## Chapter 3 outline

- 3.1 transport-layer services
- 3.2 multiplexing and demultiplexing
- 3.3 connectionless transport: UDP
- 3.4 principles of reliable data transfer

- 3.5 connection-oriented transport: TCP
  - segment structure
  - reliable data transfer
  - flow control
  - connection management
- 3.6 principles of congestion control
- 3.7 TCP congestion control

### TCP flow control

application may remove data from TCP socket buffers ....

... slower than TCP receiver is delivering (sender is sending)

### application process application OS TCP socket receiver buffers **TCP** code ΙP code from sender

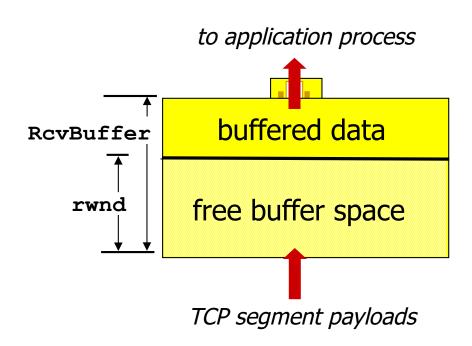
receiver protocol stack

#### flow control

receiver controls sender, so that sender won't overflow receiver's buffer by transmitting too much, too fast

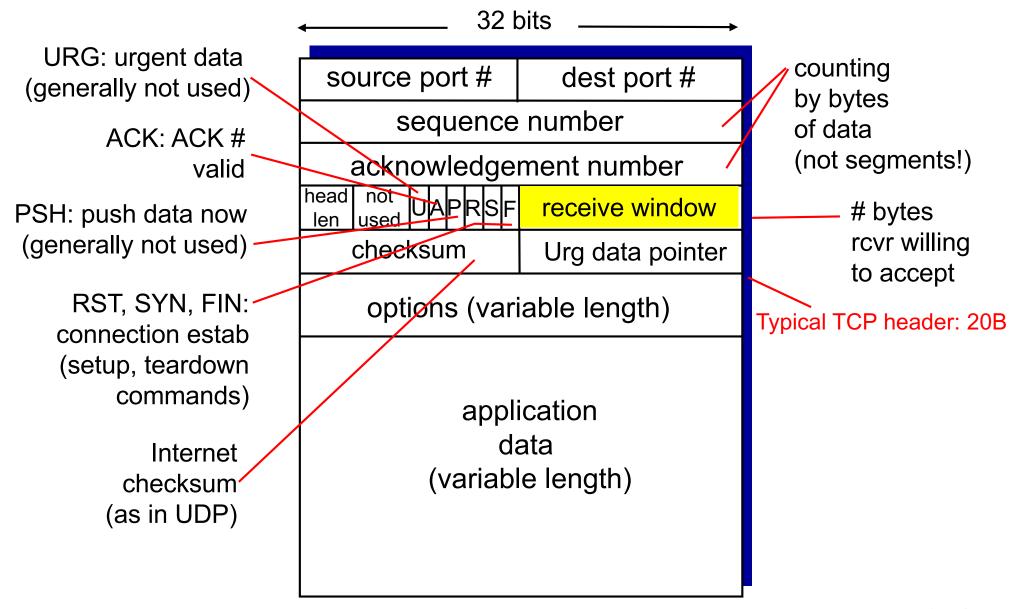
### TCP flow control

- receiver "advertises" free buffer space by including rwnd value in TCP header of receiver-to-sender segments
  - RcvBuffer size set via socket options (typical default is 4096 bytes)
  - many operating systems autoadjust RcvBuffer
- sender limits amount of unacked ("in-flight") data to receiver's rwnd value
- guarantees receive buffer will not overflow



receiver-side buffering

## TCP segment structure



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## TCP segment structure

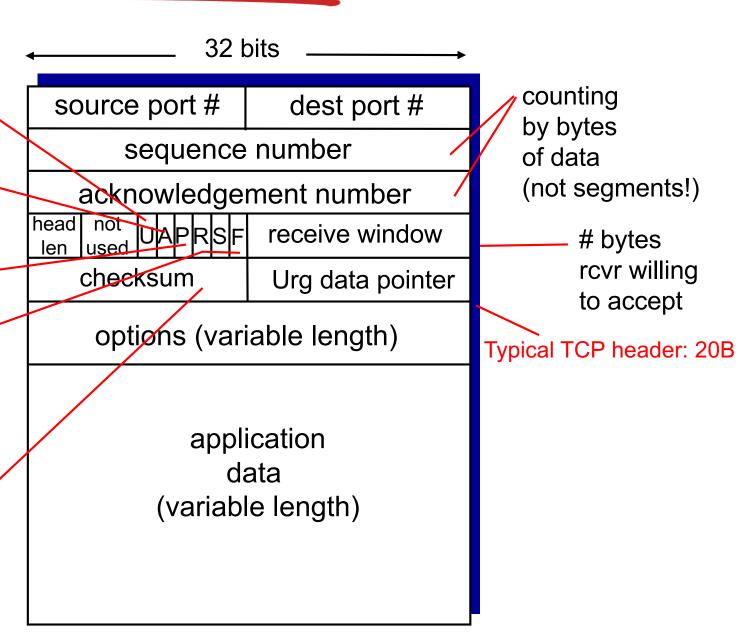
URG: urgent data (generally not used)

ACK: ACK # valid

PSH: push data now (generally not used)

RST, SYN, FIN: connection estab (setup, teardown commands)

> Internet checksum<sup>\*</sup> (as in UDP)



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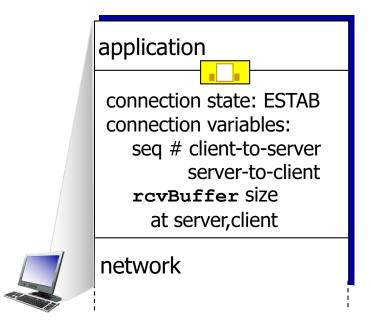
## TCP segment structure

32 bits **URG**: urgent data counting source port # dest port # (generally not used) by bytes sequence number of data ACK: ACK # (not segments!) acknowledgement number valid head not PRSFI receive window # bytes PSH: push data now used len rcvr willing (generally not used) cheeksum Urg data pointer to accept RST, SYN, FIN: options (variable length) Typical TCP header: 20B connection estab (setup, teardown commands) application data Internet (variable length) checksum (as in UDP)

### Connection Management

before exchanging data, sender/receiver "handshake":

- agree to establish connection (each knowing the other willing to establish connection)
- agree on connection parameters



```
Socket clientSocket =
  newSocket("hostname","port
  number");
```

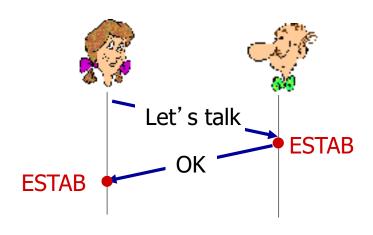
```
connection state: ESTAB
connection Variables:
seq # client-to-server
server-to-client
rcvBuffer size
at server,client

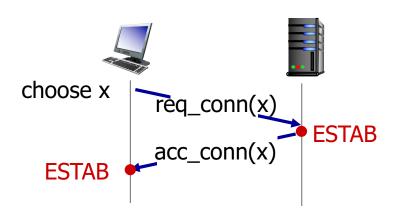
network
```

```
Socket connectionSocket =
  welcomeSocket.accept();
```

### Agreeing to establish a connection

#### 2-way handshake:

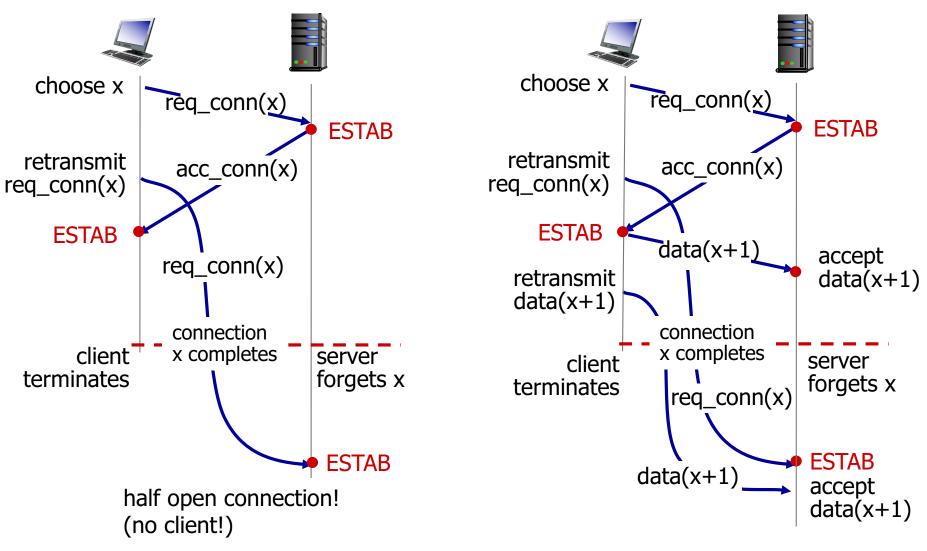




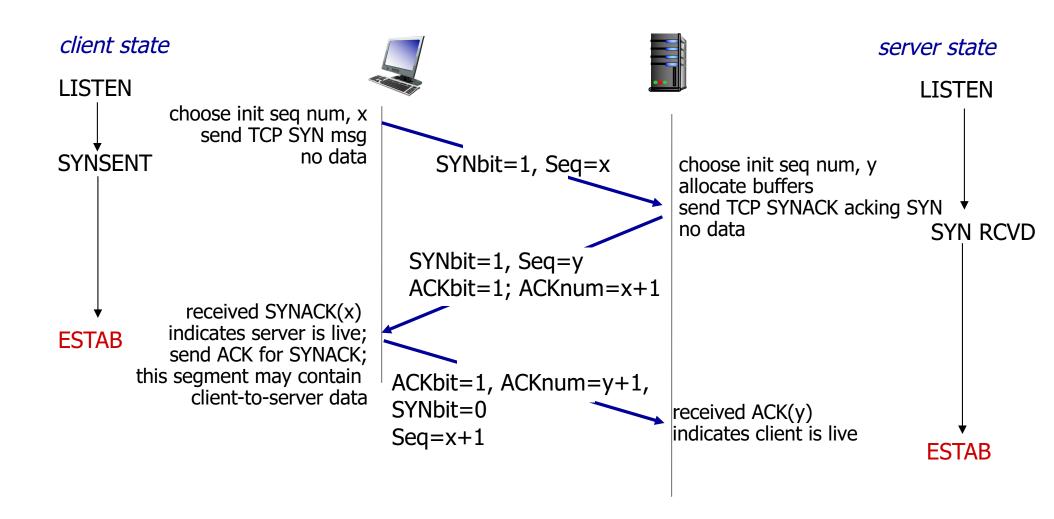
- Q: will 2-way handshake always work in network?
- variable delays
- retransmitted messages (e.g. req\_conn(x)) due to message loss
- message reordering
- can't "see" other side

### Agreeing to establish a connection

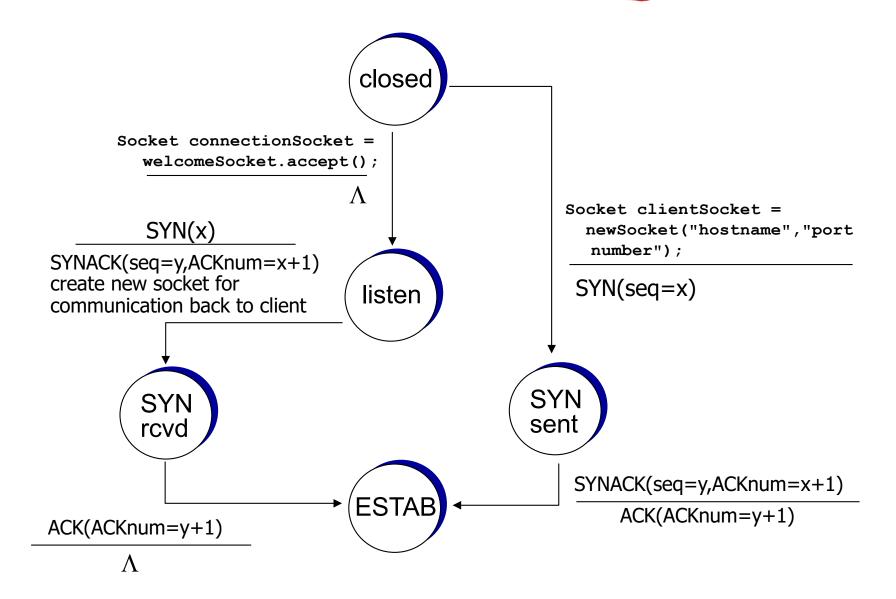
#### 2-way handshake failure scenarios:



### TCP 3-way handshake



### TCP 3-way handshake: FSM

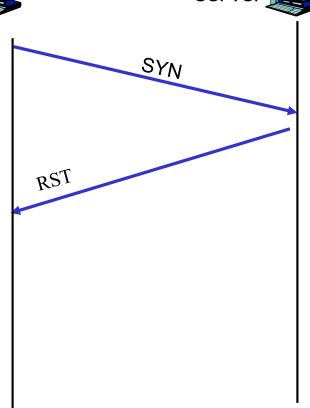


#### Server may not accept the connection

#### Why?

 Server may not be accepting TCP connections § to that port

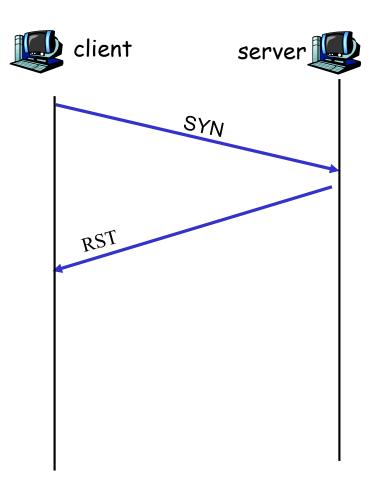
- Server may be out of resources
- What happens?
  - Sends RST
  - No connection established, no resources allocated
  - But client learns that port is open
- UDP servers do not have connections
  - just listen to a socket on a dest port#
  - If server receives a UDP packet with dest port# that does not match an existing UDP socket → Sends ICMP message back



### Attack 0: Scanning ports

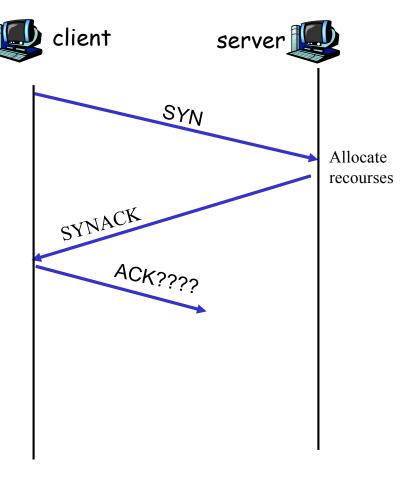
www.nmap.org

- Scanning TCP ports
  - Send TCP SYN
  - receive SYNACK, RST, nothing
- Scanning UDP ports
  - Receive ICNP messages



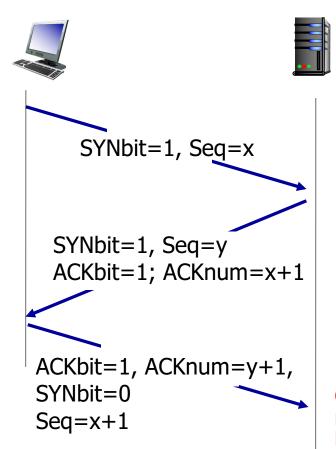
### Attack I: SYN FLOOD attacks

- Client never completes the handshake
- Server allocates resources
- SYN FLOOD: the oldest (D)DoS attack
  - Server does not have resources to server legit clients
- A Solution: SYN Cookies
  - Server creates a cookie:=a hash of the client IP, port on the SYN segment (and of a secret number known to the server
  - Server sends SYNACK with that initial seqno=cookie, but does NOT allocates resources (half open connection)
  - If client is legit, it will send an ACK with initialseqno; server can verify that ACK corresponds to SYN; only then sender allocates resources (full open connection)



## Attack II: Spoofing

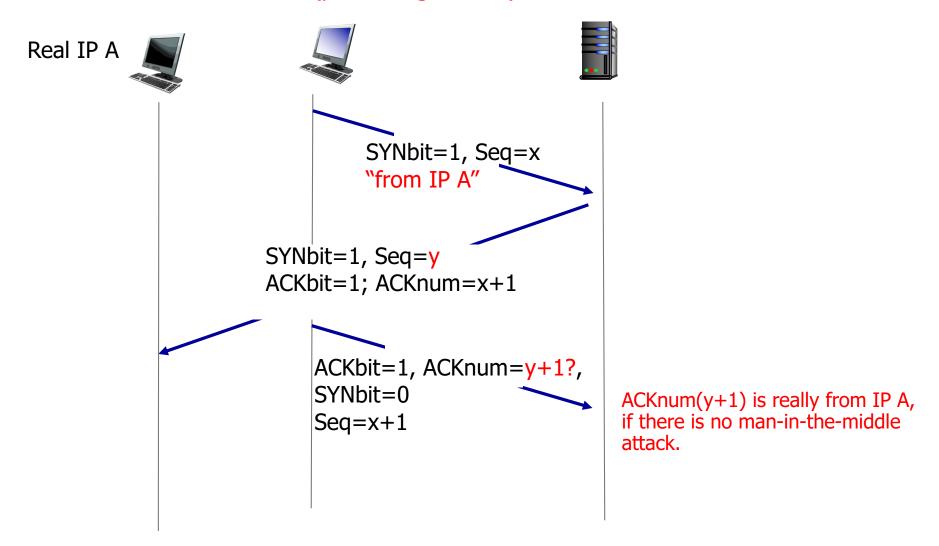
#### Client with IP A



Can the server be sure that this is really A, and not B pretending to be A?

# Attack II: Spoofing

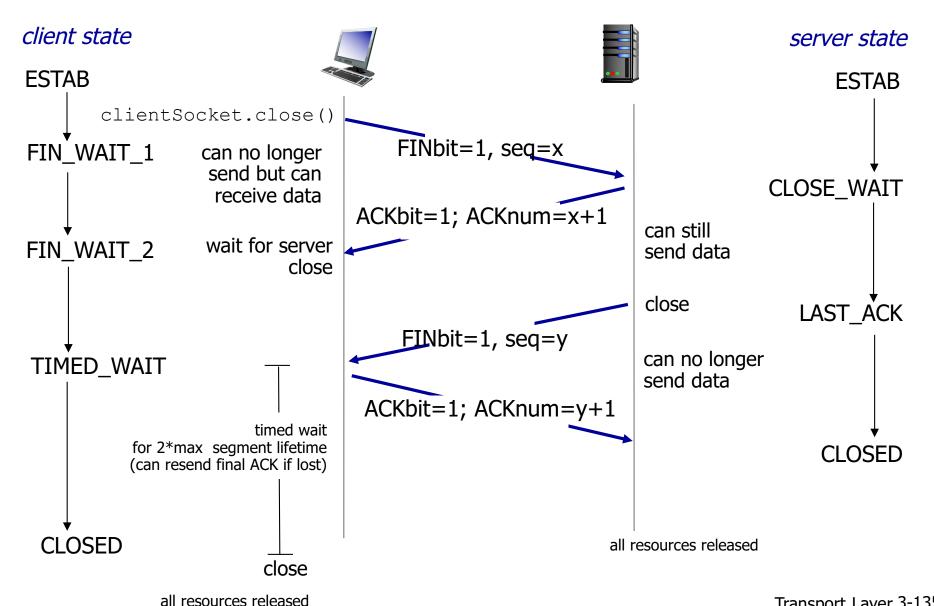
Attacker with IP B (pretending to be A)



## TCP: closing a connection

- remember: this is a duplex connection
- client, server each close their side of connection
  - send TCP segment with FIN bit = I
  - either of the two can initiate the closing
- respond to received FIN with ACK
  - on receiving FIN, ACK can be combined with own FIN
- simultaneous FIN exchanges can be handled

# TCP: closing a connection



### TCP Connection States

