



# 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

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# **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 is "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 too many flows share 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 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	
<ul> <li>If U(bandwidth) is concave</li> <li>→ elastic applications</li> </ul>	Elastic
<ul> <li>Incremental utility is decreasing with increasing bandwidth</li> </ul>	
<ul> <li>It is always advantageous to have more flows with lower bandwidth</li> </ul>	BW
<ul> <li>Increases total utility of flows served</li> </ul>	
<ul> <li>No need of admission control</li> </ul>	
This is why the Internet works!	
<ul> <li>Not so for delay-adaptive and real-time applications</li> </ul>	

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#### Overview

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



## **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

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# Approximating Bit-by-bit RR

- · Single flow: clock ticks when a bit is transmitted. For packet i:
  - $A_i$  = arrival time,  $S_i$  = transmit start time,  $P_i$  = transmission time,  $F_i$  = 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
- FYI Only 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  $W_{\rm i}$ 
  - Bandwidth of flow i is  $W_i / \sum W_j$



# Overview • What is QoS? • Queuing discipline and scheduling • Traffic Enforcement • Integrated services





### **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







#### 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

