

Department of Electronics and Communication Engineering

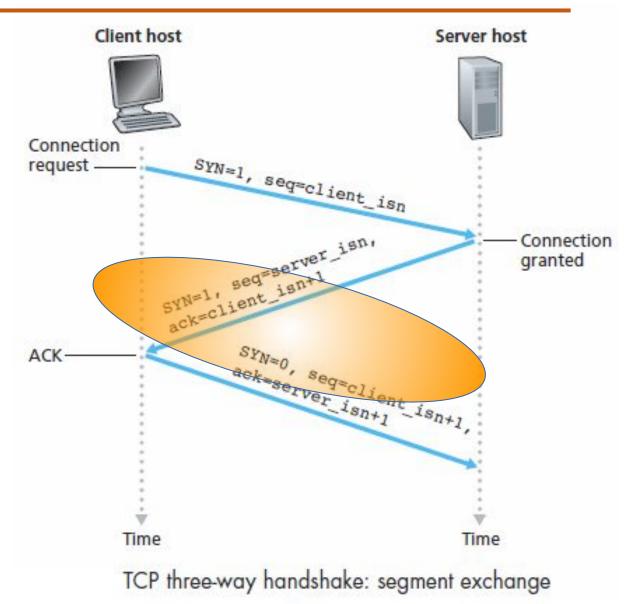


# TCP Connection Opening-Closing, Timeout

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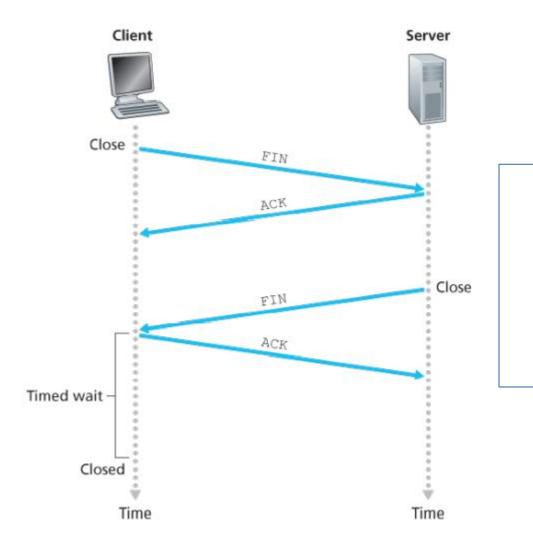
# TCP three-way handshake



# **TCP Connection Closing/Termination**

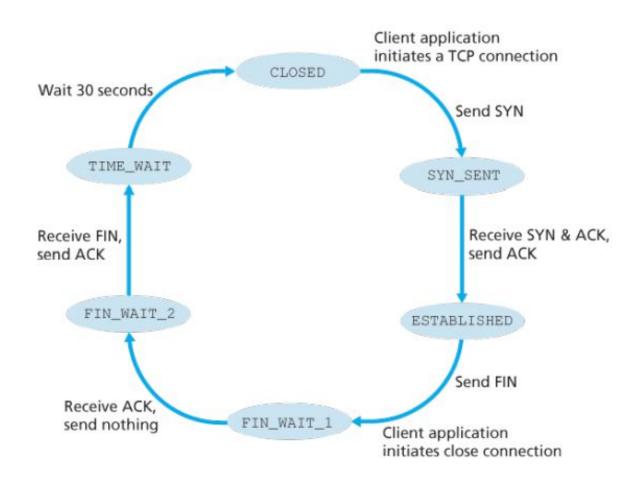
- client, server each close their side of connection
  - send TCP segment with FIN bit = 1
- respond to received FIN with ACK
  - on receiving FIN, ACK can be combined with own FIN
- simultaneous FIN exchanges can be handled

# **TCP Connection Closing**

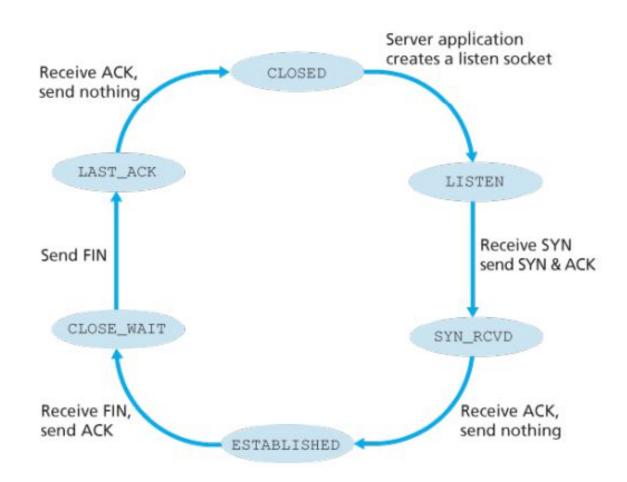


At this point, all the resources in the two hosts are now deallocated.

## A typical sequence of TCP states visited by a client TCP



# A typical sequence of TCP states visited by a server-side TCP



#### **TCP: RTT & Timeout**

- Q: how to set TCP timeout value?
- longer than RTT, but RTT varies!
- •too short: premature timeout, unnecessary retransmissions
- too long: slow reaction to segment loss

- Q: how to estimate RTT?
- SampleRTT:measured time from segment transmission until ACK receipt
  - ignore retransmissions
- SampleRTT will vary, want estimated RTT "smoother"
  - average several recent measurements, not just current SampleRTT

### **TCP: Sender Events**

# event: data received from application

- create segment with seq #
- seq # is byte-stream number of first data byte in segment
- start timer if not already running
  - think of timer as for oldest unACKed segment
  - expiration interval:TimeOutInterval

#### event: timeout

- retransmit segment that caused timeout
- restart timer

#### event: ACK received

- if ACK acknowledges previously unACKed segments
  - update what is known to be ACKed
  - start timer if there are still unACKed segments

#### **TCP Sender Events**

#### **Timeout interval estimation:**

- Based on round trip time (RTT)
  - Time taken to get acknowledgement for a TCP segment
- RTT is measured for one of the TCP segments at a time
  - Instantaneous value of RTT is referred to as SampleRTT
- RTT is measured only for freshly transmitted segments
  - Segments are chosen randomly
- Time averaged statistics are generated for the RTTs
  - Mean value is denoted as *EstimatedRTT* and standard deviation is denoted as DevRTT
- Timeout interval is chosen as a random number based on the time averaged mean and standard deviation of the RTTs

#### **TCP Sender Events**

## Timeout interval estimation(condt.)

 $\mu_n$ : Estimated RTT in round n

 $\sigma_n$ : DevRTT in round n

 $r_n$ : RTT measured in round n

#### $T_n$ : Timeout interval in round n

Update for  $\mu_n$  and  $\sigma_n$  are given by

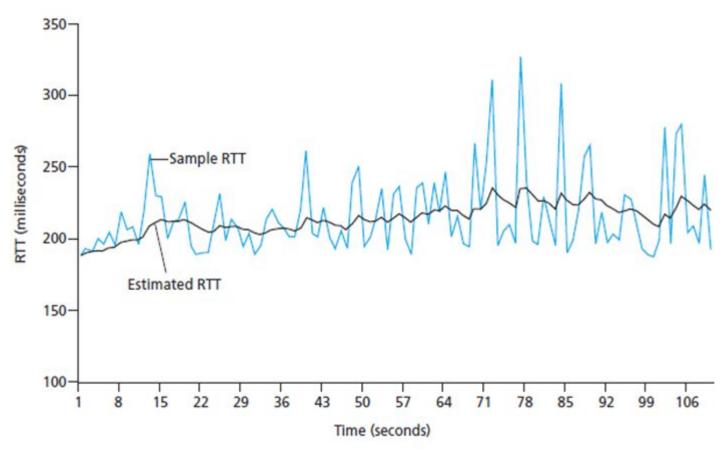
$$\mu_n = (1 - \alpha)\mu_{n-1} + \alpha r_n$$

$$\sigma_n = (1 - \beta)\sigma_{n-1} + \beta|r_n - \mu_n|$$

Update of timeout  $T_n$  is given by  $T_n = \mu_n + 4\sigma_n$ 

## **TCP RTT & Timeout**

# Timeout interval estimation(condt.)



# **COMPUTER COMMUNICATION NETWORKS TCP RTT & Timeout**

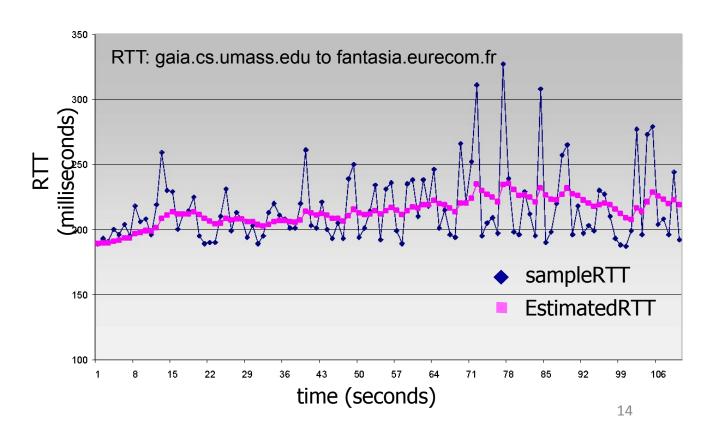
## Timeout interval estimation(condt.)

- Recommended value of  $\alpha$  is 0.125
- Recommended value of  $\beta$  is 0.25
- Initial value of Timeout interval is 1 sec
- RTT is measured and stored as SampleRTT (x<sub>n</sub>) in each round
- After each success (i.e., acknowledgement is received), the Timeout interval for round n is updated using the equations in previous slide
- In case a timeout occurs (i.e., acknowledgements missed), Timeout interval is doubled
- Packets whose acknowledgements were missed in previous round are retransmitted and timer is set to the Timeout interval
- The above steps repeats while data carrying TCP segments are sent

**TCP: RTT & Timeout** 

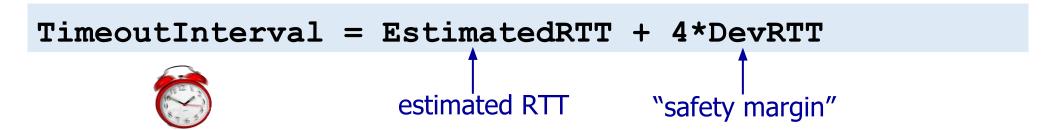
# EstimatedRTT = $(1-\alpha)$ \*EstimatedRTT + $\alpha$ \*SampleRTT

- <u>exponential weighted moving average (EWMA)</u>
- influence of past sample decreases exponentially fast
- typical value:  $\alpha$  = 0.125



**TCP: RTT & Timeout** 

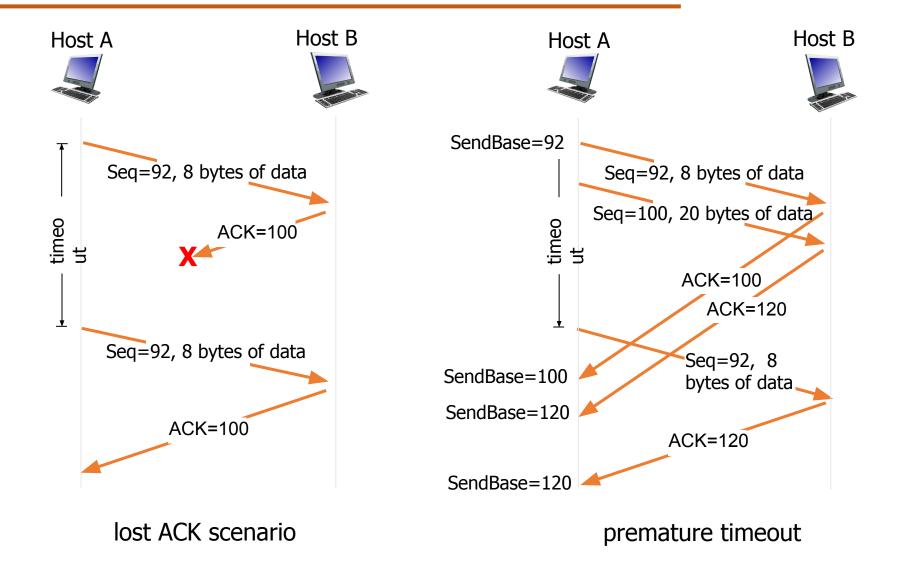
- timeout interval: EstimatedRTT plus "safety margin"
  - large variation in **EstimatedRTT**: want a larger safety margin



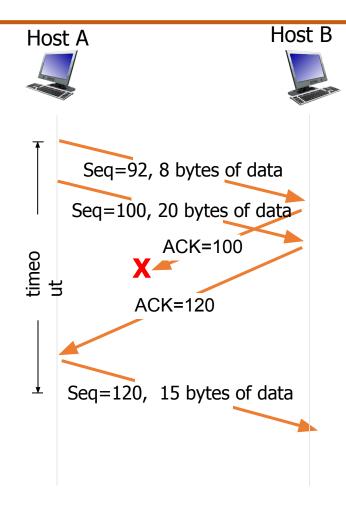
•DevRTT: EWMA of SampleRTT deviation from EstimatedRTT:

DevRTT = 
$$(1-\beta)*DevRTT + \beta*|SampleRTT-EstimatedRTT|$$
 (typically,  $\beta = 0.25$ )

#### **TCP Retransmission Scenarios**



## **TCP Retransmission Scenarios**



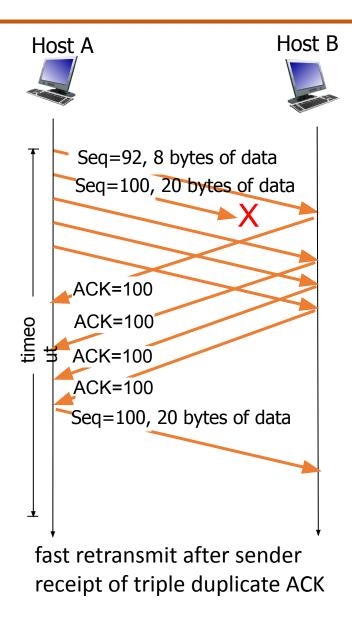
cumulative ACK

## **TCP ACK Generation**

event at receiver	TCP receiver action
arrival of in-order segment with expected seq #. All data up to expected seq # already ACKed	delayed ACK. Wait up to 500ms for next segment, send ACK
arrival of in-order segment with expected seq #. One other segment has ACK pending	immediately send single cumulative ACK, ACKing both in-order segments
arrival of out-of-order segment higher-than-expect seq. # . Gap detected	immediately send duplicate ACK, indicating seq. # of next expected byte
arrival of segment that partially or completely fills gap	immediate send ACK, provided that segment starts at lower end of gap

#### **TCP Fast Retransmit**

# TCP fast retransmit





# **THANK YOU**

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