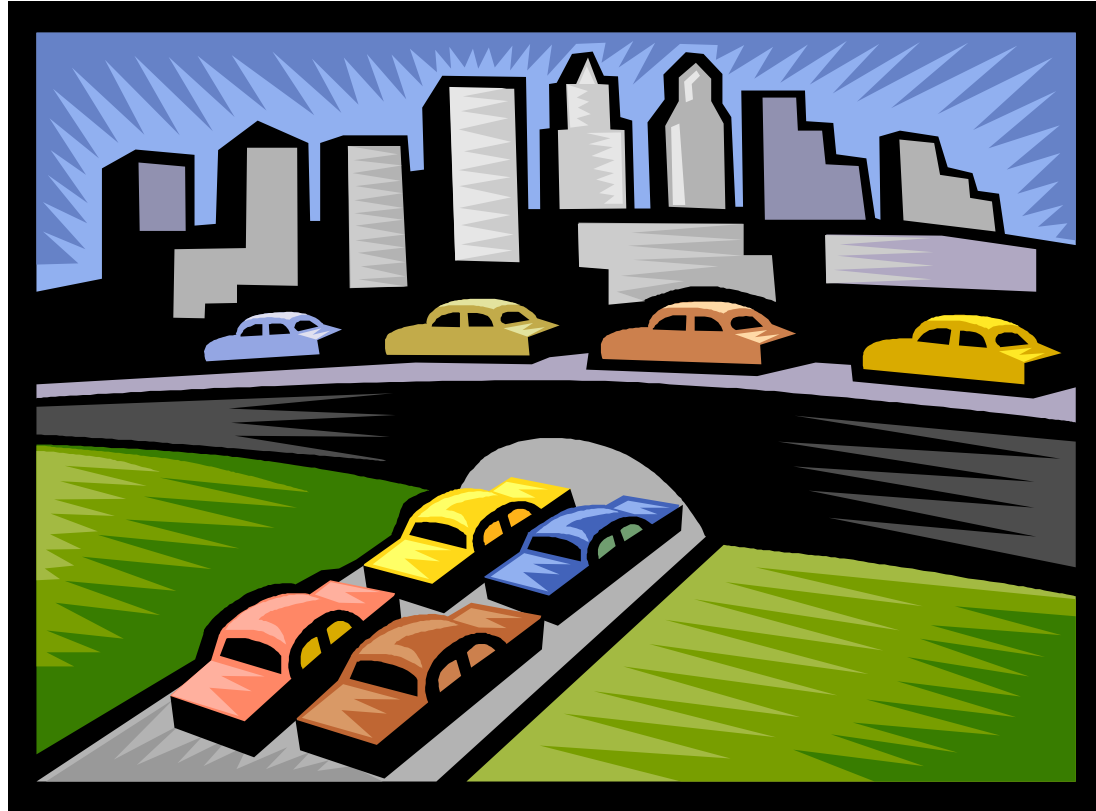


Principles of congestion control



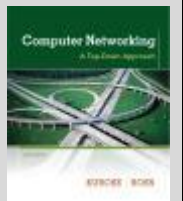
Computer Networking: A Top Down Approach

6th edition

Jim Kurose, Keith Ross

Addison-Wesley

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Chapter 3 outline

3.1 Transport-layer services

3.2 Multiplexing and demultiplexing

3.3 Connectionless transport: UDP

3.4 Principles of reliable data transfer

3.5 Connection-oriented transport: TCP

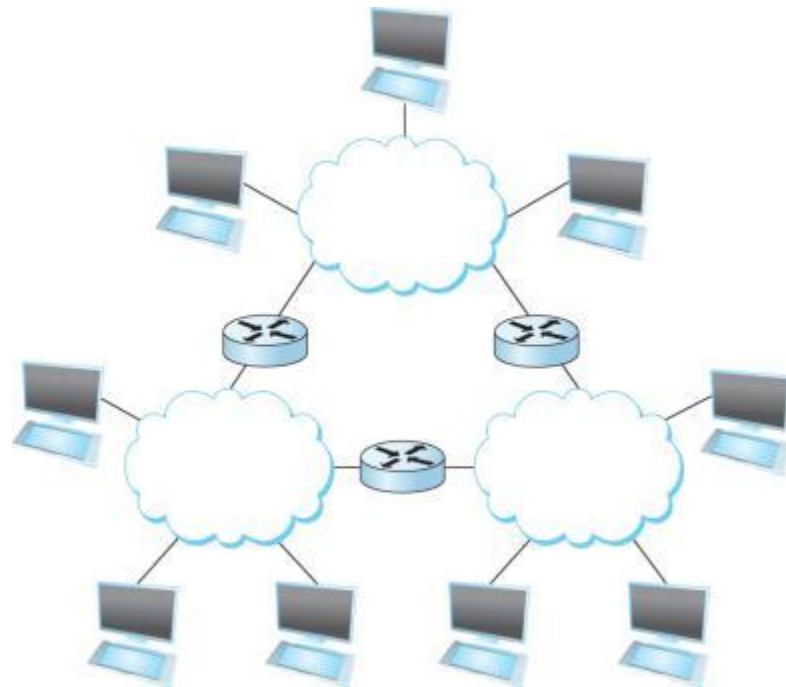
- Segment structure
- Reliable data transfer
- Flow control
- Connection management

3.6 Principles of congestion control

3.7 TCP congestion control

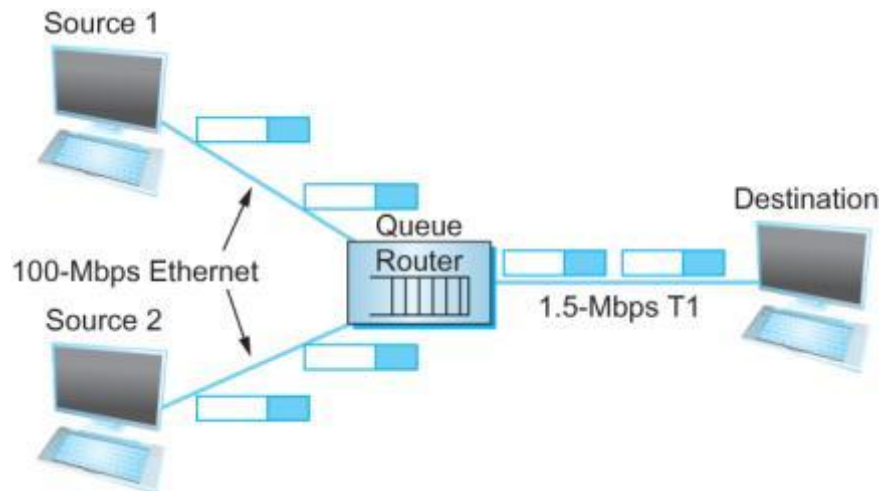
IP best-effort network

- Best-effort model
 - Everybody can send
 - Network does the best it can to deliver
 - Delivery not guaranteed, some traffic may be dropped



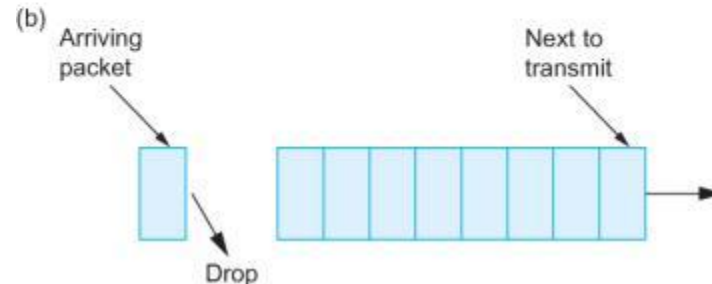
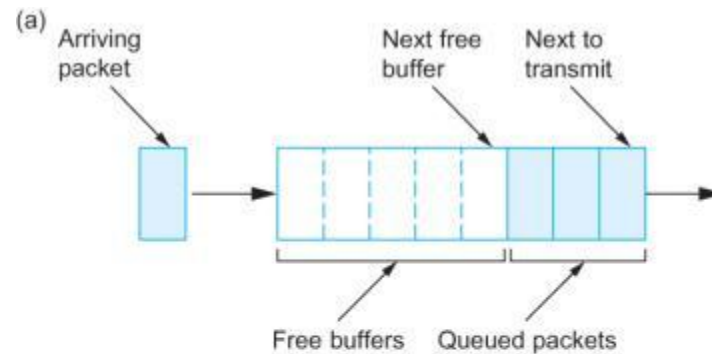
Congestion unavoidable

- Multiple packets arrive at same time
 - Router can only transmit one
 - Router has to buffer remaining
- If too many arrive in a short time window
 - Buffer may overflow
 - Router has to choose some packets to drop



What routers do

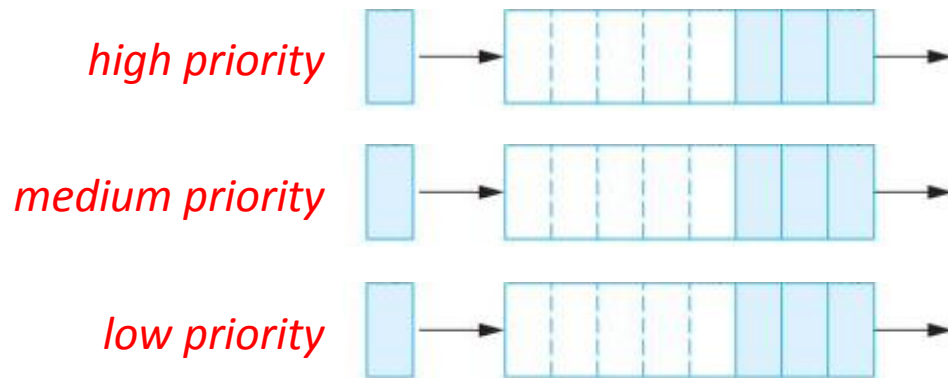
- Too many packets arrive too quickly
 - Which packets should we drop?
- First-in first-out (FIFO) with tail drop
 - Simple, drop the new guy that doesn't fit in your buffer



Queuing disciplines

- Priority queuing

- Packets marked with priority in header
- Multiple FIFO queues, one for each priority class
- Transmit high priority queues first
- Who is allowed to set priority bit?



Principles of congestion control

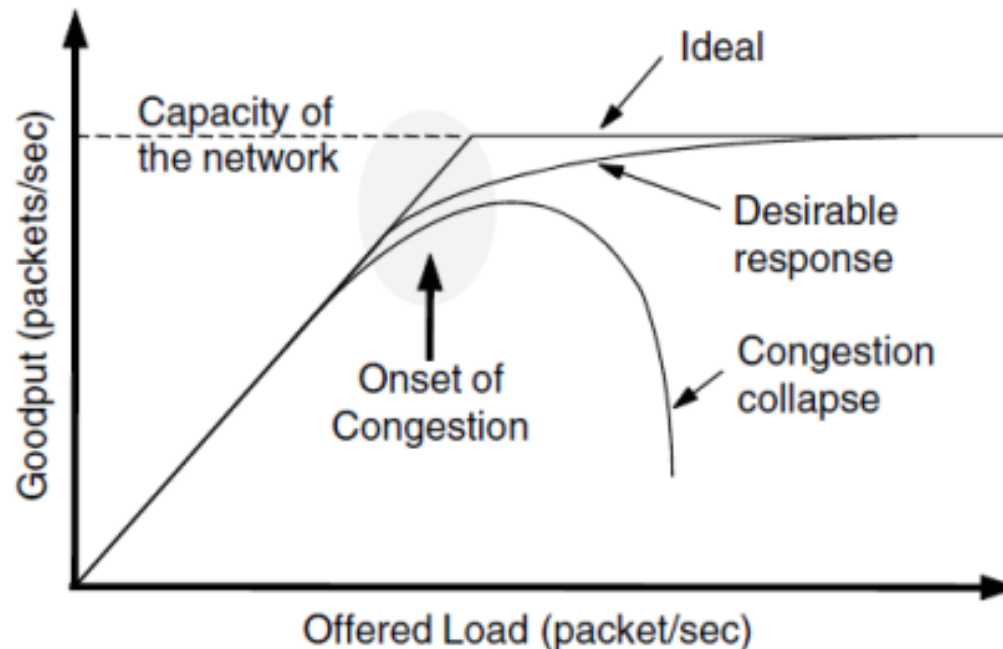
Congestion:

- Informally: "too many sources sending too much data too fast for *network* to handle"
- Different from flow control!
- Manifestations:
 - Lost packets (buffer overflow at routers)
 - Long delays (queueing in router buffers)
- A top-10 problem!

Congestion collapse

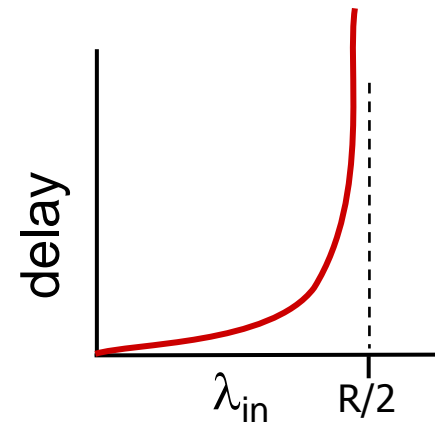
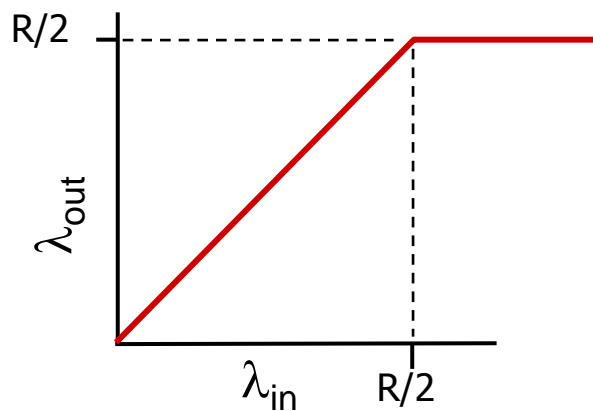
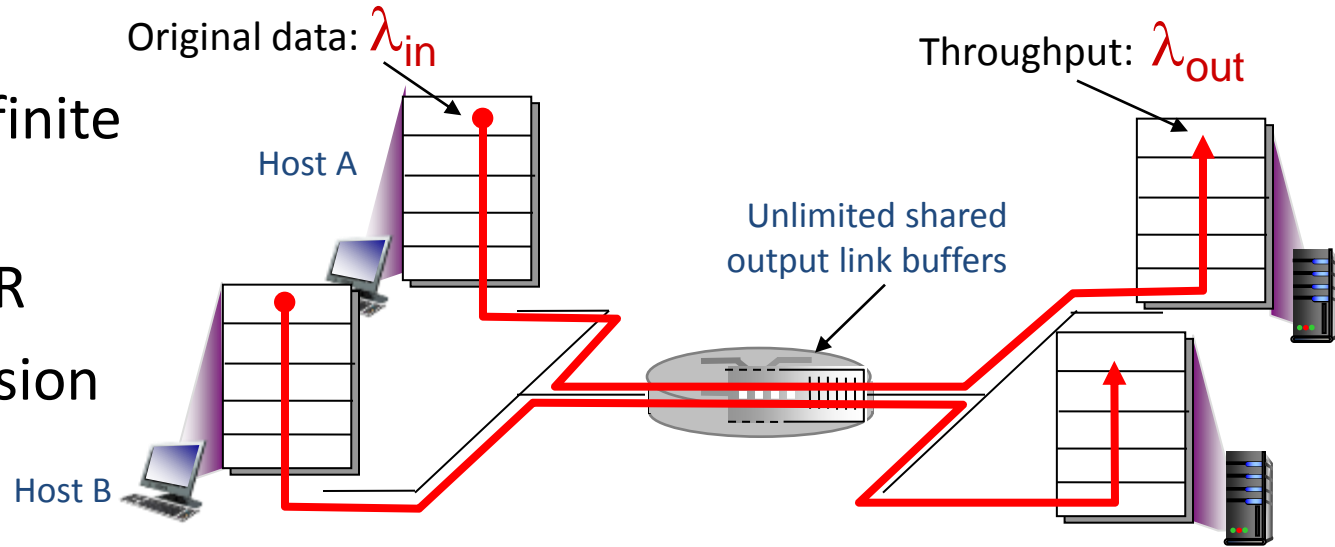
- Congestion collapse

- 1986, NSF backbone dropped from 32 kbps to 40 bps
 - Hosts send packets as fast as advertised window allowed
 - When packets dropped, hosts retransmit causing more congestion
- Goodput = useful bits delivered per unit time
 - Excludes header overhead, retransmissions, etc.



Causes/costs of congestion: scenario 1

- ❖ Two senders, two receivers
- ❖ One router, infinite buffers
- ❖ Link capacity: R
- ❖ No retransmission

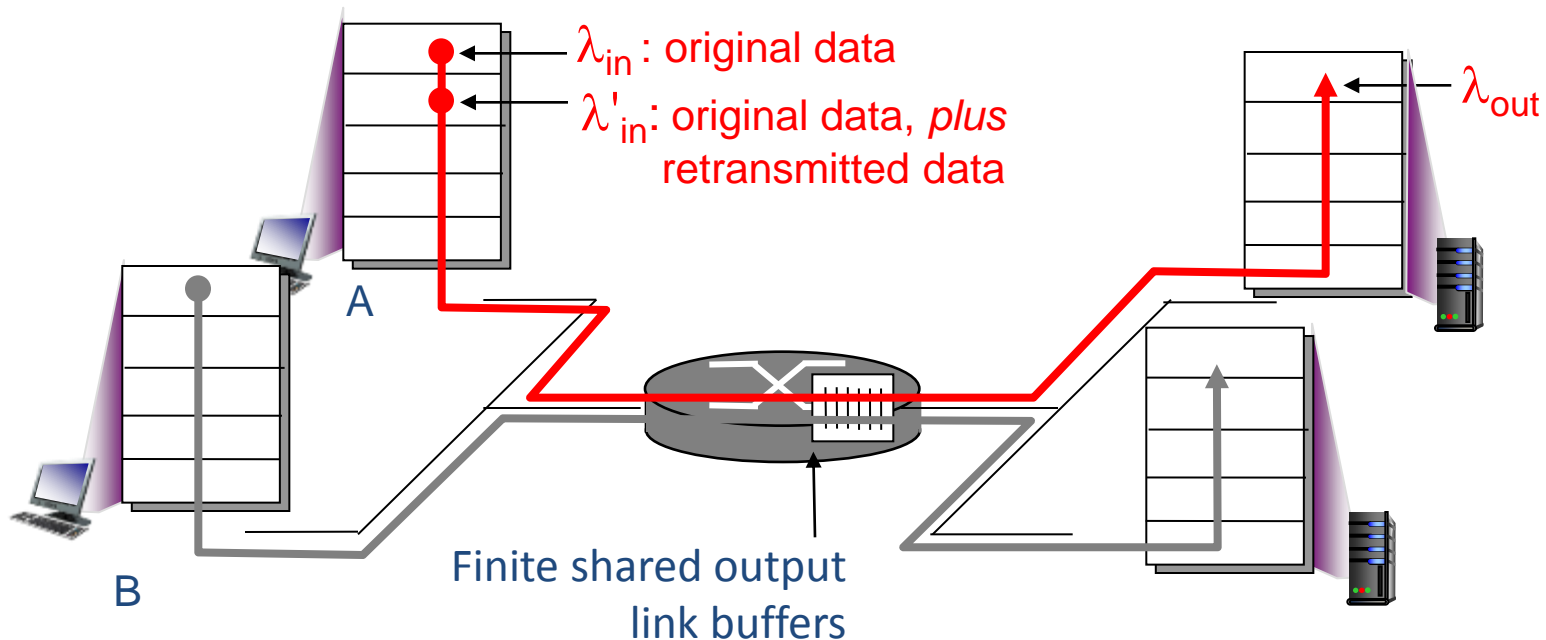


- ❖ Maximum per-connection throughput: $R/2$

- ❖ Large delays as arrival rate, λ_{in} , approaches capacity

Causes/costs of congestion: scenario 2

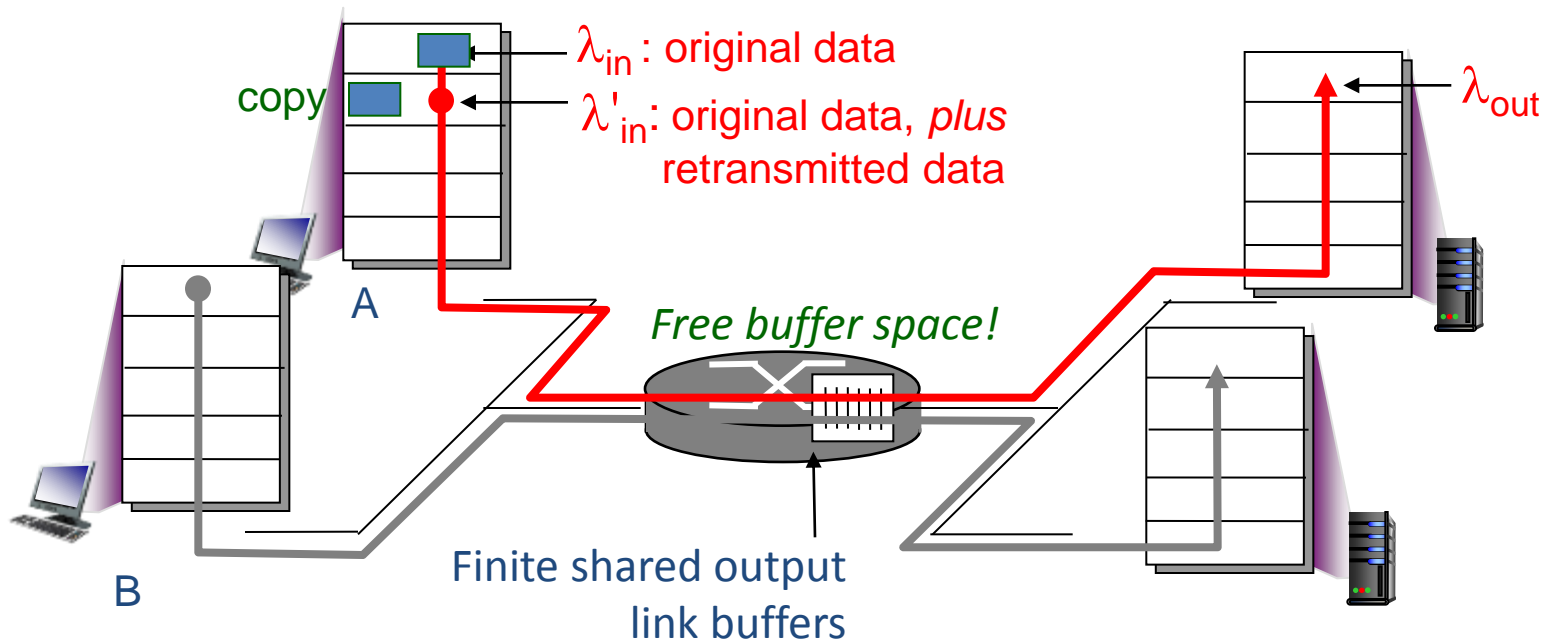
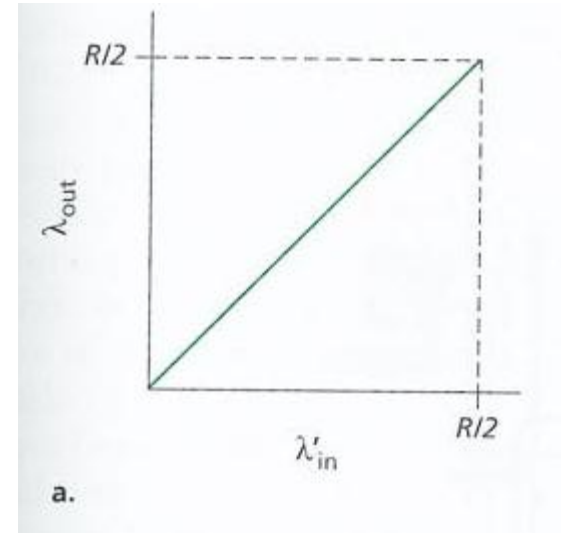
- ❖ One router, *finite* buffers, reliable connection
- ❖ Sender retransmission of timed-out packet
 - Application-layer input = application-layer output: $\lambda_{in} = \lambda_{out}$
 - Transport-layer input includes *retransmissions* : $\lambda'_{in} \geq \lambda_{in}$
 - $\lambda'_{in} = \textit{offered load}$



Congestion scenario 2a: ideal case

Idealization: *perfect knowledge*

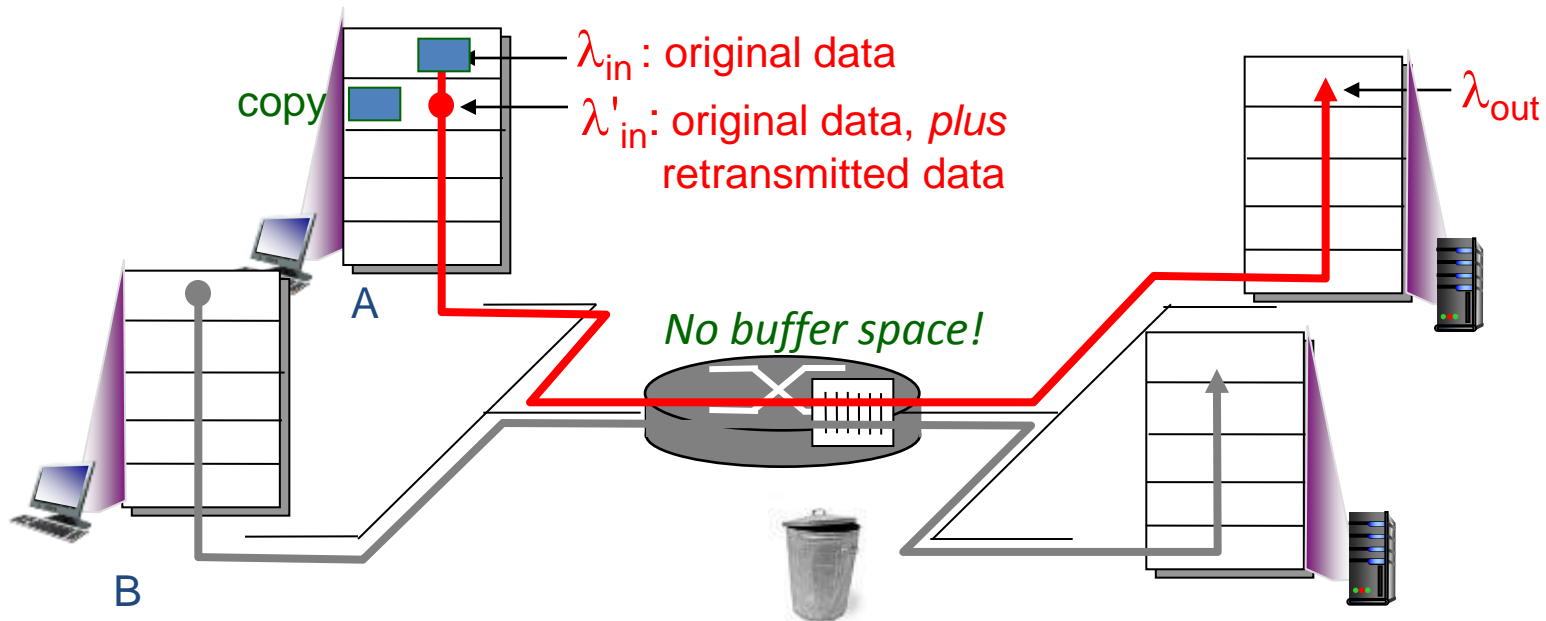
- ❖ Sender magically sends only when router buffers available
- ❖ No loss, $\lambda'_{in} = \lambda_{in}$
- ❖ Hosts won't send faster than $R/2$



Congestion scenario 2b: known loss

Idealization: *known loss*

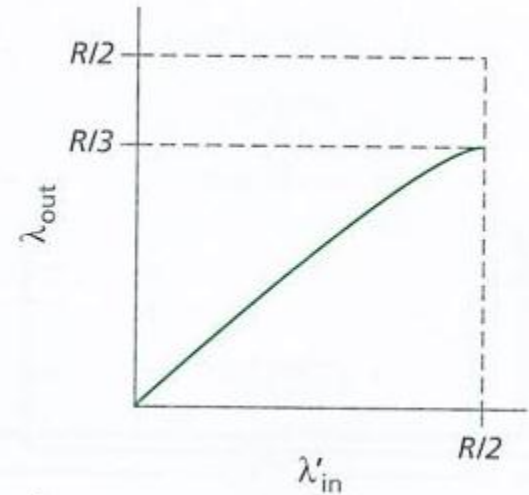
- ❖ Packets can be lost, dropped at router due to full buffers
- ❖ Sender only resends if packet *known* to be lost



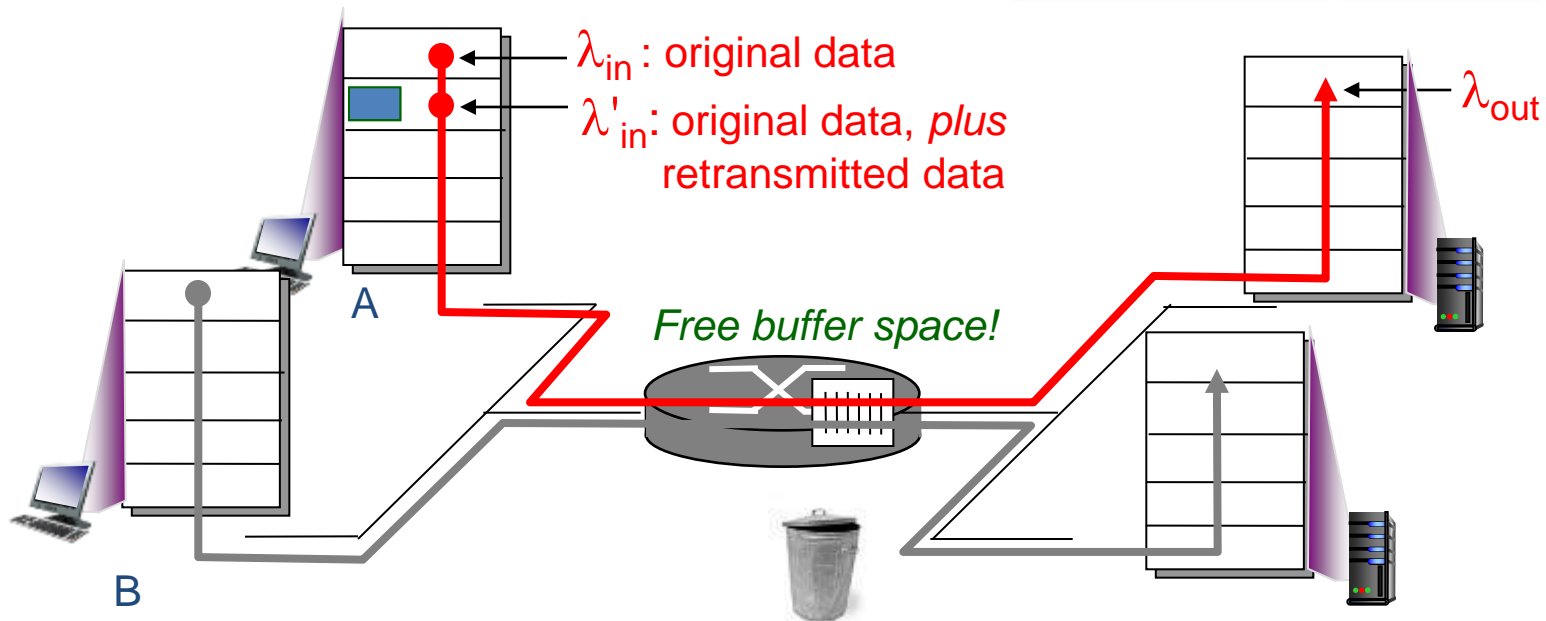
Congestion scenario 2b: known loss

Idealization: *known loss*

- ❖ Packets can be lost, dropped at router due to full buffers
- ❖ Sender only resends if packet *known* to be lost



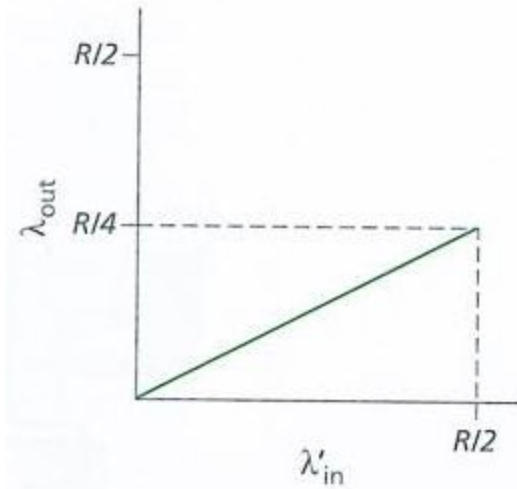
b.



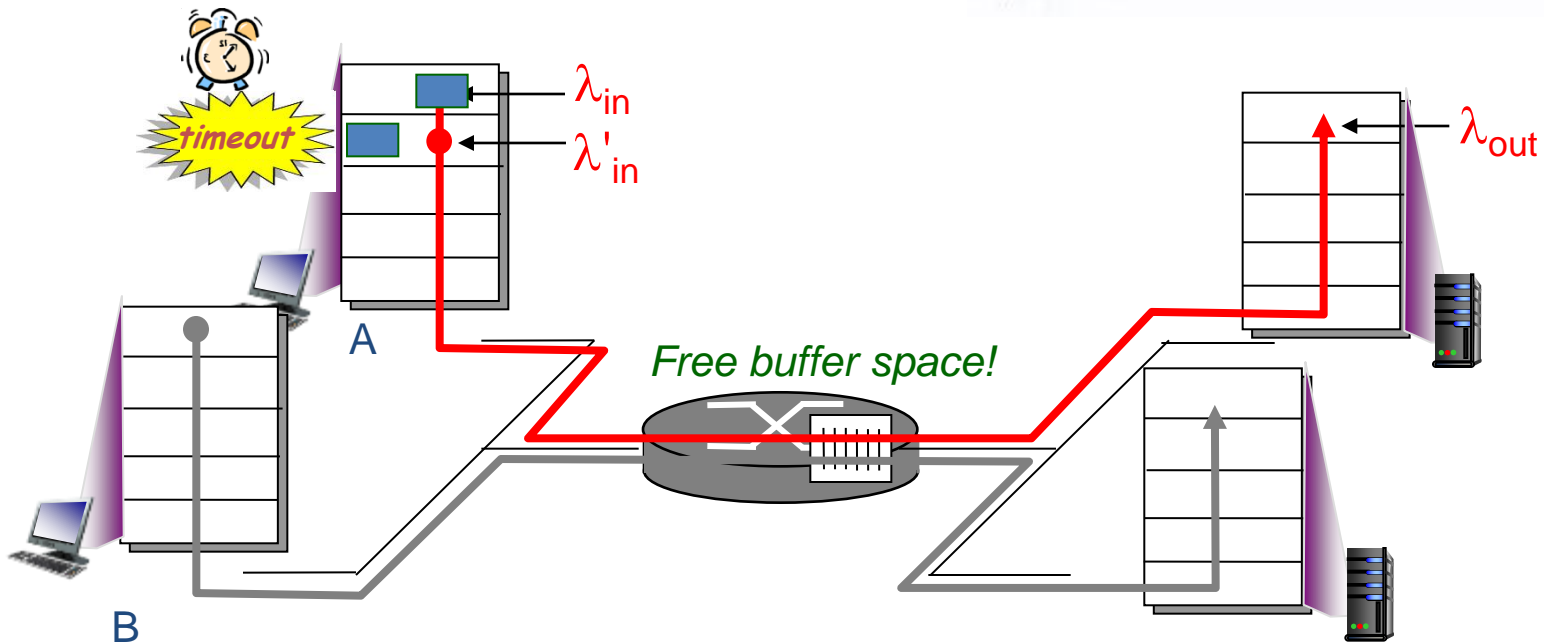
Congestion scenario 2c: duplicates

Realistic: *duplicates*

- ❖ Packets can be lost, dropped at router due to full buffers
- ❖ Sender times out prematurely, sending *two* copies, both of which are delivered



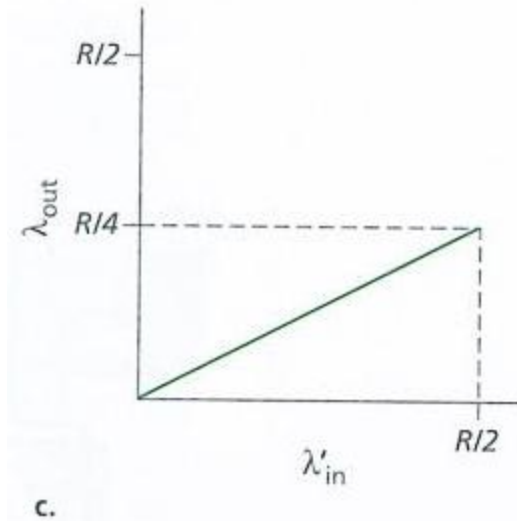
c.



Congestion scenario 2c: duplicates

Realistic: *duplicates*

- ❖ Packets can be lost, dropped at router due to full buffers
- ❖ Sender times out prematurely, sending *two* copies, both of which are delivered



Costs of congestion:

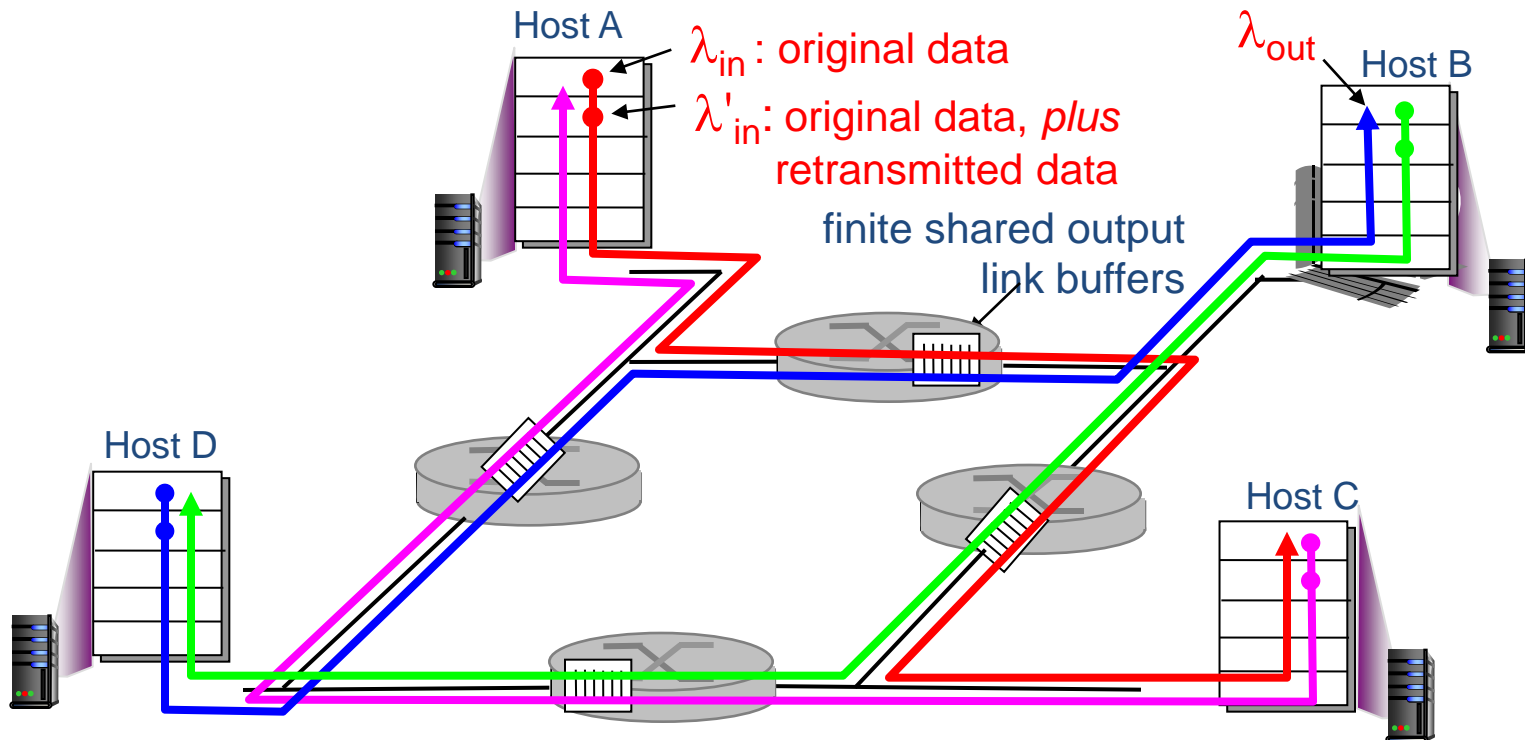
- ❖ More work (retransmissions) for given goodput
- ❖ Unneeded retransmissions
 - Link carries multiple copies of packet
 - Decreases goodput

Causes/costs of congestion: scenario 3

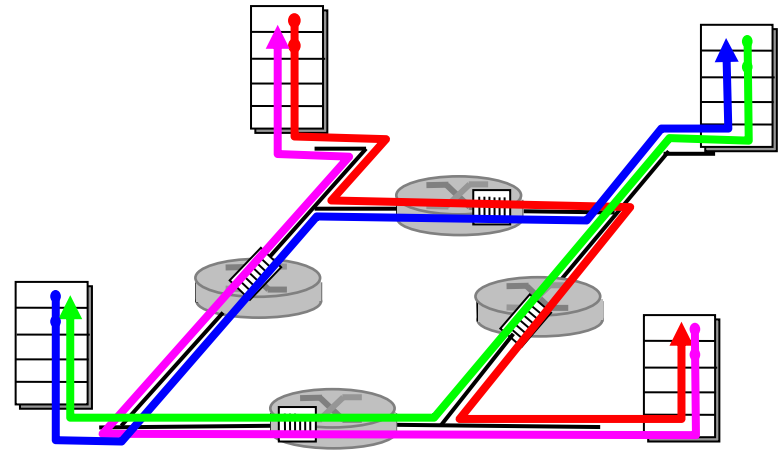
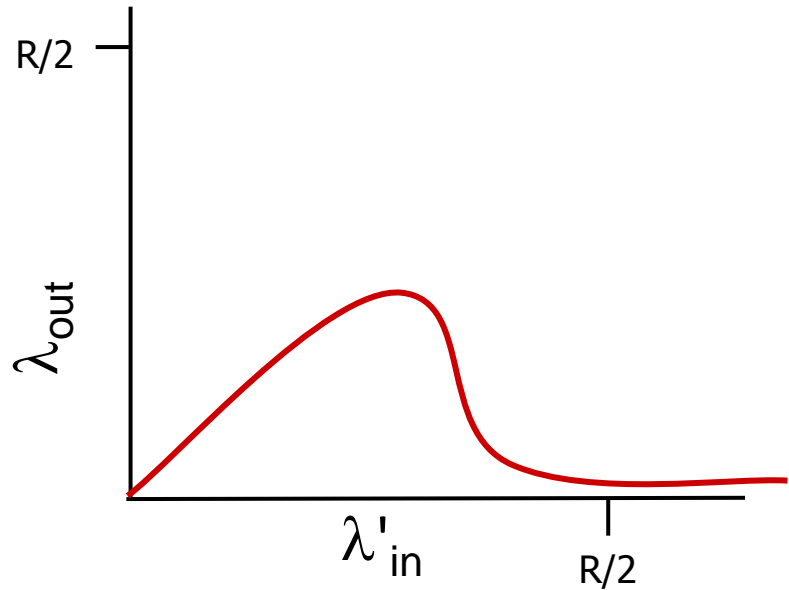
- ❖ Four senders
- ❖ Multihop paths
- ❖ Timeout/retransmit

Q: What happens as λ_{in} and λ'_{in} increase?

A: As red λ'_{in} increases, all arriving blue pkts at upper queue are dropped, blue throughput $\rightarrow 0$



Causes/costs of congestion: scenario 3



Another cost of congestion:

- ❖ When packet dropped, any upstream transmission capacity used for that packet was wasted!

Approaches to congestion control

Two broad approaches towards congestion control:

End-end:

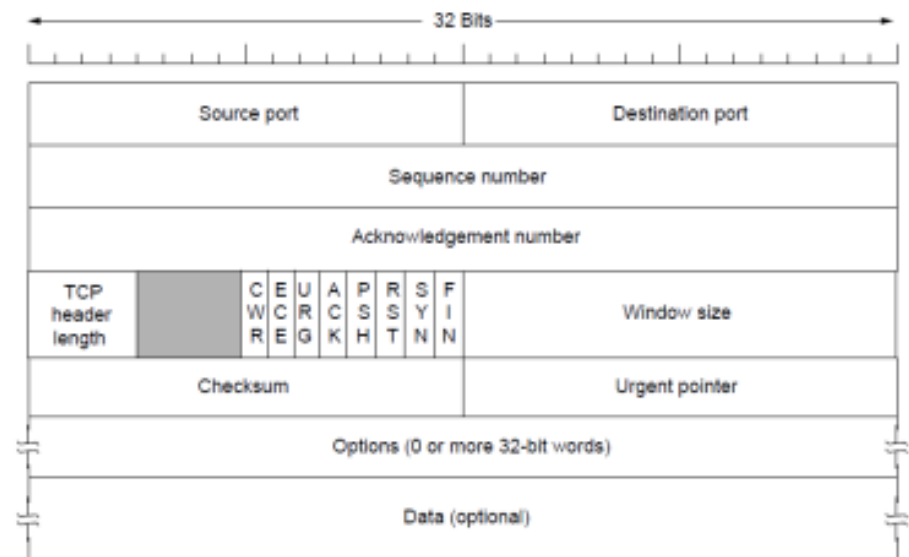
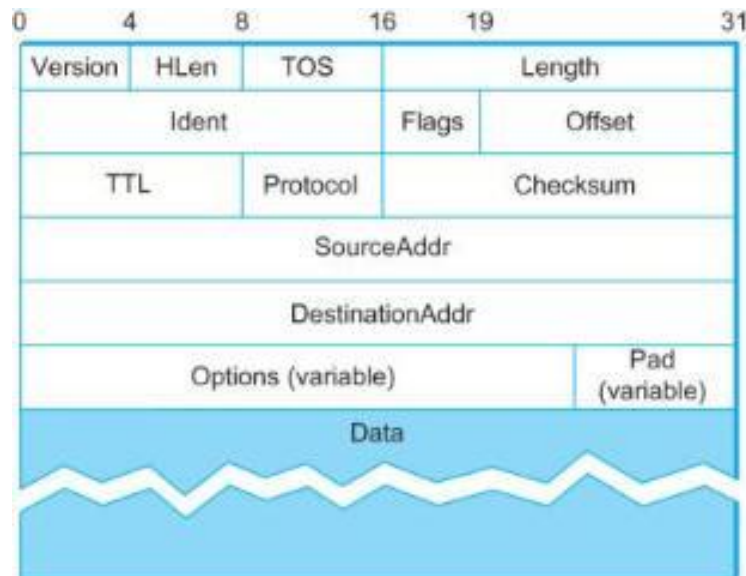
- ❖ No explicit feedback from network
- ❖ Congestion inferred from end-system observed loss, delay
- ❖ Approach taken by TCP

Network-assisted:

- ❖ Routers provide feedback to end systems
 - Single bit indicating congestion (SNA, DECbit, TCP/IP ECN, ATM)
 - Explicit rate for sender to send at

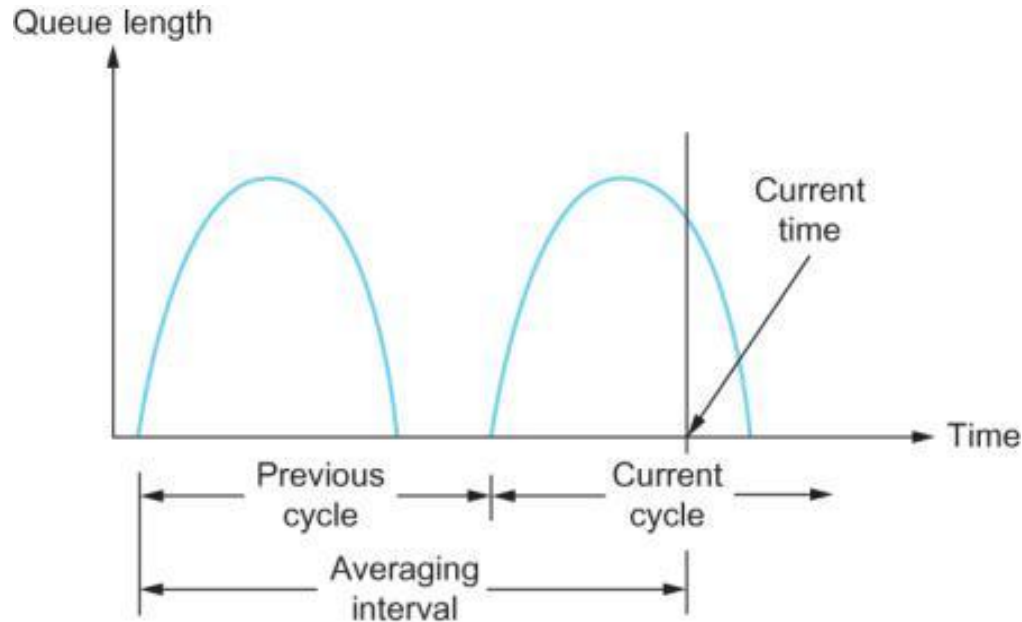
Router signaling

- **Explicit Congestion Notification (ECN)**
 - Sender sets TOS IP header bit saying it supports ECN
 - If ECN-aware router is congested, marks another TOS bit
 - TCP receiver sees IP congestion bit, informs sender via TCP segment ECN-Echo (ECE) bit
 - TCP sender confirms receipt of ECE with Congestion Window Reduced (CWR) bit



Router signaling

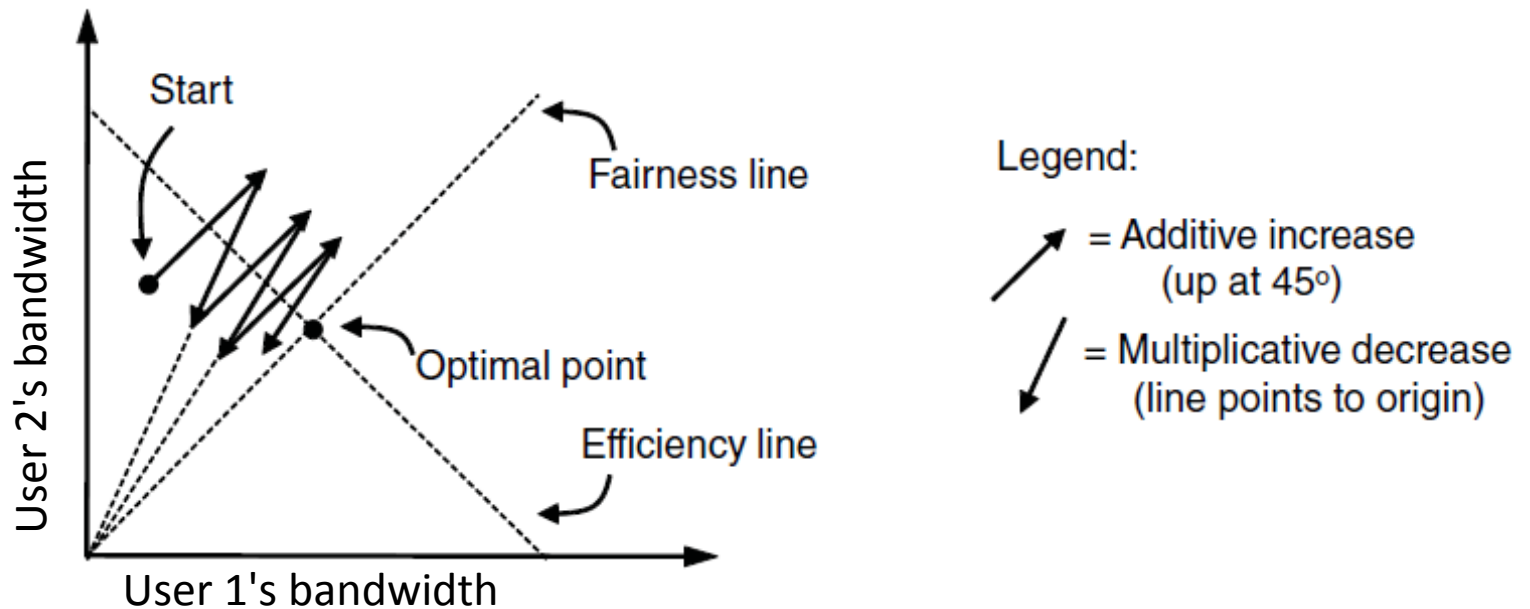
- How does router determine congestion?
 - Checks avg. queue length spanning last busy + idle cycle



- What does TCP sender do with congestion signals?
 - Checks fraction of last window's worth of packets
 - If $< 50\%$, increase congestion window
 - If $> 50\%$, decrease congestion window by 0.875

AIMD principle

- Additive increase, multiplicative decrease (AIMD)
 - Additive increase: On success of last packet, increase number of packets in-flight by one
 - Multiplicative decrease: On loss of packet, divide number of allowed in-flight packets in half



Summary

- Principles of congestion control
 - Too many senders can lead to congestion collapse
 - Links between routers have limited bandwidth
 - Router queues are finite
 - Traffic patterns are unpredictable
 - **Goodput** = useful bits delivered per unit time
 - Broad approaches
 - **End-to-end**, no information from routers
 - **Network assisted**, routers warn when congestion occurring (or about to)
 - AIMD principle
 - Two competing senders achieve efficiency & fairness