# 15-441/641: Content Delivery Peer-to-Peer

15-441 Spring 2019 Profs **Peter Steenkiste** & Justine Sherry



Spring 2019 https://computer-networks.github.io/sp19/



#### Overview

- · What is QoS?
- · Queuing discipline and scheduling
- Traffic Enforcement
- · Integrated services



#### What is QoS?

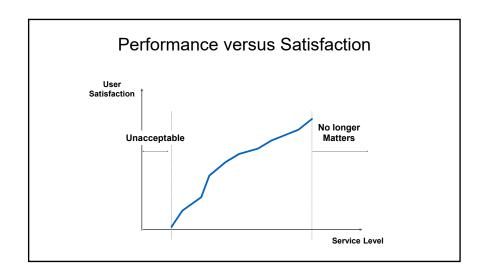
- The Internet supports best effort packet delivery
- · Sufficient for most applications
- But some applications require or can benefit from a "higher" level of service
- "Higher" quality of service can mean that bounds are provided for one or more performance parameters
  - · Bandwidth: fast data transfers, video
  - · Delay, jitter: telephony, interactive video
  - · Packet loss: update services
- · QoS can also mean that a user gets "better" treatment (than other users)
- But no guarantees are given, e.g., the "10 items or less" line in the grocery store

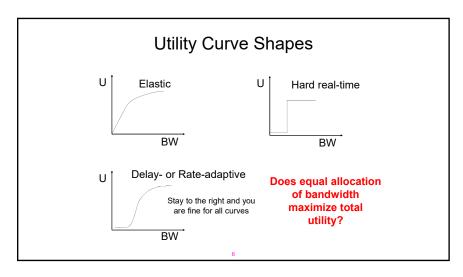


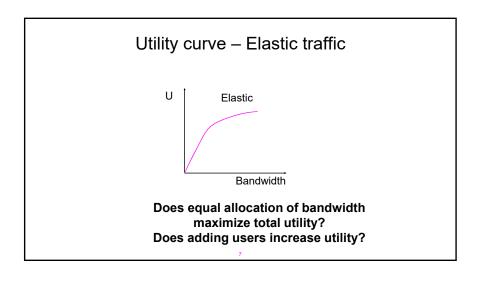
# Why Should we Consider QoS?

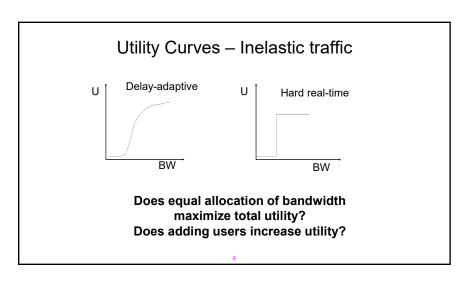
- · What is the **basic objective** of network design?
  - Maximize total bandwidth? Minimize latency?
  - Maximize user satisfaction the total utility given to users
  - · Maximize profit?
- · What does utility vs. bandwidth look like?
- Utility: represents how satisfied a user is with the service
- · Shape depends on application
- · Must be non-decreasing function











# **Inelastic Applications**

- · Continuous media applications
  - · Lower and upper limit on acceptable performance.
  - · BW below which video and audio are not intelligible
  - Internet telephones, teleconferencing with high delay (200 300ms) impair human interaction
  - Sometimes called "tolerant real-time" since they can adapt to the performance of the network
- · Hard real-time applications
  - · Require hard limits on performance
  - E.g. control applications



#### Quality of Service versus Fairness

- · Traditional definition of fairness: treat all users equally.
  - · E.g., share bandwidth on bottleneck link equally
- · QoS: treat users differently.
  - For example, some users get a bandwidth guarantee, while others have to use best effort service
- · The two are not in conflict
  - · All else being equal, users are treated equally
  - · Unequal treatment is based on policies, price:
  - · Administrative policies: rank or position
  - · Economics: extra payment for preferential treatment



# QoS Analogy: Surface Mail

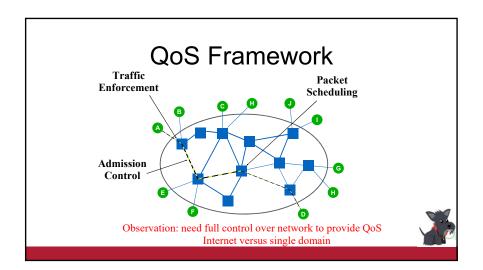
- · The defaults if "first class mail".
  - · Usually gets there within a few days
  - · Sufficient for most letters
- Many "guaranteed" mail delivery services: next day, 2-day delivery, next day am, .....
  - · Provide faster and more predictable service at a higher cost
  - Providers differentiate their services: target specific markets with specific requirements and budgets
- Why don't we do the same thing in networks?



#### How to Provide QoS?

- · Admission control limits number of flows
  - You cannot provide guarantees if there are too many flows sharing the same set of resources (bandwidth)
  - · For example, telephone networks busy tone
  - · This implies that your request for service can be rejected
- Traffic enforcement limits how much traffic flows can inject based on predefined limits.
  - · Make sure user respects the traffic contract
  - Data outside of contract can be dropped (before entering the network!) or can be sent at a lower priority
- Scheduling support in the routers guarantee that users get their share of the bandwidth.
- · Again based on pre-negotiated bounds
- · Analogy: service in a grocery store





#### What is a flow?

- · Defines the granularity of QoS and fairness
- · TCP flow
- Traffic to or from a device, user, or network
- Bigger aggregates for traffic engineering purposes
- · Routers use a classifier to determine what flow a packet belongs to
  - Classifier uses a set of fields in the packet header to generate a flow ID
  - Example: (src IP, dest IP, src port, dest port, protocol)
  - · Or: (src prefix, dest prfix), i.e., some fields are wildcards



# Admission Control - Elastic

- · If U(bandwidth) is concave
- → elastic applications
- Incremental utility is decreasing with increasing bandwidth
- It is always advantageous to have more flows with lower bandwidth
- · Increases total utility of flows served
- · No need of admission control

This is why the Internet works!

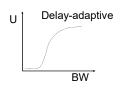
Not so for delay-adaptive and real-time applications





#### Admission Control - Guarantees

- If U is convex → inelastic applications
- U(number of flows) is no longer monotonically increasing
- Need admission control to maximize total utility
- Admission control → deciding when adding more people would reduce overall utility
  - · E.g., bandwidth or latency guarantees
- · Basically avoids overload





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# **Queuing Disciplines**

- · Each router must implement some queuing discipline
- Since you have queues you will need a policy
- · Queuing allocates both bandwidth and buffer space:
- · Bandwidth: which packet to serve (transmit) next
- Buffer space: which packet to drop next (when required)
- · Queuing discipline affects latency, bandwidth, ..



# **Network Queuing Disciplines**

- · First-in-first-out (FIFO) + drop-tail
  - · Simplest choice used widely in the Internet
  - · FIFO means all packets treated equally
  - · Drop-tail: new packets gets dropped when queue is
  - · Important distinction:
    - · FIFO: scheduling discipline
    - · Drop-tail: drop policy
- · Alternative is to do Active Queue Management
  - · To improve congestion response
  - · Support fairness in presence of non-TCP flows
  - · To give flows different types of service QoS



# **Alternative Drop Policies**

- · Avoid lockout and full queue problems
- · Random drop and drop front policies
  - Drop random packet or packet at the head of the queue if the queue is full and a new packet arrives
- · Solve the lock-out problem but not the full-queues problem
- · May trigger congestion response faster
- Random Early Discard (RED) and Explicit Congestion Notification (ECN) slow down receivers before queues are full
- · RED: drop some packets before queue is full
- ECN: mark a bit in the headers to notify receiver (who notifies the sender) of congestion onset without dropping a packet



# Problems in Achieving fairness

- In the Internet, fairness is only achieved if all flows play by the same rules
  - But it is complicated: fairness is poorly defined for short flows, many versions of TCP co-exist, etc.
- In practice: most sources must use TCP or be "TCP friendly"
  - · Most sources are cooperative
- · Most sources implement homogeneous/compatible control law
  - · Compatible does not mean identical
  - · Typically means less aggressive than TCP
- · What if sources do not play by the rule?
  - E.g., TCP versus UDP without congestion control



#### Fairness Goals In Practice

- · Allocate resources fairly
- · Partially achieved by using similar congestion control rules
- · Isolate ill-behaved users
- · This is challenging
- How about users who start with a large initial congestion window
- How about UDP flows (good news: uncommon)
- · How about users who modify TCP (good news: very hard)
- · Still achieve statistical multiplexing
  - · One flow can fill entire pipe if no contenders
  - Work conserving → scheduler never idles link if it has a packet



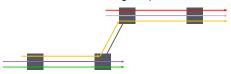
#### What is Fairness?

- · At what granularity?
- · Flows, connections, domains?
- · What if users have different RTTs/links/etc.
  - · Should it share a link fairly or be TCP fair?
- · Maximize fairness index?
  - Fairness =  $(\Sigma x_i)^2/n(\Sigma x_i^2)$  0<fairness<1
- · Basically a tough question to answer!
- · Good to separate the design of the mechanisms from definition of a policy
  - · User = arbitrary granularity
- · One example: max-min fairness



# Max-min Fairness

- Give users with "small" demand what they want, evenly divide unused resources to "big" users
- Formally:
  - · Resources allocated in terms of increasing demand
  - · No source gets resource share larger than its demand
  - · Sources with unsatisfied demands get equal share of resource





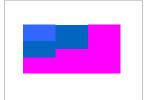
# Implementing Max-min Fairness

- · Generalized processor sharing
- Fluid fairness
- Bitwise round robin among all queues
- · Why not simple round robin?
  - Variable packet length → can get more service by sending bigger packets
  - · Unfair instantaneous service rate
    - · What if packets arrive just before/after packet departs?
- · We will use bit-bit round robin as an example
  - · Many other algorithms exist



# Bit-by-bit RR Illustration

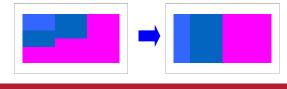
- Send one bit for every flow that has data queued – perfect!
- ... but not feasible to interleave bits on real networks
- FQ simulates bit-by-bit RR

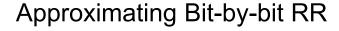




# Fair Queuing

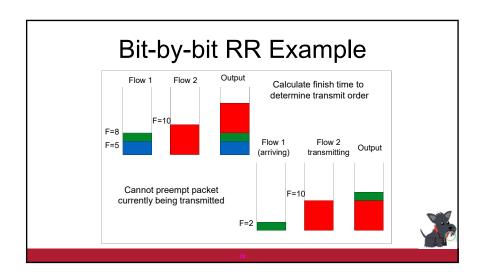
- · Mapping bit-by-bit schedule onto packet transmission schedule
- Transmit packet sequentially but in bit RR order
  - · How do you compute this packet order?
  - · Must be efficient and work for any order

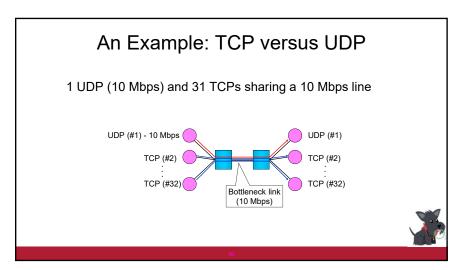


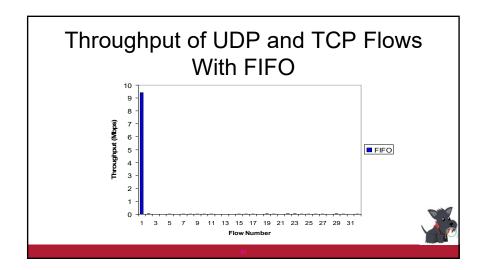


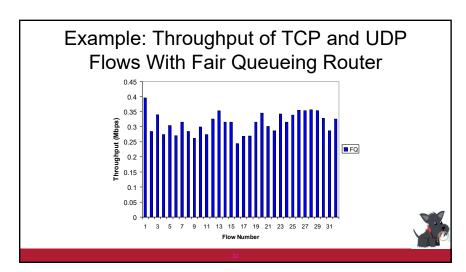
- Single flow: clock ticks when a bit is transmitted. For packet i:
  - A<sub>i</sub> = arrival time, S<sub>i</sub> = transmit start time,
     P<sub>i</sub> = transmission time, F<sub>i</sub> = finish transmit time
  - $F_i = S_i + P_i = \max(F_{i-1}, A_i) + P_i$
- Multiple flows: clock ticks when a bit from all active flows is transmitted → round number
- Models the fact that you would transmit one bit from each flow in bit RR
- Can now calculate F<sub>i</sub> for each packet if number of flows is know at all times – determines packet order
  - · Need to know flow count to calculate clock tick time





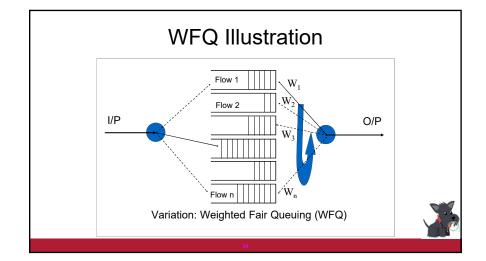






# Fair Queuing Tradeoffs

- Complex computation
- · Overhead of classification and scheduling
- · Must keep queues sorted by finish times
- · Computation changes whenever the flow count changes
- · Complex state must keep queue per flow
  - · Hard in routers with many flows (e.g., backbone routers)
  - Flow aggregation is a possibility (e.g. do fairness per domain)
- FQ can control congestion by monitoring flows
- Weighted fair queuing can give flows a different fraction of the bandwidth controlled by a weight W<sub>i</sub>
  - Bandwidth of flow i is  $W_i / \sum W_j$

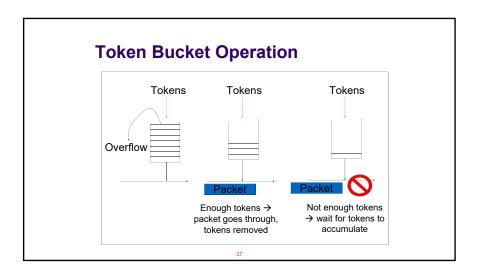


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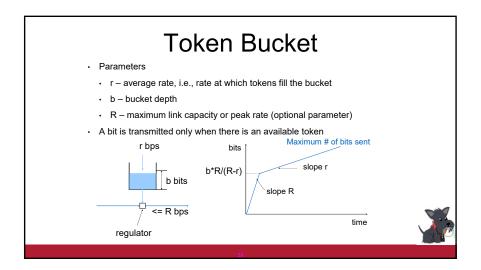
# Traffic Enforcement: Token Bucket Filter Tokens enter bucket at rate r Operation: Bucket depth b: capacity of bucket Sending a packet of size P uses P tokens If bucket has P tokens, packet sent at max rate, else must wait for tokens to accumulate

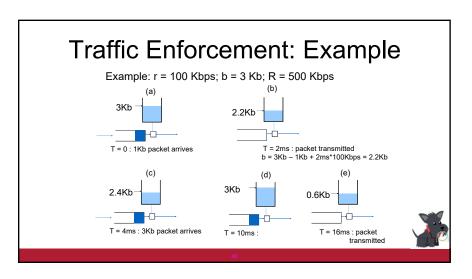


#### **Token Bucket Characteristics**

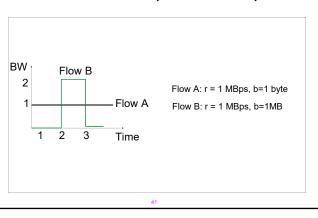
- Can <u>characterize</u> flow using a token bucket: smallest parameters for which no packets will be delayed
- · On the long run, rate is limited to r
- · On the short run, a burst of size b can be sent
- Maximum amount of traffic that can enter the network in time interval T is bounded by:
  - Simple case: Traffic = b + r\*T
- · Information useful to admission algorithm







#### Token Bucket Specs - Example



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# **Integrated Services Traffic Classes**

- · IETF RFC 1633 (1994)
- · Guaranteed service
- · For hard real-time applications
- · Fixed guarantee rate, assuming clients send at agreed-upon rate
- Predicted service
  - · For delay-adaptive applications
  - · Two components
    - · If conditions do not change, commit to current service
    - If conditions change, take steps to deliver consistent performance (help apps minimize playback delay)
    - · Implicit assumption network does not change much over time
- · Datagram/best effort service
- Also includes Resource reSerVation Protocol (RSVP) for establishing paths; may also need routing support



#### Lessons

- What type of applications are there? → Elastic, adaptive real-time, and hard real-time.
- Why do we need admission control → to maximize utility
- · How do token buckets + WFQ provide QoS guarantees?

