Quality of Service

Quality of Service

- How "good" are late data and lowthroughput channels?
- It depends on the application. Do you care if...
 - Your e-mail takes 1/2 hour to reach your friend?
 - You have to spend 1/2 hour to make a cheaper plane reservation on the Web?
 - Your call to 911 takes 1/2 hour to go through your nifty new IP phone service?



Application Requirements

- Internet currently provides one single class of "best-effort" service
 - No assurances about delivery
- High speed networks have enabled new applications
 - Require "deliver on time" assurances from the network
 - Real-time applications
 - Sensitive to the timeliness of their data
 - Voice
 - Video
 - Industrial control



Timely Delivery

- How to achieve timely delivery
 - When actual RTT small (< 2/3) relative to acceptable delay
 - Retransmit
 - When base RTT (no queuing delay) large
 (> 2) relative to acceptable delay
 - Impossible
 - Otherwise possible, but not through retransmission



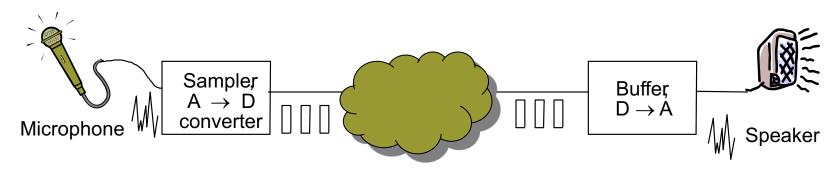
Timely Delivery

- Within the 48-state U.S
 - Base RTT (no queueing delay) peaks around 75 msec
 - Actual RTT is often 10-100 msec
- Humans notice about 50 msec delay for voice
 - Use erasure codes across packets, or
 - Support delay preferences in the network;
 called quality of service, or QoS



Real-time Applications

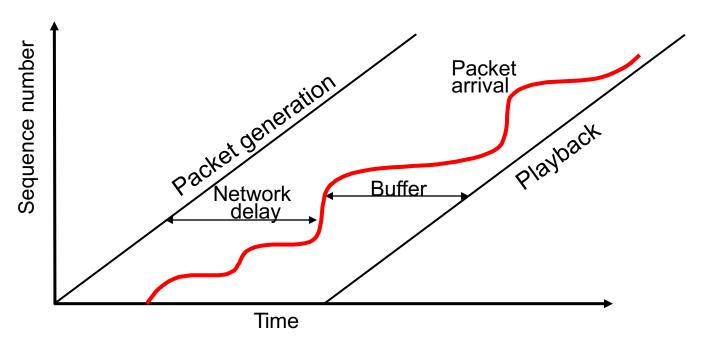
- Two types or applications
 - Hard real-time
 - Elastic (soft real-time)
- Example real-time application requirements audio
 - Sample voice once every 125μs
 - Each packet has a playback time
 - Packets experience variable delay in network
 - Add constant factor to playback time playback point





Real-Time Applications

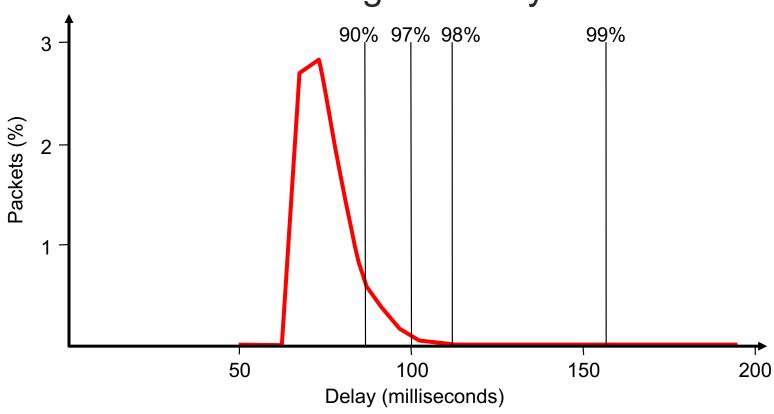
Playback Buffer





Delay Distribution

What is a good delay?





Quality of Service Approaches

- Approach : Admission control
 - Flow tells the network what it wants
 - Network decides if flow can be admitted
- Fine-grained
 - Provide QoS to individual applications or flows
 - Example: Resource Reservation Protocol (RSVP)
- Coarse-grained
 - Provide QoS to large classes of data or aggregated flows
 - Example: Differentiated Services (DIFFSERV)



Mechanisms

- Flow specification
 - Tell the network what the flow wants
- Admission control
 - Network decides if it can handle flow
- Reservation
 - Enable admission control
- Packet classification
 - Map packets to flows
- Scheduling
 - Forwarding policy



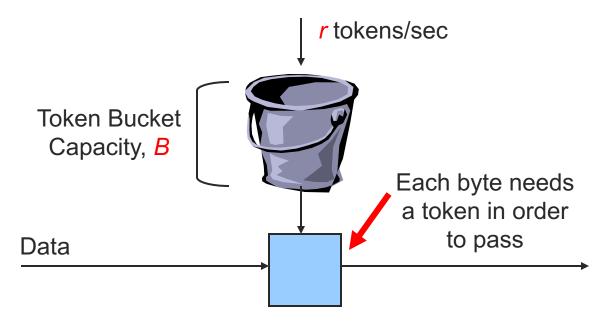
Characterizing a Flow

- Describe flow's traffic characterization
 - Average bandwidth + burstiness: token bucket filter
 - o Token rate:
 - Bucket depth:

Use

- Must have a token to send a byte
- Must have n tokens to send n bytes
- Start with no tokens
- Accumulate tokens at rate of r per second
- Can accumulate no more than B tokens



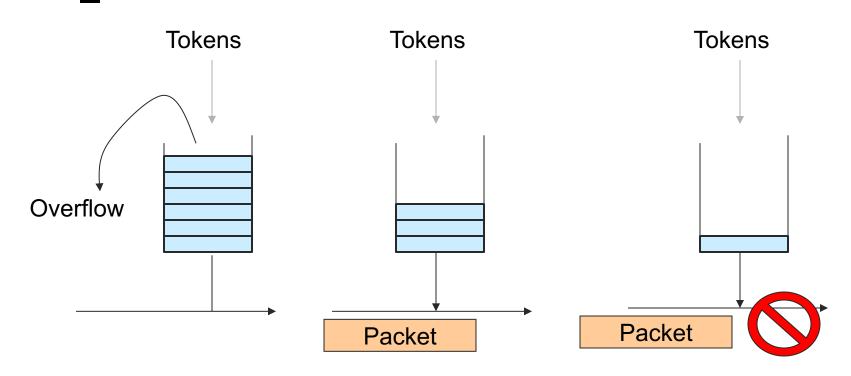


Dropping Filter: drops packets if token is not available

Buffered Filter: buffers data until tokens become available



Token Bucket Operation



Buffer tokens up to capacity of bucket

Enough tokens → packet goes through, tokens removed

Not enough tokens

→ wait for tokens to
accumulate

Question

 Given a finite length data stream, will it be affected by a token bucket filter?



Not if during every time interval, the number of bytes is less than or equal to B + rt, where t is the length of the interval

Given a token rate r and a finite data trace, how can the minimum token bucket size B be found such that the filter has no effect?

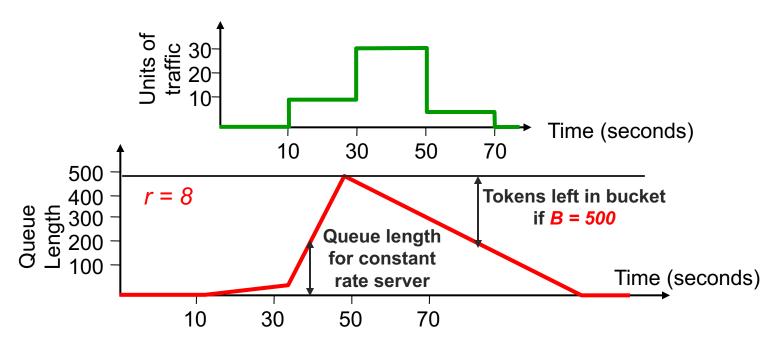


Given a token rate r and a finite data trace, how can the minimum token bucket size B be found such that the filter has no effect?



- Simply observe the maximum buffer size
 - Why?
 - If the buffer is truncated to size B, then the number of empty buffer positions is equivalent to the number of tokens in an (r,B) token bucket filter





The number of empty buffer positions for buffer size B and a constant rate server is equivalent to the number of tokens in an (r,B) token bucket filter



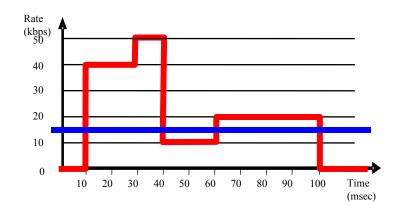
r = 15 kbps



What is the minimum size of B required so that the filter lets the stream pass with no loss or delay?



r = 15 kbps



 \blacksquare Min B =

$$(40 - 15) * 20 +$$

$$(50 - 15) * 10 -$$

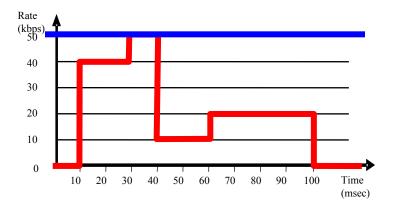
$$(15-10)*20+$$

$$(20-15)*40$$

= 950 bits



- What is the minimum B needed for arbitrary r > 0
- If $r \ge 50$ B = 0
- If $50 > r \ge 40$ Min B = (50 - r) * 10
- If $40 > r \ge 20$ Min B = (40 - r) * 20 + (50 - r) * 10
- If $20 > r \ge 10$ Min B = (40 - r) * 20 + (50 - r) * 10 - (r - 10) * 20 + (20 - r) * 40
- If 10 > r ≥ 0Min B = (40 - r) * 20 + (50 - r) * 10 + (10 - r) * 20 + (20 - r) * 40



Differentiated Services

Goal

- Scalability through the use of only a small number of service classes
 - Two classes
 - Regular and premium (i.e. first class and bulk mail)
 - Diffserv
 - Proposes 6 bits of IP ToS field (26 = 64 classes)

Questions

- Who is allowed to set the premium bit?
 - Typically an ISP
 - Should we allow an individual customer or application?
- How do routers react to such a classification?
 - IETF has specified per-hop behavior



Differentiated Services

- Expedited forwarding
 - Per-hop behavior
 - Need to strictly limit the load of traffic receiving expedited forwarding
 - Give strict priority
 - Use weighted fair queueing (WFQ) and assign sufficiently large weights for traffic receiving expedited forwarding



Differentiated Services

- Assured forwarding
 - Per-hop behavior
 - Like RED but with "in" and "out" packets (RIO)
 - Does not reorder packets
 - For more than two classes of traffic, use weighted RED
- Profile meters at the edges of ISP networks could mark packets as "in" or "out"

