

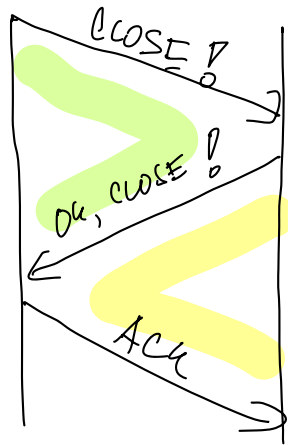
CS 725/825 & IT 725

Lecture 18

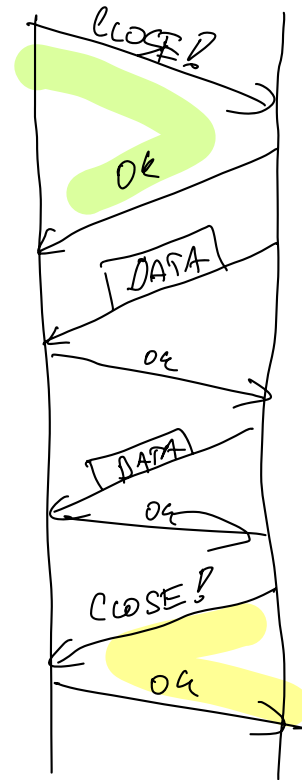
Transport Layer

November 6, 2023

CLOSING CONNECTION

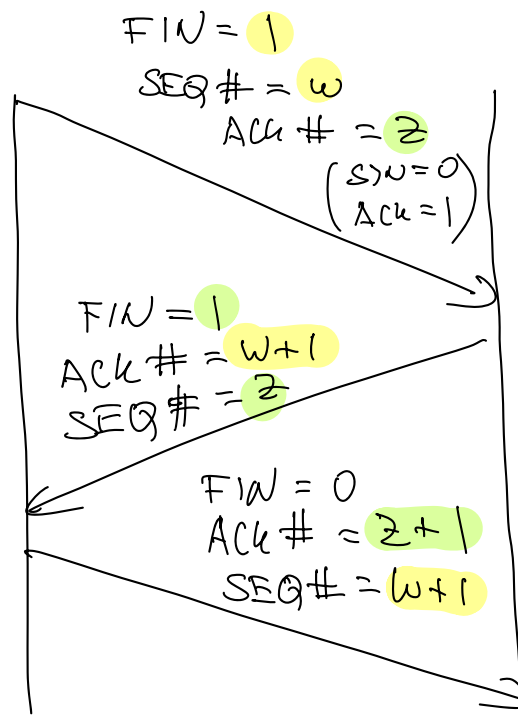


SYM. RELEASE



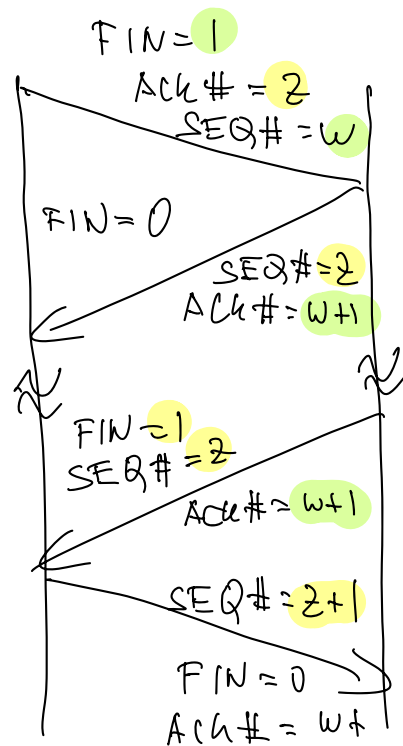
ASYM. RELEASE

TCP - SYM. RELEASE

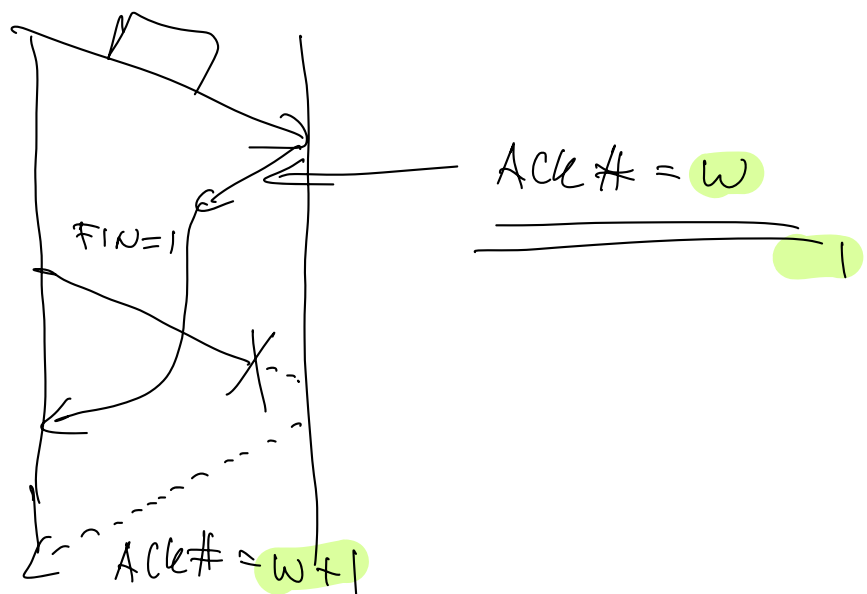


FIN - finish

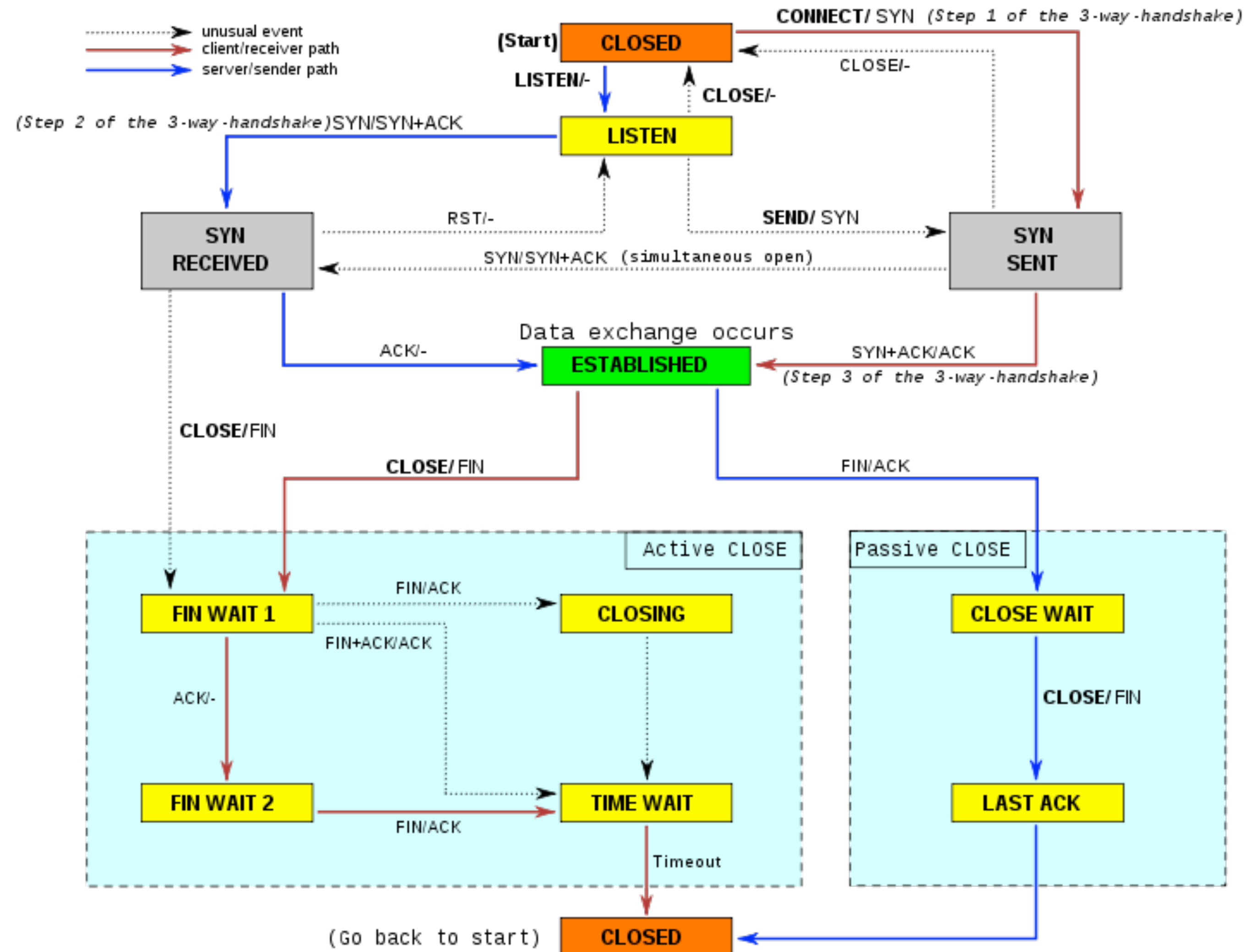
TCP - ASYM. RELEASE



MOTIVATION FOR "COUNTING" $FIN=1$
AS ONE BYTE in $SEQ/ACK\#$ CALCULATION



TCP State Diagram



TCP Congestion Control

- ▶ Flow (receiver congestion) control
 - Window Size field - explicitly reported by the receiver
 - TCP Window Scale Option
- ▶ Network congestion control
 - Retransmission timeout - based on observed RTT
 - Transmission window - based on detected packet loss

Implicit Congestion Notification

► Round Trip Time (RTT)

- time between data packet transmission and reception of it acknowledgement
- increase in RTT could be interpreted as due to an increase in queue lengths in nodes (congestion)
- ... or it could be due to a route change...

► Packet Loss

- packet loss due to queue overflow (congestion)
- ... or it could be packet loss due to random packet errors

Retransmission Timeout

Initialization:

RFC 6298

$$\text{RTO} \leftarrow 1 \text{ sec}$$

After the first measurement:

$$\text{SRTT} \leftarrow R$$

$$\text{RTTVAR} \leftarrow R/2$$

$$\text{RTO} \leftarrow \text{SRTT} + \max(G, K * \text{RTTVAR})$$

After subsequent measurements:

$$\text{RTTVAR} \leftarrow (1 - \text{beta}) * \text{RTTVAR} + \text{beta} * |\text{SRTT} - R'|$$

$$\text{SRTT} \leftarrow (1 - \text{alpha}) * \text{SRTT} + \text{alpha} * R'$$

$$\text{RTO} \leftarrow \text{SRTT} + \max(G, K * \text{RTTVAR})$$

Where:

R - first RTT measurement

R' - subsequent RTT measurement

RTTVAR - RTT variance

SRTT - smoothed RTT estimate

RTO - retransmission timeout

G - clock granularity

Recommended values:

alpha=1/8, beta=1/4, K=4

Exponential Back-off

RTO after a timeout:

$$\text{RTO} \leftarrow q * \text{RTO}$$

Recommended value: $q = 2$

This is a **congestion control mechanism** since retransmissions are delayed after packet loss detected. The delay is increasing **exponentially** with more packet losses.

Transmission Window

- ▶ Network provides no explicit indication of congestion
- ▶ Source observes RTT and packet loss and adjusts transmission rate according to its estimate of the congestion state of the network
- ▶ Transmission window size is proportional to the maximum transmission rate
- ▶ Additive Increase Multiplicative Decrease (AIMD)
 - better safe than sorry

Network Congestion Control

► Method:

$$\text{TransWind} = \min(\text{RecvWind}, \text{CongWind})$$

- **TransWind** - used in transmission
- **RecvWind** - from Window Size field
- **CongWind** - transmitter's estimate of how many unacknowledged packets can be pushed onto the network without causing congestion

TCP CONGESTION WINDOW

