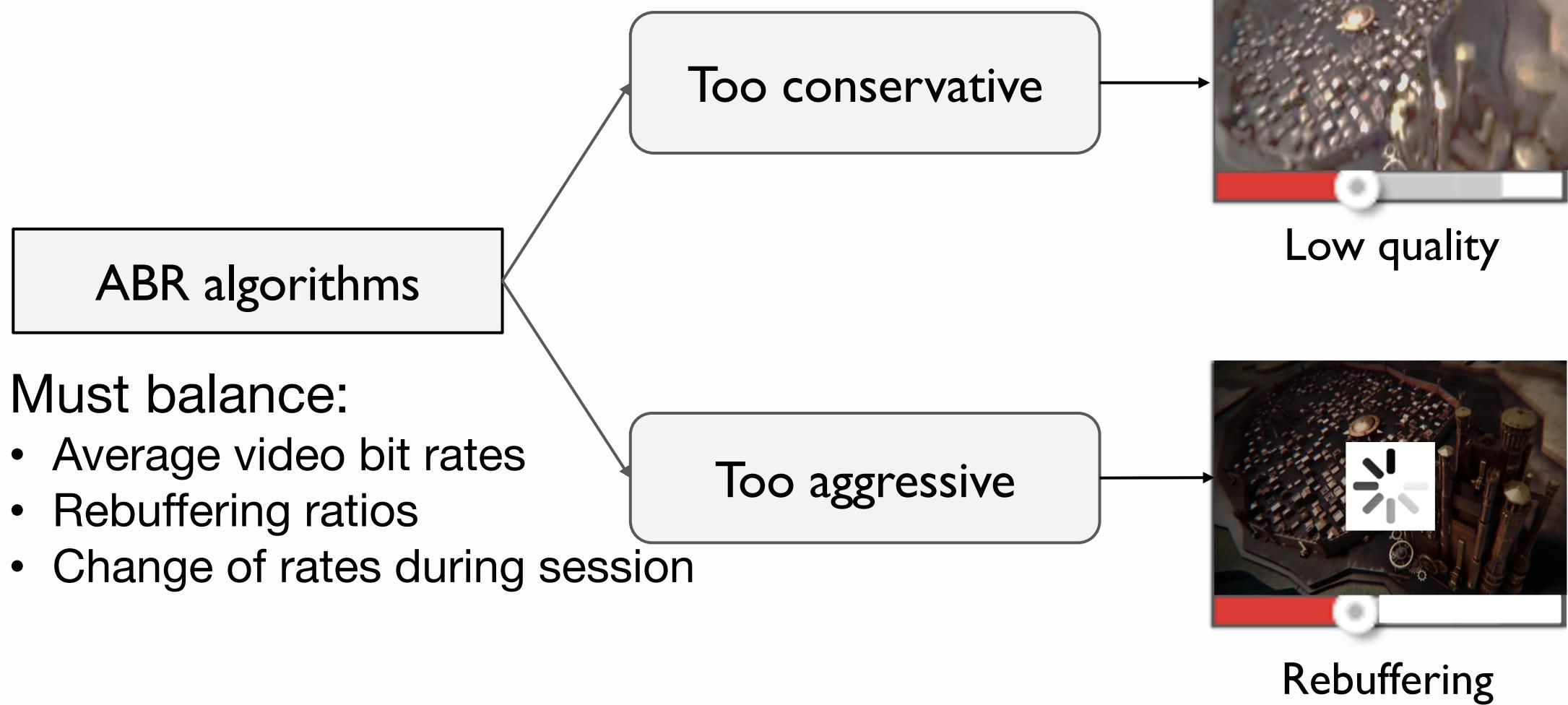
Video encoded at each bitrate is split into chunks

# Adaptive Bitrate Streaming



**Adaptive Bitrate Algorithms(ABR)**

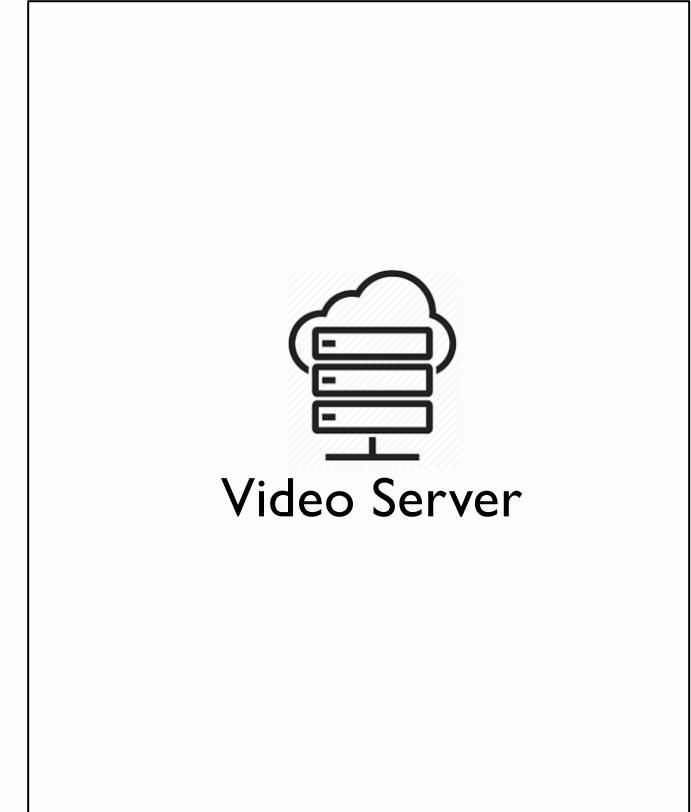
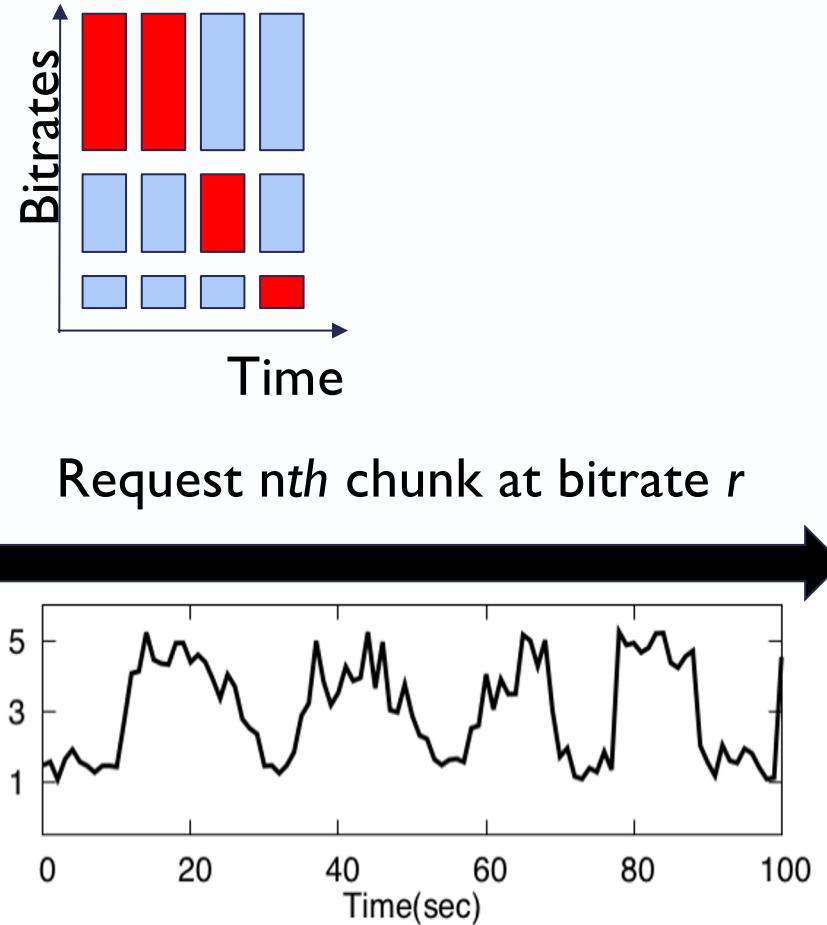
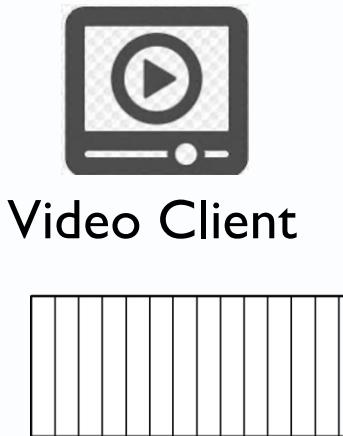
# Adaptive Bitrate Streaming



# Examples of Adaptive BitRate Algorithms

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# Recap: Adaptive Bitrate Streaming



**Adaptive Bitrate Algorithms(ABR)**

## How do we design ABR algorithms?

- Many different algorithms used in the industry.
- Active area of research in networking community
- Discuss two example research algorithms
  - Do not accept research publications as gospel!
  - Critique them, question their claims, ask if you can do better

# Decision that an ABR algorithm needs to make



**4 sec of chunks  
in the player buffer**

Bitrate  
Decision



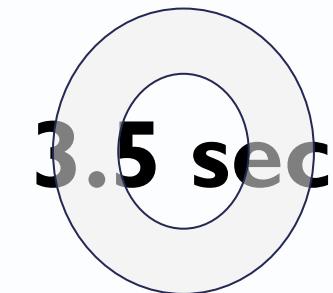
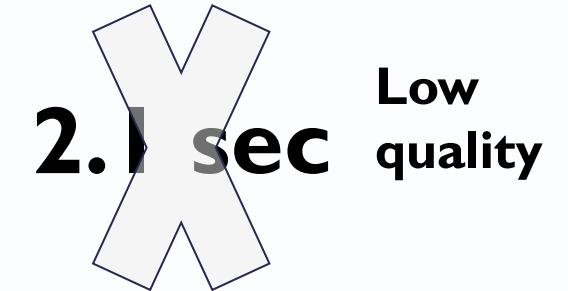
Given I have a certain amount of data in my client buffer, which quality should I download next chunk?

# What if we could perfectly predict download times?



**4 sec of chunks  
in the player buffer**

Bitrate  
Decision



## Example Algorithm 1

- ***“A buffer-based approach to rate adaptation: evidence from a large video streaming service”, Huang et al., ACM Sigcomm 2014.***
- Do not try to predict throughput (and hence chunk download times)
  - Prediction is hard and often erroneous
- Simply consider the occupancy of the client buffer in the decision process
  - Low occupancy => lower bit rate, High occupancy => higher bit rate
- Three versions of the algorithm, each considers additional enhancements not considered in the previous version.

## Example Algorithm 2

- ***“A Control-Theoretic Approach for Dynamic Adaptive Video Streaming over HTTP”, Yin et al., ACM Sigcomm 2015***
- Predict throughput, and consider buffer occupancy
- Makes decisions considering multiple factors:
  - Average video bit rates
  - Rebuffering ratios
  - Change of rates during session
- Optimizes “QoE”, a composite metric that combines the above three metrics

## How the control theoretic algorithm works

- Lets assume chunks  $1, 2, \dots, i$  have been downloaded.
- Algorithm considers a lookahead of  $k$  chunks
- Predicts throughput for this period of time.
- Pick quality for next  $K$  chunks based on permutation with best QoE score.
  - $Q$  different qualities ( $q_1, q_2, q_3$  etc.)  $\Rightarrow Q^K$  possible permutations
- Download chunk  $i+1$ , adjust predictions and go back to first step.

$i+1$	$i+2$	....	$i + K$
$q_1$	$q_1$	...	$q_1$
$q_3$	$q_3$	....	$q_3$
$q_1$	$q_3$	...	$q_1$
$q_2$	$q_1$	...	$q_3$

## Questions to think about..

- How do we predict throughput based on throughput of last few chunks?
- Does a purely buffer-based approach suffice? When can throughput prediction help?
- How could we be robust to prediction errors?
- How well can the algorithm cope with “sudden and prolonged” bandwidth dips?
- Strengths and weaknesses of each algorithm?