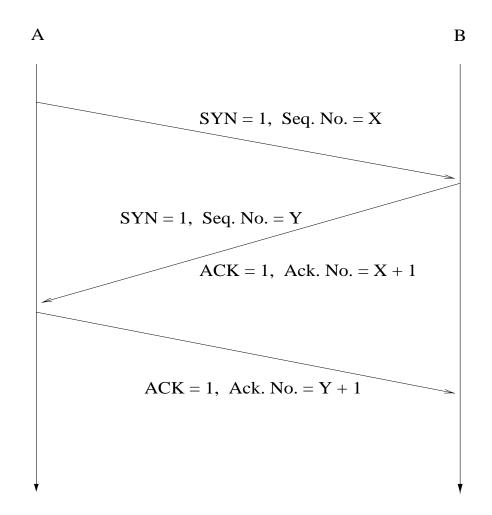
#### TCP connection establishment (3-way handshake):

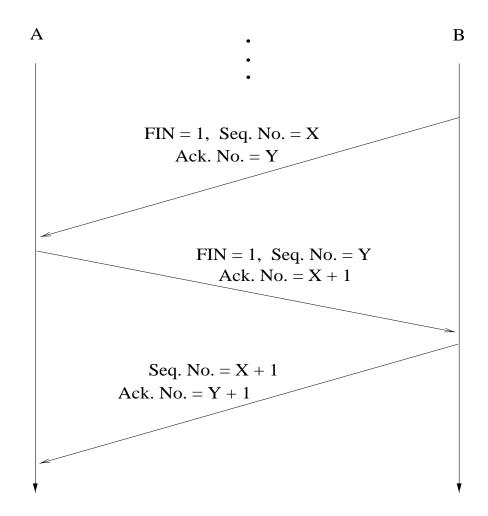


- X, Y are chosen randomly
  - $\rightarrow$  sequence number prediction
- piggybacking

2-person consensus problem: are A and B in agreement about the state of affairs after 3-way handshake?

- $\longrightarrow$  in general: impossible
- $\longrightarrow$  can be proven
- $\longrightarrow$  "acknowledging the ACK problem"
- $\longrightarrow$  also TCP session ending
- $\longrightarrow$  lunch date problem

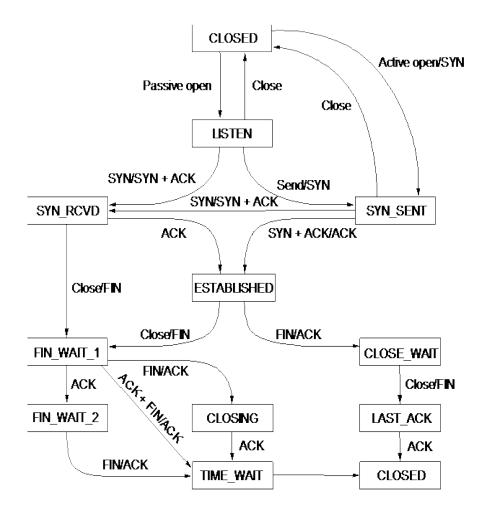
# TCP connection termination:



- full duplex
- half duplex

More generally, finite state machine representation of TCP's control mechanism:

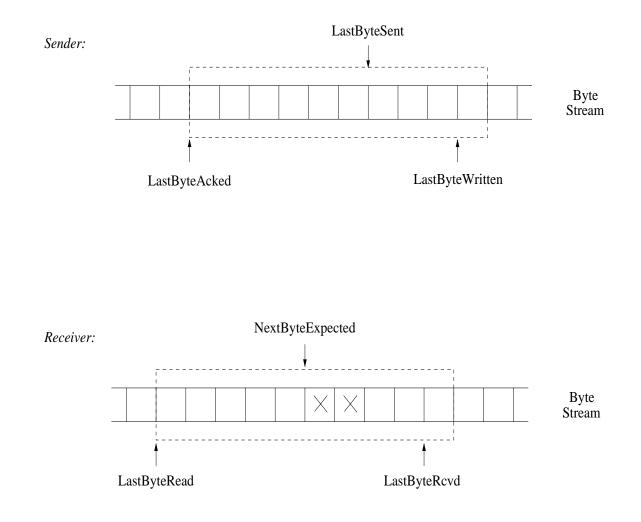
 $\longrightarrow$  state transition diagram



Features to notice:

- Connection set-up:
  - client's transition to ESTABLISHED state without ACK
  - how is server to reach ESTABLISHED if client ACK is lost?
  - ESTABLISHED is macrostate (partial diagram)
- Connection tear-down:
  - three normal cases
  - $-\operatorname{special}$  issue with <code>TIME WAIT</code> state
  - $-\operatorname{employs}$  hack

# TCP's sliding window protocol



• sender, receiver maintain buffers MaxSendBuffer, MaxRcvBuffer Same as generic sliding window

 $\longrightarrow$  data unit: byte, not packet

Sender side: maintain invariants

- LastByteAcked  $\leq$  LastByteSent  $\leq$  LastByteWritten
- $\bullet \texttt{LastByteWritten-LastByteAcked} < \texttt{MaxSendBuffer}$

 $\longrightarrow$  buffer flushing (advance window)

- $\longrightarrow$  application blocking
- $\bullet \texttt{LastByteSent-LastByteAcked} \leq \texttt{AdvertisedWindow}$ 
  - $\longrightarrow$  AdvertisedWindow: receiver side free space
  - $\longrightarrow$  throttling effect

How much sender can still send:

```
\label{eq:linear} \begin{split} \texttt{EffectiveWindow} &= \texttt{AdvertisedWindow} - \\ & (\texttt{LastByteSent} - \texttt{LastByteAcked}) \end{split}
```

- $\longrightarrow$  upper bound
- $\longrightarrow$  sender may choose to send less
- $\longrightarrow$  self-throttling

Affected through sender side variable

 $\longrightarrow$  CongestionWindow

```
\label{eq:linear} \begin{split} \texttt{EffectiveWindow} &= \texttt{MaxWindow} - \\ & (\texttt{LastByteSent} - \texttt{LastByteAcked}) \end{split}
```

where

```
MaxWindow =
```

 $\min\{\texttt{AdvertisedWindow}, \texttt{CongestionWindow}\}$ 

How to set CongestionWindow.

 $\longrightarrow$  TCP congestion control

Receiver side: maintain invariants

- LastByteRead < NextByteExpected  $\leq$  LastByteRcvd + 1
- $\bullet \texttt{LastByteRcvd} \texttt{NextByteRead} < \texttt{MaxRcvBuffer}$

 $\longrightarrow$  buffer flushing (advance window)

 $\longrightarrow$  application blocking

Thus,

```
\label{eq:advertisedWindow} \begin{split} \texttt{AdvertisedWindow} &= \texttt{MaxRcvBuffer} - \\ & (\texttt{LastByteRcvd} - \texttt{LastByteRead}) \end{split}
```

How to let sender know of change in receiver window size after AdvertisedWindow becomes 0?

- trigger ACK event on receiver side when
  AdvertisedWindow becomes positive
- $\bullet$  sender periodically sends 1-byte probing packet
  - $\longrightarrow$  design choice: smart sender/dumb receiver
  - $\longrightarrow$  same situation for congestion control

Silly window syndrome: Assuming receiver buffer is full,

what if application reads one byte at a time with long pauses?

- can cause excessive 1-byte traffic
- if AdvertisedWindow < MSS then set

```
\texttt{AdvertisedWindow} \gets 0
```

Do not want to send too many 1 B payload packets.

Nagle's method:

- rule: connection can have only one such unacknowledged packet outstanding
- while waiting for ACK, incoming bytes are accumulated (i.e., buffered)

 $\ldots$  compromise between real-time constraints and efficiency.

 $\rightarrow$  useful for telnet/ssh-type interactive applications

#### RTT estimation

... important to not underestimate nor overestimate.

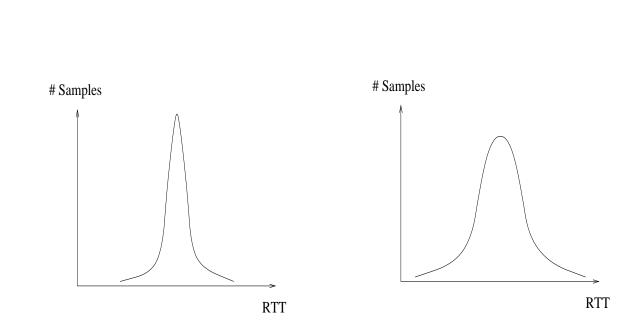
Karn/Partridge: Maintain running average with precautions

 $\texttt{EstimateRTT} \leftarrow \alpha \cdot \texttt{EstimateRTT} + \beta \cdot \texttt{SampleRTT}$ 

• SampleRTT computed by sender using timer

• 
$$\alpha + \beta = 1; \ 0.8 \le \alpha \le 0.9, \ 0.1 \le \beta \le 0.2$$

- TimeOut  $\leftarrow 2 \cdot \texttt{EstimateRTT}$  or TimeOut  $\leftarrow 2 \cdot \texttt{TimeOut}$  (if retransmit)
  - $\longrightarrow$  need to be careful when taking **SampleRTT**
  - $\longrightarrow$  infusion of complexity
  - $\longrightarrow$  still remaining problems



- $\longrightarrow$  need to account for variance
- $\longrightarrow$  not nearly as nice

Hypothetical RTT distribution:

### Jacobson/Karels:

- Difference = SampleRTT EstimatedRTT
- EstimatedRTT = EstimatedRTT +  $\delta \cdot \text{Difference}$
- Deviation = Deviation +  $\delta(|\text{Difference}| \text{Deviation})$

Here  $0 < \delta < 1$ .

Finally,

• TimeOut =  $\mu \cdot \texttt{EstimatedRTT} + \phi \cdot \texttt{Deviation}$ 

where  $\mu = 1, \phi = 4$ .

- $\longrightarrow$  persistence timer
- $\longrightarrow$  how to keep multiple timers in UNIX