

**CS3640** 

# Transport Layer (3): Congestion Control

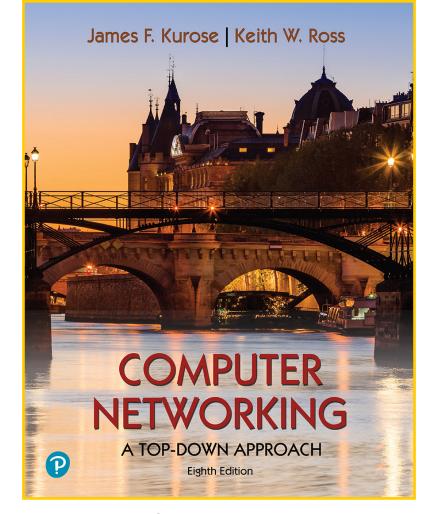
**Prof. Supreeth Shastri** 

Computer Science
The University of Iowa

## Lecture goals

continued discussion of reliable data transfer, followed by congestion control

- Pipelined RDT protocols
- Congestion control



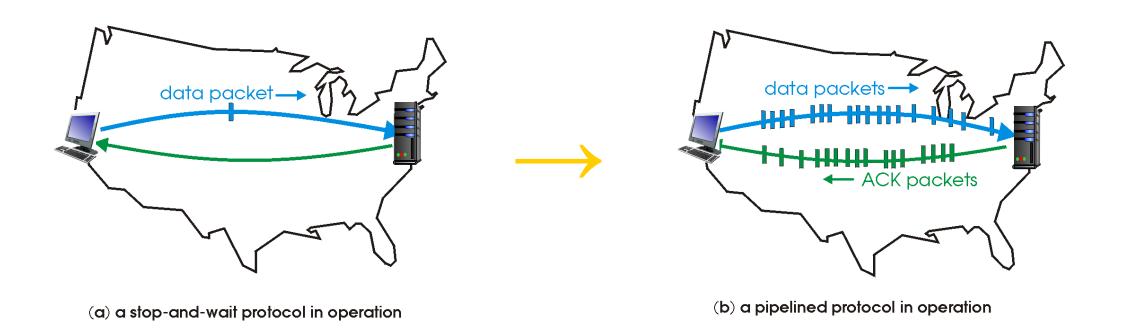
Chapter 3.4, 3.6



## Reliable Data Transfer: techniques and mechanisms

Checksum	detect bit errors in packet	
ACK	report reception of a packet correctly	
NAK	report error(s) in a received packet	
Sequence numbers	detect any missing packets	
Timers	detect and recover from packet loss	
Pipelining	increase channel utilization	

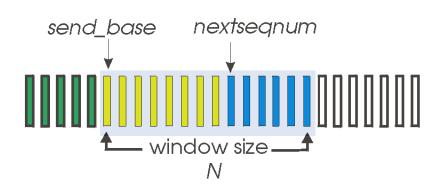
## Increased Utilization w/ Pipelining



#### Pipelined transfer of packets, where several could be in-transit and unacknowledged

- change-1: increase the range of sequence# in RDT protocols
- change-2: add buffer capability to both sender and receiver sides
- two variants: Go-Back-N and Selective Repeat

## Go-Back-N (GBN) Protocol



already ack'ed

sent, not yet ack'ed

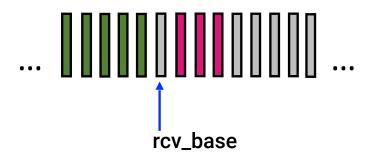
usable, not yet sent

not usable

#### **Sender**

- Packet window: sender defines a window of up to N consecutive transmitted but not yet ACKed packets
- Cumulative ACK: if the sender receives ACK(k), then it considers all packet up to k to be ACKed i.e., it moves the window forward to k+1
- Timer: sender always maintains a timer for the oldest not yet ACKed packet
- Retransmissions: Upon timer interrupt, sender retransmits the first not yet ACKed packet followed by all higher seq# packets in window

## Go-Back-N (GBN) Protocol



received and ACKed

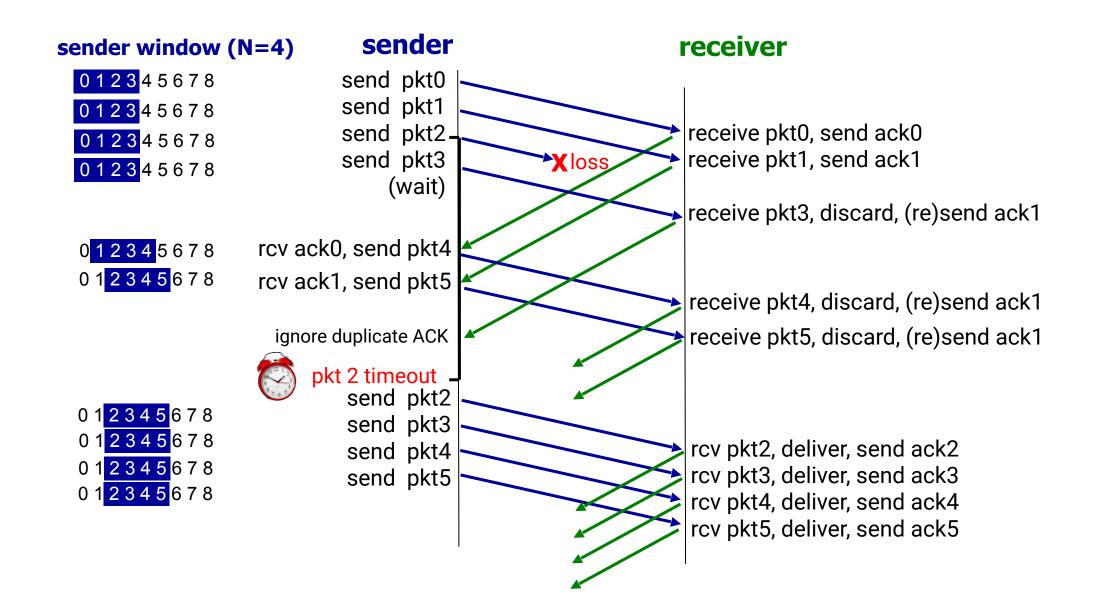
Out-of-order: received but not ACKed

Not received

#### Receiver

- Out-of-order packets: are typically discarded
- Discarding results in lower channel utilization, whereas buffering introduces extra state management
- Acknowledging: always send ACK for any correctlyreceived packet; however, the ACK will always carry the seq# of the highest in-order packet
- This ACK scheme may generate duplicate ACKs

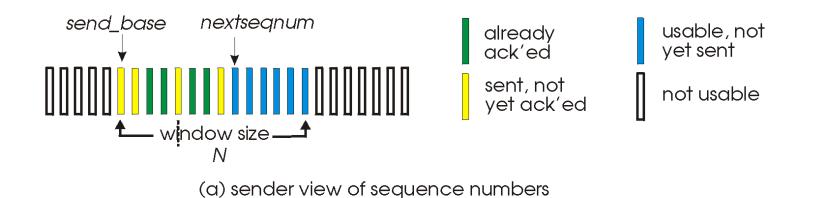
#### **Go-Back-N** in action

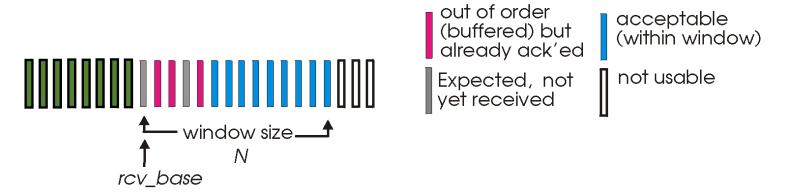


## **Selective Repeat (SR) Protocol**

	Go-Back-N	Selective Repeat
ACKs	<b>Cumulative</b> i.e., ACK(k) will ACK all packets up to and including #k	Individual i.e., ACK(k) just ACKs packet #k
Out of order packets	Receiver discards all out of order packets	Buffers out-of-order packet for later delivery
Buffer size	Sender buffer = N; receiver buffer = 1	Sender buffer = N; Receiver buffer = N
Sender timer	Set for only the oldest unacknowledged packet	Set for every transmitted packet

## Selective Repeat: sender and receiver windows





(b) receiver view of sequence numbers

## **Selective Repeat: sender and receiver**

#### sender

#### data from above:

if next available seq # in window, then send packet and start timer

#### timeout(k):

resend packet k, restart timer

#### ACK(k) in [sendbase, sendbase+N]:

- mark packet k as received
- remove the timer for packet k
- if k smallest unACKed packet, advance window base to next unACKed seq #

#### receiver

#### packet k in [rcvbase, rcvbase+N]

- send ACK(k)
- out-of-order: buffer
- in-order: deliver (also deliver buffered, inorder packets), advance window to next not-yet-received packet

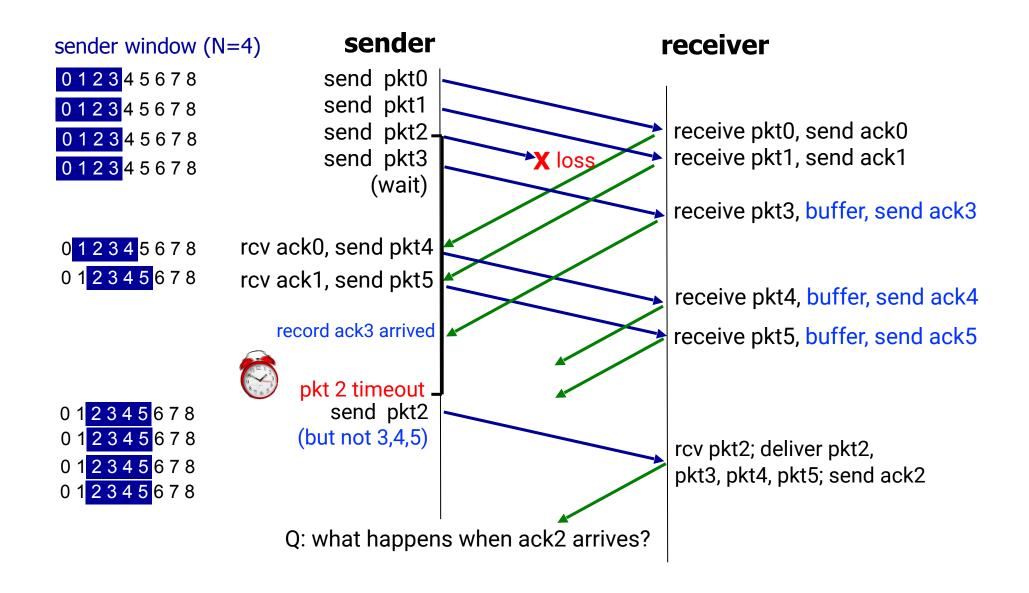
#### packet k in [rcvbase-N, rcvbase-1]

ACK(k)

#### otherwise:

ignore

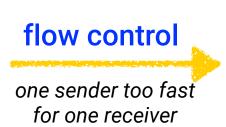
## **Selective Repeat in action**



## **Congestion Control**

## Distinguishing Flow Control from Congestion Control







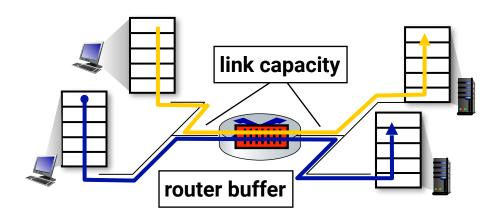


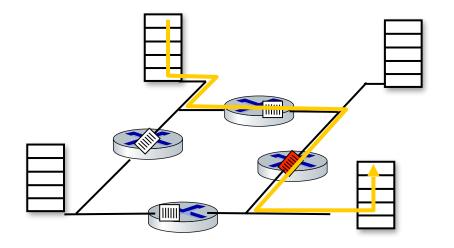


many senders, too fast for the network to handle



## **Causes and Implications of Congestion**





#### 1. When the router's link capacity gets saturated

- packets experience queuing delays
- senders and receivers experience lower throughput

#### 2. When the router's buffer overflows

- some packets get dropped, causing sender to retransmit
- some packets get delayed, causing sender to retransmit duplicate packets
- both cases lead to lower throughput and higher latency

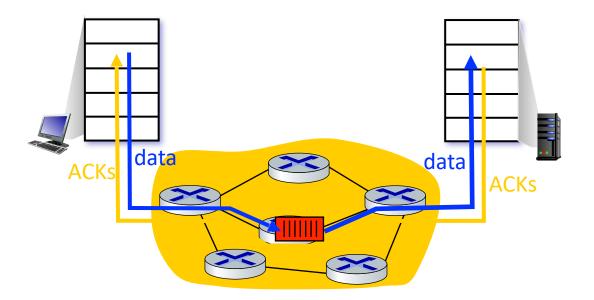
#### 3. When an upstream router drops a packet

- transmission capacity used up by the packet so far is wasted
- this reduces the utilization of the overall network

## Approaches towards congestion control

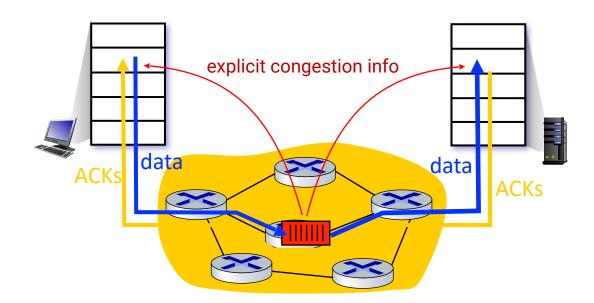
#### **End-to-end** (reactive)

- no explicit feedback from network layer
- congestion is *inferred* from observed loss and delay (via timeouts, duplicate ACKs, or RTT measurements)
- this is the approach taken by the original TCP



#### **Network-assisted** (proactive)

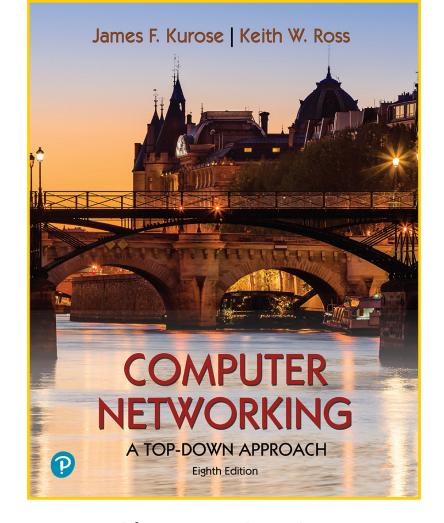
- a congested router provides direct feedback to all hosts with flows passing through it
- router acts proactively to indicate its congestion level or to explicitly set sending rate
- RFC3168 (in 2001) added explicit congestion notification (ECN) to TCP and IP



## **Next lecture**

from principles to practice: design and operation of TCP

- Protocol structure
- Connection management
- Reliable data transfer
- Flow and congestion control



Chapters 3.5, 3.7



# **Spot Quiz (ICON)**