

#### Lecture 10

#### Quality of Service in IP Networks

ELEC 3506/9506

Communication Networks

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## Topics of the day

- Introduction to Quality of Service
- Flow Characteristics
- QoS Metrics
- QoS Principles
- Traffic Management
- Resource Management



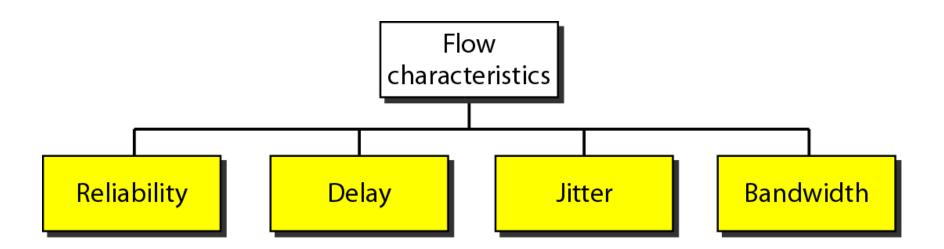
# QoS (Quality of Service) in IP Networks

- What is Network QoS?
  - Description or measurement of the overall performance of the network
  - The Internet was originally designed for best-effort service (elastic traffic)
  - A set of techniques and mechanisms that guarantee the performance of the network to deliver predictable service
  - We may loosely define quality of service as something a flow seeks to attain.



#### Flow Characteristics

 Traditionally, the following characteristics are attributed to a flow.





## Reliability

- Lack of reliability means losing a packet or acknowledgment, which entails retransmission.
- However the sensitivity of application programs to reliability is not the same.
  - Reliability is more important for email, file transfer, web browsing.
  - Reliability is less important for telephony and audio/video conferencing.



### Delay

- Source to destination delay is another characteristic.
- This consists of:
  - Sum of delays at routers
  - Propagation delay in the media
  - Set up mechanisms
- Delay tolerance is also application specific
  - Min delay tolerant telephony, audio/video conferencing, remote log-in
  - High delay tolerant file transfer and e-mail



#### **Network Delay**

- Consists of: Transmission delay, Propagation delay, Processing delay, and Queuing delay
  - Transmission Delay = (Packet length) / (Transmission rate)
  - Propagation Delay = (Distance) / (Propagation speed)
  - Processing Delay = Time required to process a packet in a router or a destination host
  - Queuing Delay = The time a packet waits in input and output queues in a router



#### **Jitter**

- Variation of delay for packets in the same flow.
- Low jitter
  - Difference is small
- High jitter
  - Difference between delays is large
- Variation of delay is not acceptable for audio and video applications.
  - Except with sufficient application buffering



#### Bandwidth / Capacity

- Maximum rate at which data can be transferred over a given link.
- Measured in bits per second (bps).
- Bandwidth / Capacity requirements.
  - Application specific.
  - Flow specific.



### QoS Metrics

- Throughput and Goodput
- Packet Loss
- Delay
- Jitter



## Throughput

- The terms bandwidth and throughput are not same.
- Bandwidth is the maximum rate at which data can be moved over a given link.
- Throughput, sometimes called effective bandwidth, is the actual speed over a link or circuit.
- Throughput is less than the potential bandwidth.
- For example, we may have a link with bandwidth of 1 Mbps, but the throughput may be 200 kbps if the devices connected to the link may handle only 200kbps.



#### Goodput

- Goodput is the application level throughput
  - Number of useful/correct bits per unit of time
  - From source address to the certain destination
  - Excluding protocol overhead (E.g. Headers)
  - Excluding retransmitted data packets.
- Goodput < Throughput due to:</p>
  - Protocol overhead
  - Transport layer flow control and congestion avoidance
  - Retransmission of lost or corrupt packets



#### Packet Loss

- Some packets fail to reach the destination.
- Packet loss can be caused by a number of factors:
  - Signal degradation
  - Channel congestion
  - Corrupted packets rejected in-transit
  - Faulty networking hardware / routing configurations
- Packet recovering protocols TCP
- Acceptable packet loss
  - Packet loss does not always indicate a problem



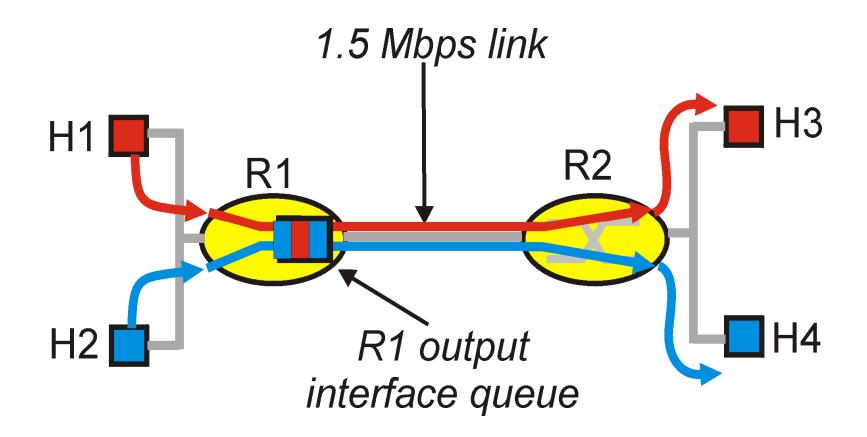
#### **Service Classes**

ID	Description	Priority	Duration (s) (min-max)		Data rate (kb/s) (min-max)		BER (min-max)		Delay (ms) (min-max)	
SC1	Large files exchange	8	50MB	500MB	1000	50000	1,00E-06	1,00E-06	200	
SC2	HQ video streaming	6	300	600	2000	40000	1,00E-09	1,00E-09	200	
SC3	LAN access and file service	4	120	300	500	50000	1,00E-06	1,00E-06	100	200
SC4	Interactive ultra high media	1	120	500	1000	50000	1,00E-03	1,00E-06	20	100
SC5	Lightweight brows- ing	5	300	900	64	512	1,00E-06	1,00E-06	200	
SC6	Data and media telephony	2	60	120	64	512	1,00E-03	1,00E-06	100	200
SC7	Simple telephony and messaging	3	10	120	8	64	1,00E-03	1,00E-06	100	200
SC8	Multimedia messaging	7	5	15	8	64	1,00E-06	1,00E-09	200	



# Sharing and Congestion Studies

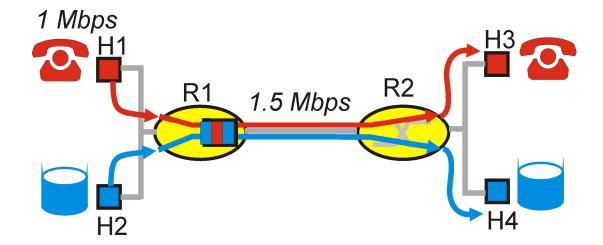
Simple model





#### Principles of QoS Guarantees

- Example: 1Mbps IP phone and FTP share 1.5 Mbps link.
  - Bursts of FTP can congest router, cause audio loss
  - Want to give priority to audio over FTP



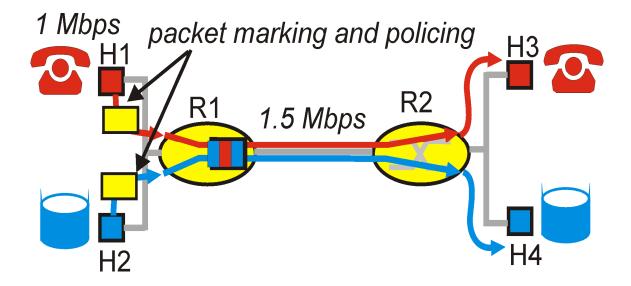
 Principle 1: Packet marking/classification needed for router to distinguish between different classes; and new router policy to treat packets accordingly.

**Different Traffic Classification** 



# Principles for QoS Guarantees (more)

- What if applications misbehave (audio sends higher than declared rate)
  - Policing: force source adherence to bandwidth allocations



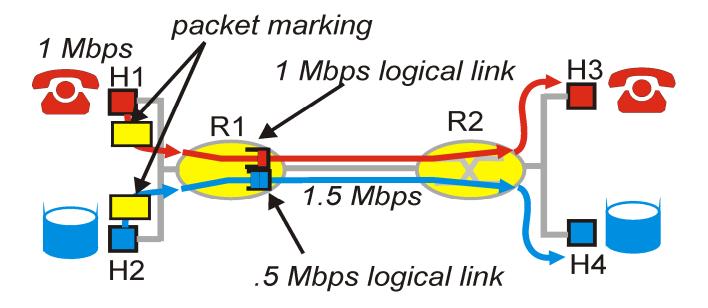
Principle 2: Provide protection (isolation) for one class from others.

Separate Resources for Different Traffic Classes



# Principles for QoS Guarantees (more)

Allocating fixed (non-sharable) bandwidth to flows: inefficient use
of bandwidth if flows doesn't use its allocation



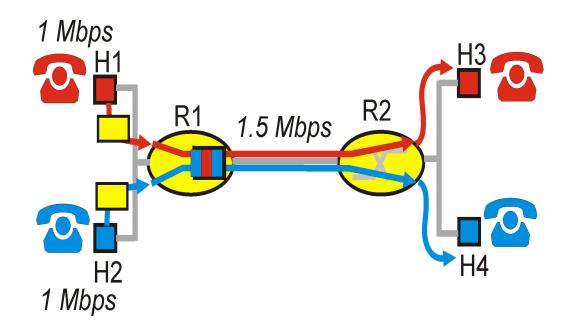
 Principle 3: While providing isolation, it is desirable to use resources as efficiently as possible.

Separate and Flexible Resources for Different Traffic Classes



# Principles for QoS Guarantees (more)

 Basic fact of life: can not support traffic demands beyond link capacity.

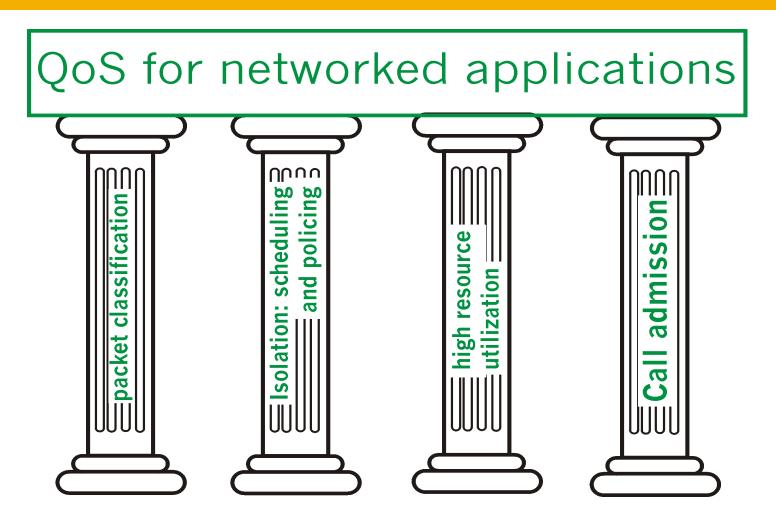


Principle 4: Call Admission: flow declares its needs, network may block call (e.g., busy signal) if it cannot meet needs.

Admission control to ensure sufficient separate and flexible resources for different traffic classes



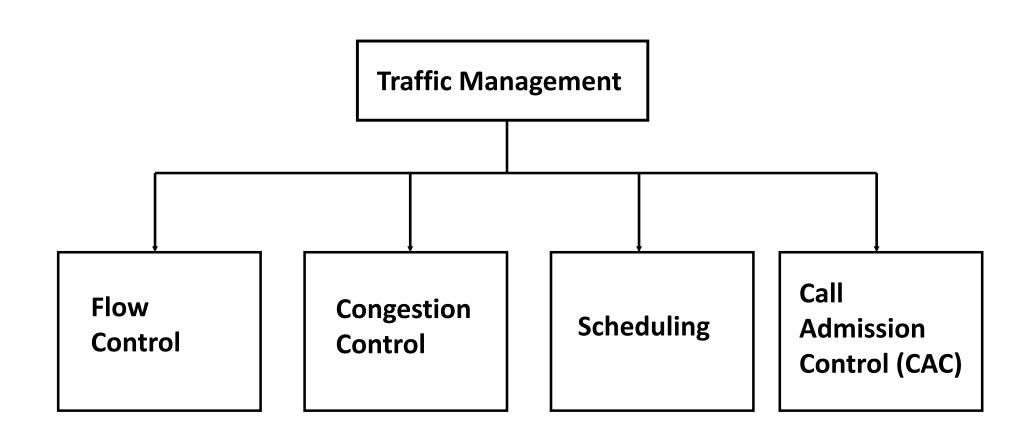
# Summary of QoS Principles



Let's next look at mechanisms for achieving this ...



### Traffic Management





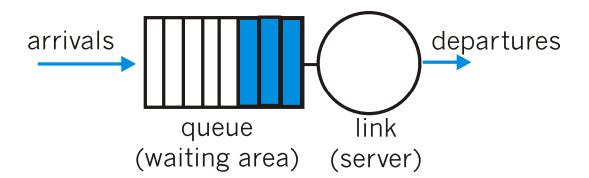
#### Scheduling Mechanisms

- Scheduling: Choose next packet to send on link
- Examples of Scheduling Mechanism:
  - FIFO- First in First Out
  - Priority Scheduling
  - Round Robin Scheduling
  - WFQ- Weighted Fair Queuing



#### Scheduling - FIFO

- FIFO (first in first out) scheduling: send in order of arrival to queue.
  - Discard policy: if packet arrives to full queue.
  - Who to discard?
    - Tail drop: Drop arriving packet
    - Priority: Drop/remove on priority basis
    - Random: Drop/remove randomly



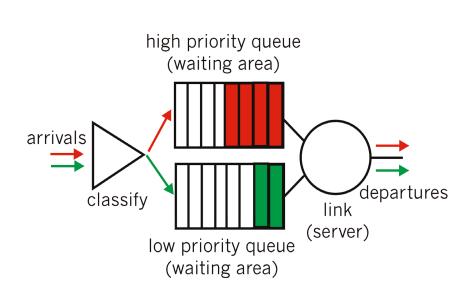


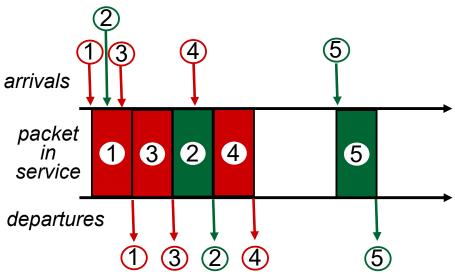
## Scheduling - Priority

Priority scheduling: Transmit highest priority queued packet

- Multiple classes, with different priorities
  - Class may depend on marking or other header info.

E.g. IP source/destination, port numbers, etc...





High priority: packet 1,3,4

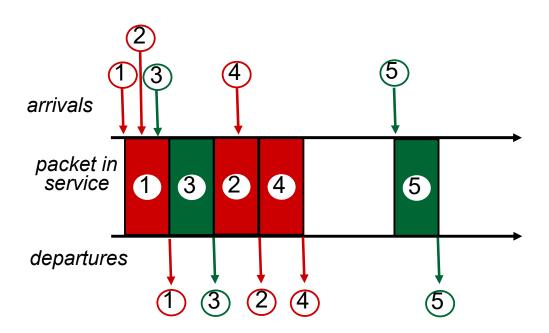
Low priority: packet 2, 5



## Scheduling – Round Robin

#### Round Robin Scheduling:

- Packets sorted into classes, but without priority class
- Cyclically scan class queues, serving one from each class (if available)
- Example:
  - Class 1: packet 1, 2, 4
  - Class 2: packet 3, 5



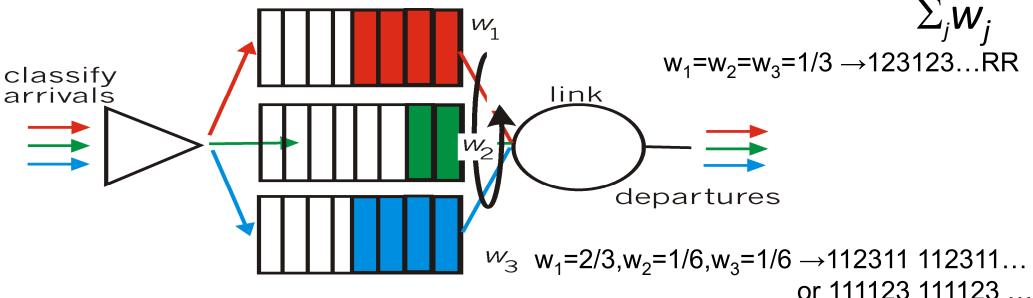


## Scheduling - WFQ

#### Weighted Fair Queuing:

- Generalized Round Robin
- each class, *i*, has weight,  $w_i$ , and gets weighted amount of service in each cycle:  $\frac{W_i}{\sum_i w_i}$

For a link rate of R, each class will then receive a rate of R  $\frac{W_i}{\sum_i W_i}$ 





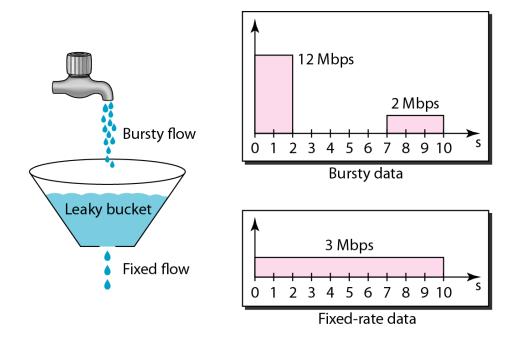
# Traffic Shaping

- Controls the amount and the rate of the traffic sent to the network.
- Two techniques can shape traffic:
  - Leaky bucket
  - Token bucket



# Leaky Bucket

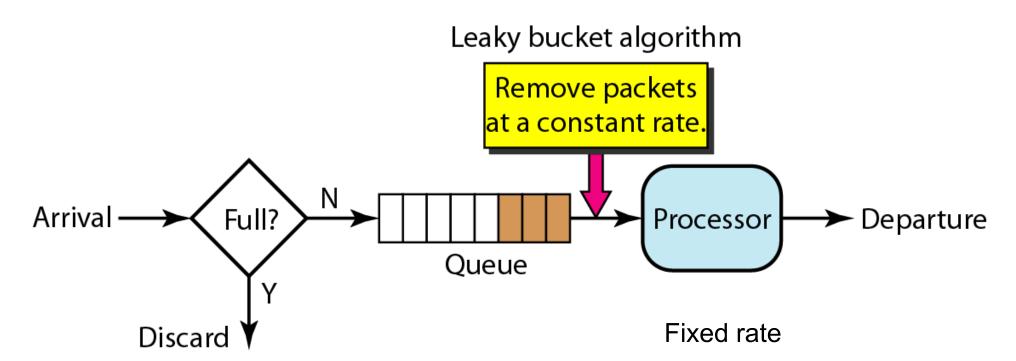
- Shapes bursty traffic into fixed-rate traffic by averaging the data rate
- It may drop packets if the bucket is full
- It also prevents congestion





#### Leaky Bucket Algorithm

This is a simple leaky bucket implementation



The leaky bucket is very restrictive. It does not credit an idle host. For example, if a host is not sending for a while, its bucket becomes empty → buffer/queue status unaware



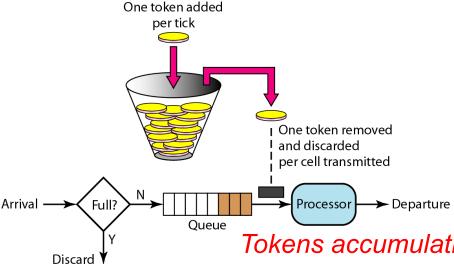
#### Token Bucket

- The leaky bucket imposes a hard limit on the data transmission rate.
- The token bucket allows a certain amount of burstiness while imposing a limit on the average data transmission rate.
- The token bucket allows bursty traffic at a regulated maximum rate.
- Allows idle hosts to accumulate credit for the future in the form of tokens → buffer/queue status aware



#### Token Bucket Algorithm

- Bucket contains tokens
- Tokens are added periodically at desired maximum average rate
- Packets can only be sent when there are tokens in the bucket.
- Tokens are removed when sending packets



#### **Analysis**

- Capacity of the bucket: c tokens
- Tokens rate entering bucket: r tokens/sec
- The system removes one token for every cell of data sent. The maximum number of cells that can enter the network during any time interval of length t is:

Maximum number of packets =  $r \times t + c$ 

 The maximum average rate for the token bucket (departure) is:

Maximum average rate =  $(r \times t + c)/t$ packets per second

Tokens accumulation when buffer is empty → queue status aware



#### Resource Management

- Flows need resources
  - Buffer, capacity, CPU time, etc.
  - Ensure availability: call admission
- QoS improved if these resources are guaranteed beforehand
- Resource-reservation models for QoS
  - Integrated Services
  - Differentiated Services



#### **Admission Control**

- A mechanism used by a router or switch to accept or reject a flow,
  - based on predefined parameters called flow specifications.
- Before a router accepts a flow for processing, it checks the flow specifications to see
  - If its capacity (in terms of bandwidth, buffer size, CPU speed, etc) can handle the new flow
  - If its previous commitments to other flows can handle the new flow.

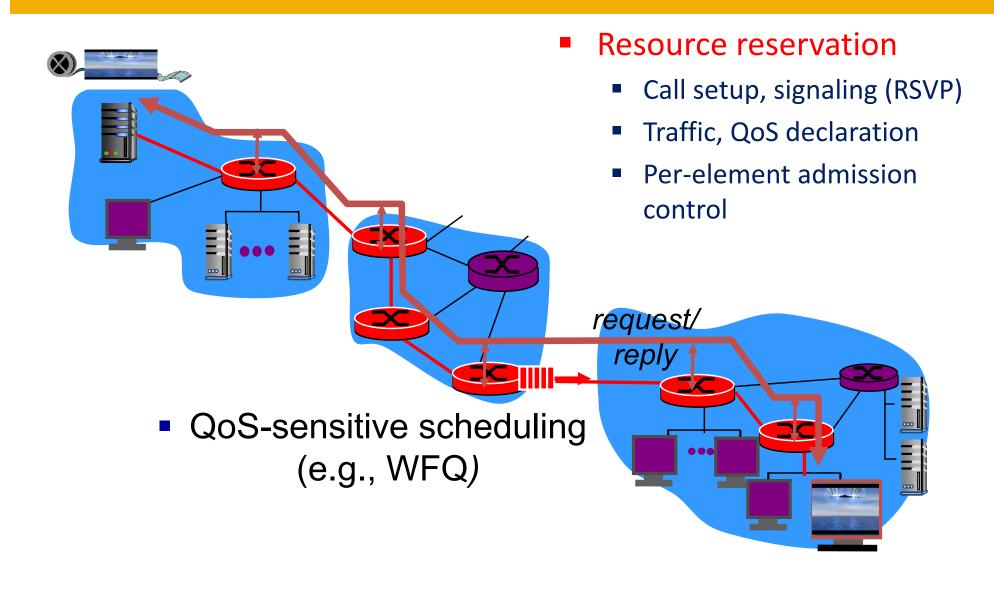


#### Integrated Services

- Traditional Internet only provides best-effort delivery for all users
- Some applications need a minimum bandwidth to function (e.g, real-time video)
- Integrated service (IntServ): architecture for providing QoS guarantees in IP networks for individual application sessions/flows (flow-based).
- Resources are explicitly reserved for a given data flow
  - Routers maintain state info of allocated resources, QoS requests for different flows.
- Admit/deny new call setup requests.



# Intserv: QoS Guarantee Scenario





#### Call Admission

#### Arriving session must:

- Declare its QoS requirement
  - R-spec: defines the QoS being requested (buffer, bandwidth, etc.)
- Characterize traffic it will send into network
  - T-spec: defines traffic characteristics of the flow (token bucket algorithm parameters)
- Signaling protocol: needed to carry R-spec and T-spec to routers (where reservation is required)
  - RSVP (Resource Reservation Protocol)



# Intserv QoS: Service Models/Classes

#### **Guaranteed service:**

- Designed for real-time traffic that needs a guaranteed minimum end-toend delay.
- Guarantees that the packets will arrive within a certain delivery time and are not discarded if flow traffic stays within the boundaries of *Tspec*.

#### Controlled load service:

 Designed for applications that can accept some delays, but are sensitive to an overload network and to the danger of losing packets.



#### Differentiated Services

- Concerns with *Intserv* 
  - Scalability: Router needs to keep information for each flow
  - Service Class Limitations: Only two types, guaranteed and control-load
- Differentiated Services (DiffServ)
  - Main processing moved from the core of the network to the edge of the network (Solves scalability problem)
  - Per-flow service changed to per-class service.
- IntServ is based on building virtual paths by using the bandwidth reservation technique
- DiffServ is based on traffic differentiation/priority without reservation



#### Recommended Reading

 J. F. Kurose and K. W. Ross, Computer Networking: A Top-Down Approach, 8<sup>th</sup> ed., 2022, Chapter 4