# **Transport Protocols & TCP**

CSE 3213 Fall 2007

13 November 2007

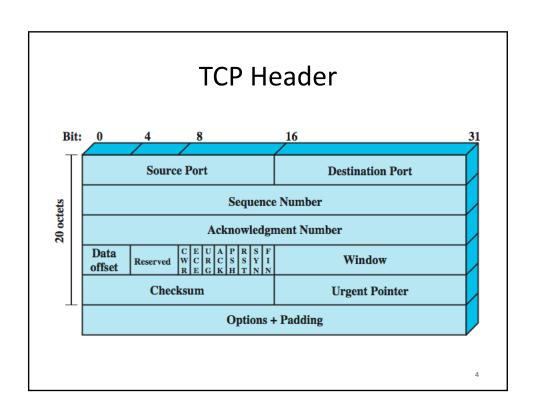
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# TCP

- Services
- Flow control
- Connection establishment and termination
- Congestion control

#### **TCP Services**

- Transmission Control Protocol (RFC 793)
- connection oriented, reliable communication
- over reliable and unreliable (inter)networks
- two ways of labeling data:
- data stream push
  - user requires transmission of all data up to push flag
  - receiver will deliver in same manner
  - avoids waiting for full buffers
- urgent data signal
  - indicates urgent data is upcoming in stream
  - user decides how to handle it



#### Issues

- ordered delivery,
- retransmission strategy,
- · duplication detection,
- flow control,
- · connection establishment & termination,
- crash recovery
- Note: since the underlying network is unreliable,
  - segments may get lost
  - segments may arrive out of order

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# **Ordered Delivery**

- segments may arrive out of order
- hence number segments sequentially
- TCP numbers each octet sequentially
- and segments are numbered by the first octet number in the segment

#### **TCP Flow Control**

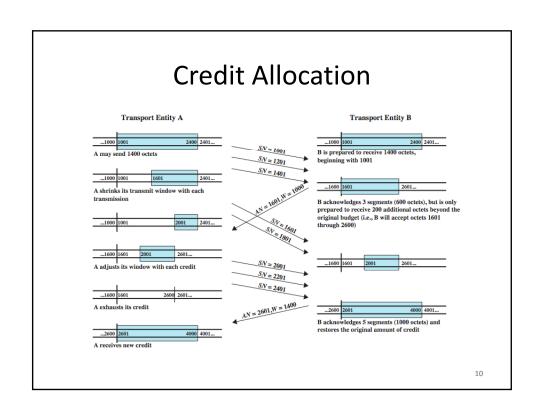
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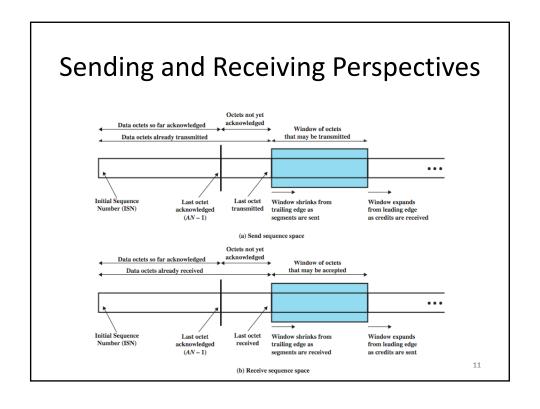
#### Flow Control

- Fixed sliding window approach
  - works well on reliable networks
  - does not work well on unreliable networks such as IP internet
- Credit scheme
  - more flexible
  - works for IP
  - used in TCP

#### **Credit Scheme**

- decouples flow control from ACK
- each octet has sequence number
- each transport segment has seq number (SN), ack number (AN) and window size (W) in header
- sends seq number of first octet in segment
- ACK includes (AN=i, W=j) which means
  - all octets through SN=i-1 acknowledged, want i next
  - permission to send additional window of W=j octets





# **Retransmission Strategy**

- retransmission of segment needed because
  - segment damaged in transit
  - segment fails to arrive
- transmitter does not know of failure
- receiver must acknowledge successful receipt
  - can use cumulative acknowledgement for efficiency
- sender times out waiting for ACK triggers re-transmission

#### **Retransmit Policy**

- TCP has a queue of segments transmitted but not acknowledged
- will retransmit if not ACKed in given time
  - first only single timer, send the front segment when timer expires, efficient, considerable delays
  - batch single timer, send all segments when timer expires, has unnecessary retransmissions
  - individual timer for each segment, lower delay, more efficient, but complex
- effectiveness depends in part on receiver's accept policy

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## **Accept Policy**

- segments may arrive out of order
- in order
  - only accept segments in order
  - discard out of order segments
  - simple implementation, but burdens network
- in windows
  - accept all segments within receive window
  - reduce transmissions
  - more complex implementation with buffering

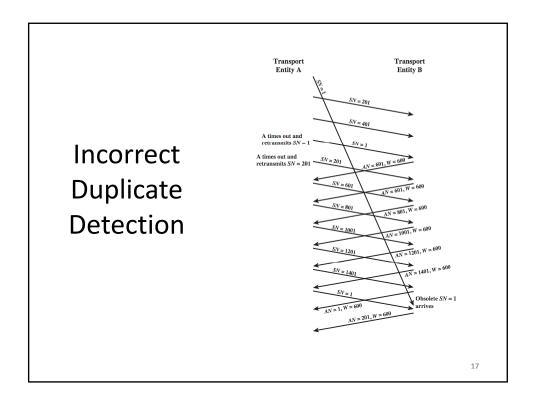
### **Acknowledgement Policy**

- immediate
  - send empty ACK for each accepted segment
  - simple at cost of extra transmissions
- cumulative
  - piggyback ACK on suitable outbound data segments unless persist timer expires
  - when send empty ACK
  - more complex but efficient

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## **Duplication Detection**

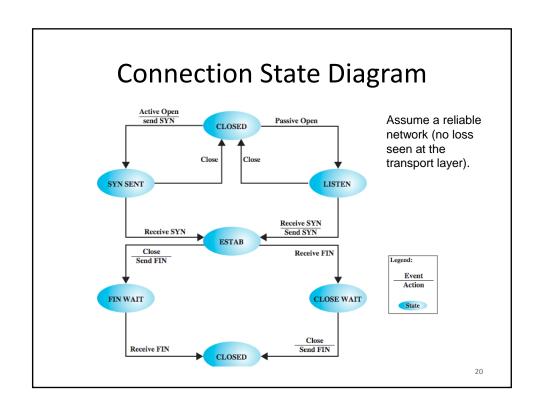
- if ACK lost, segment duplicated & re-transmitted
- receiver must recognize duplicates
- if duplicate received prior to closing connection
  - receiver assumes ACK lost and ACKs duplicate
  - sender must not get confused with multiple ACKs
  - need a sequence number space large enough to not cycle within maximum life of segment



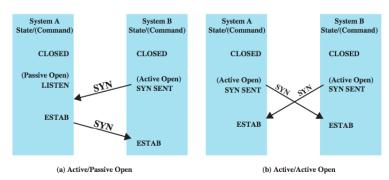
# **Connection Establishment and Termination**

# Connection Establishment and Termination

- required by connection-oriented transport protocols like TCP
- need connection establishment and termination procedures to allow:
  - each end to know the other exists
  - negotiation of optional parameters
  - triggers allocation of transport entity resources



# **Connection Establishment Diagram**



Assume a reliable network (no

loss seen at the transport layer).

What if either SYN is lost? (discussed later)

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#### **Connection Termination**

- either or both sides by mutual agreement
- graceful or abrupt termination
- if graceful, initiator must:
  - send FIN to other end, requesting termination
  - place connection in FIN WAIT state
  - when FIN received, inform user and close connection
- other end must:
  - when receives FIN must inform TS user and place connection in CLOSE WAIT state
  - when TS user issues CLOSE primitive, send FIN & close connection

#### **Connection Establishment**

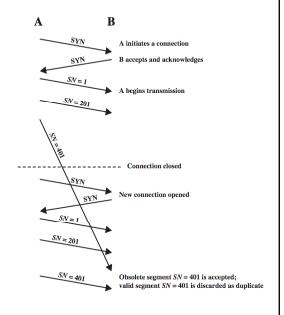
- two way handshake
  - A send SYN, B replies with SYN
  - lost SYN handled by re-transmission
  - ignore duplicate SYNs once connected
- lost or delayed data segments can cause connection problems
  - eg. segment from old connection

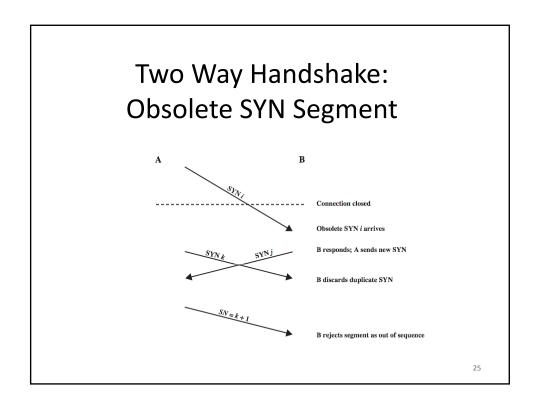
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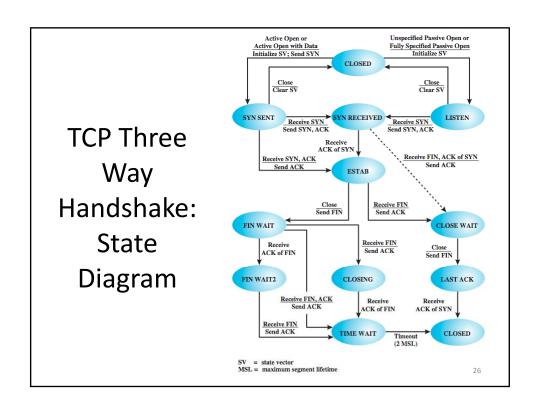
# Two Way Handshake: Obsolete Data Segment

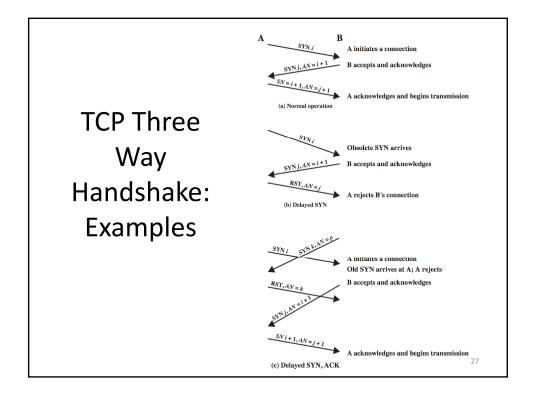
Solution: starting SN is far away from the last SN of the previous connection.

Use request of the form SYNi where i+1 is the SN of the first data segment to be sent.









# TCP Connection Establishment: Summary

- three way handshake
  - SYN, SYN-ACK, ACK
- connection determined by source and destination sockets (host, port)
- can only have a single connection between any unique pairs of ports
- but one port can connect to multiple ports

## Connection Termination (2)

- also need 3-way handshake
- misordered segments could cause:
  - entity in CLOSE WAIT state sends last data segment, followed by FIN
  - FIN arrives before last data segment
  - receiver accepts FIN, closes connection, loses data
- need to associate sequence number with FIN
- receiver waits for all segments before FIN sequence number

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# Connection Termination: Graceful Close

- also have problems with loss of segments and obsolete segments
- need graceful close which will:
- send FIN i and receive AN i+1
- receive FIN j and send AN j+1
- wait twice maximum expected segment lifetime

# **TCP Congestion Control**

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# **TCP Congestion Control**

- flow control also used for congestion control
  - recognize increased transit times & dropped packets
  - $\boldsymbol{-}$  react by reducing flow of data
- RFC's 1122 and 2581 detail extensions
  - Tahoe, Reno and New Reno implementations
- two categories of extensions:
  - retransmission timer management
  - window management

#### **Retransmission Timer Management**

- static timer likely too long or too short
- estimate round trip delay by observing pattern of delay for recent segments
- set time to value a bit greater than estimated RTT
- simple average over a number of segments
- exponential average using time series (RFC793)

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# **Computing RTT**

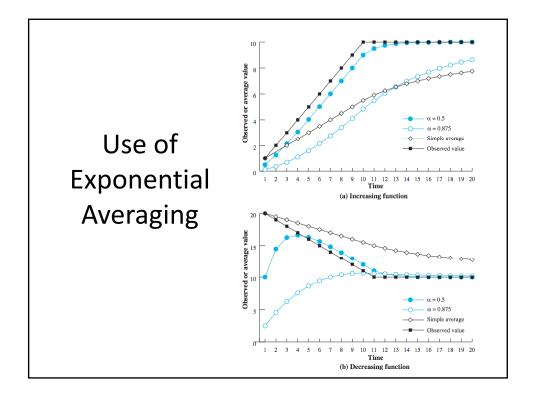
• Simple average

$$r(K+1) = \frac{1}{K+1} \sum_{i=1}^{K+1} RTT(i)$$

$$r(K+1) = \frac{K}{K+1}r(K) + \frac{1}{K+1}RTT(K+1)$$

Exponential average

$$r(K+1) = a \times r(K) + (1-a) \times RTT(K+1)$$
$$0 < a < 1$$



# **Exponential RTO Backoff**

- timeout probably due to congestion
  - dropped packet or long round trip time
- hence maintaining same RTO is not good idea
- better to increase RTO each time a segment is re-transmitted
  - -RTO = q\*RTO
  - commonly q = 2 (binary exponential backoff)
  - as in Ethernet CSMA/CD

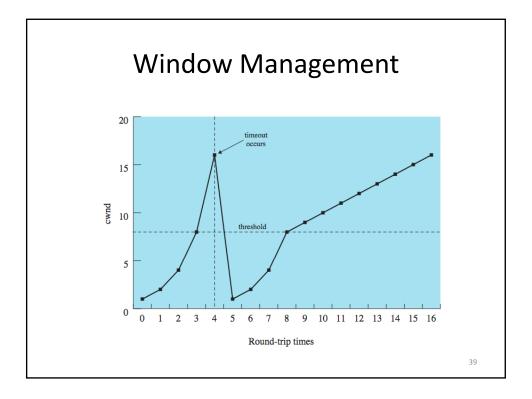
## Window Management

- slow start
  - larger windows cause problem on connection created
  - at start limit TCP to 1 segment
  - increase when data ACK, exponential growth
- dynamic windows sizing on congestion
  - when a timeout occurs perhaps due to congestion
  - set slow start threshold to half current congestion window
  - set window to 1 and slow start until threshold
  - beyond threshold, increase window by 1 for each RTT

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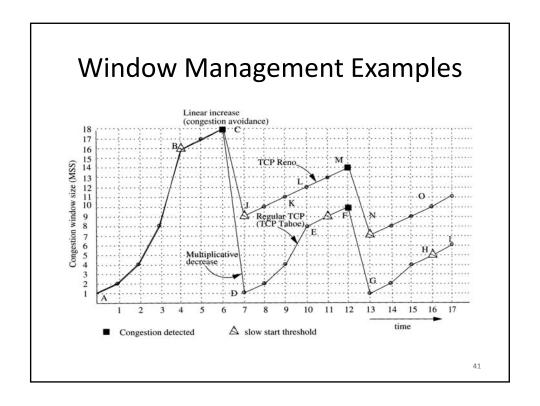
#### **Summary**

- Assigns a congestion window C<sub>w</sub>:
  - Initial value of C<sub>w</sub> = 1 (packet)
  - If trx successful, congestion window doubled. Continues until  $C_{\text{max}}$  is reached
  - After  $C_w \ge C_{max}$ ,  $C_w = C_w + 1$
  - If timeout before ACK, TCP assumes congestion
- TCP response to congestion is drastic:
  - A random backoff timer disables all transmissions for duration of timer
  - C<sub>w</sub> is set to 1
  - C<sub>max</sub> is set to C<sub>max</sub> / 2
- Congestion window can become quite small for successive packet losses.



# Fast Retransmit, Fast Recovery

- retransmit timer rather longer than RTT
- if segment lost TCP slow to retransmit
- fast retransmit
  - if receive a segment out of order, issue an ACK for the last in-order segment correctly received. Repeat this until the missing segment arrives.
  - if receive 4 ACKs for same segment then immediately retransmit (without waiting for timer) since it is likely to have been lost
- fast recovery
  - lost segment means some congestion
  - halve window then increase linearly
  - avoids slow-start



# Reading

• Chapter 20, Stallings' book