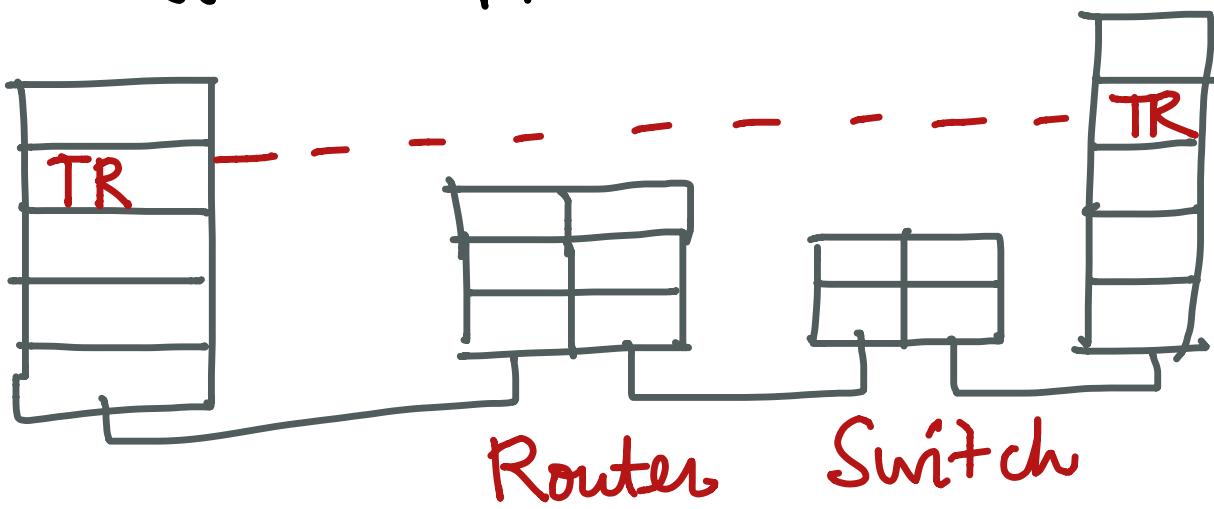
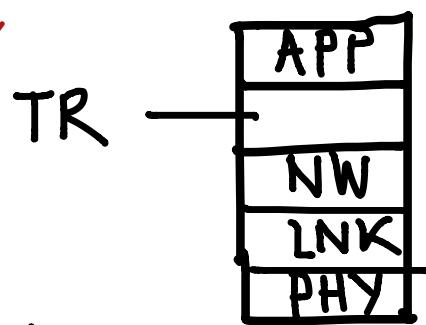


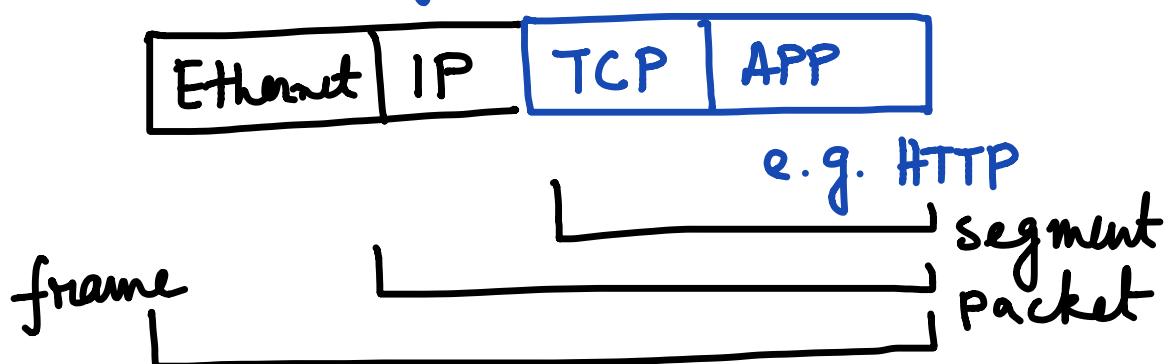
# Transport Layer

Task: deliver data end-to-end for different applications



Layer that gives end-to-end service.

Transport layer messages are "segments"



## Transport Layer Services

Unreliable messages  
reliable bytestreams

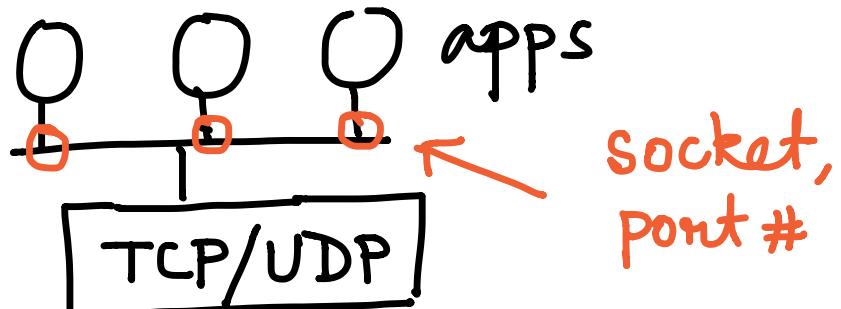
Implemented via  
UDP  
TCP

reliability does NOT mean it can't lose packets, but it is capable of handling such losses yet get the information across.

## Comparison

Telephone	Verbal
TCP	UDP
Connection-oriented	connectionless
Arbitrary length	limited size
flow control used	sends regardless of receiver status
congestion control used	sends regardless of network status

Sockets provide a simple abstraction of the network to the application



Socket API uses object called  
Sockets

can be used for streams and  
datagrams → same API

## Socket API functions (Recap)

Socket - create a new communication endpoint

Bind - a port to a socket

Listen - willing to connect

Accept - establish connection (passive)

Connect - " " (active)

[Send (to) - Send data

Receive (from) - receive data

Close

---

## Ports

Application identifier =

(IP address, protocol, port)

ports are numbered with 16 bits

Analogous to email mailboxes

Servers bind to special ports

<1024, requires admin privileges

Clients bind to "ephemeral" ports

does not need special privileges

Well known ports for server applications

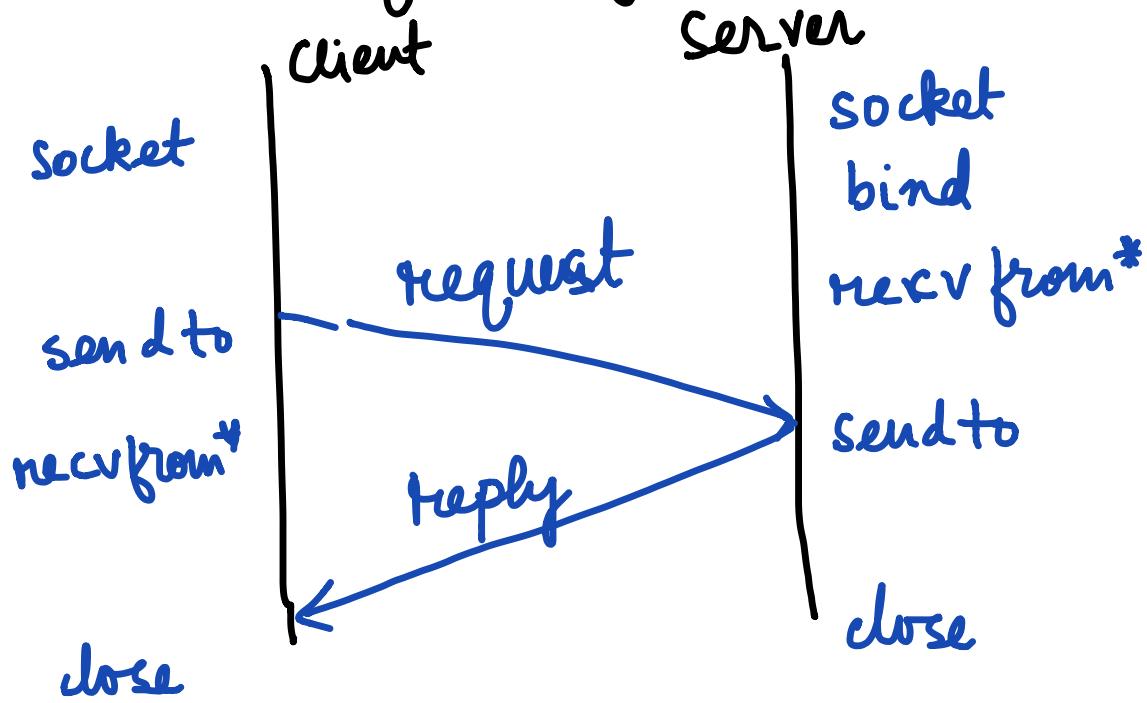
20, 21	- FTP
80	- HTTP
22	- SSH
25	- SMTP
.....	
443	- HTTPS

## User Datagram Protocol (UDP)

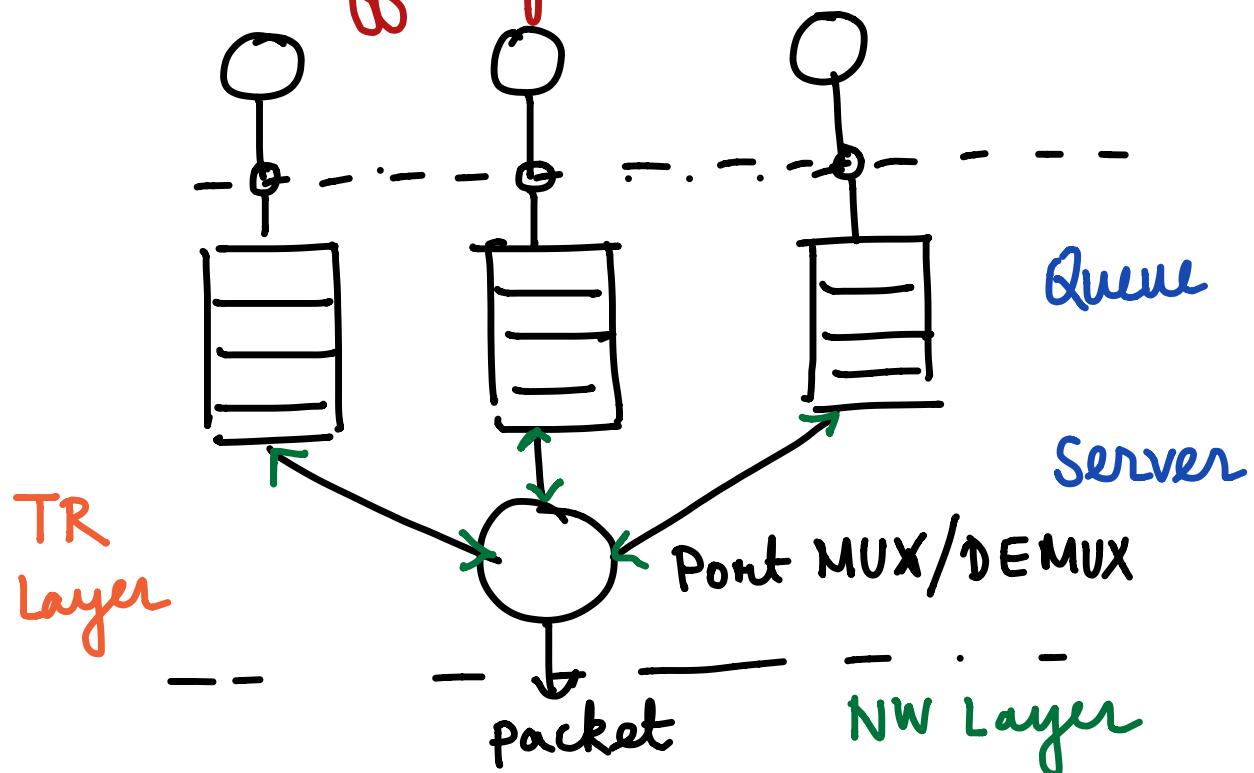
### Features

- datagram/message (contrast to bytestream)  
used for applications  
that don't need reliability of stream
- VOIP - no time to recover
- DNS - can request multiple times
- DHCP - bootstrapping

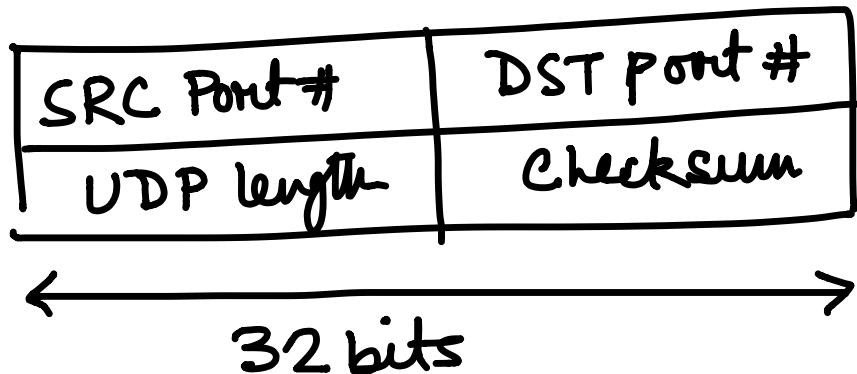
## UDP sequence of events



## UDP buffering



## UDP header



## Connection oriented service

[Connection Establishment      ] Telephone  
[Connection Release            ] call.

process to setup the infrastructure  
to send and receive data (end to end)

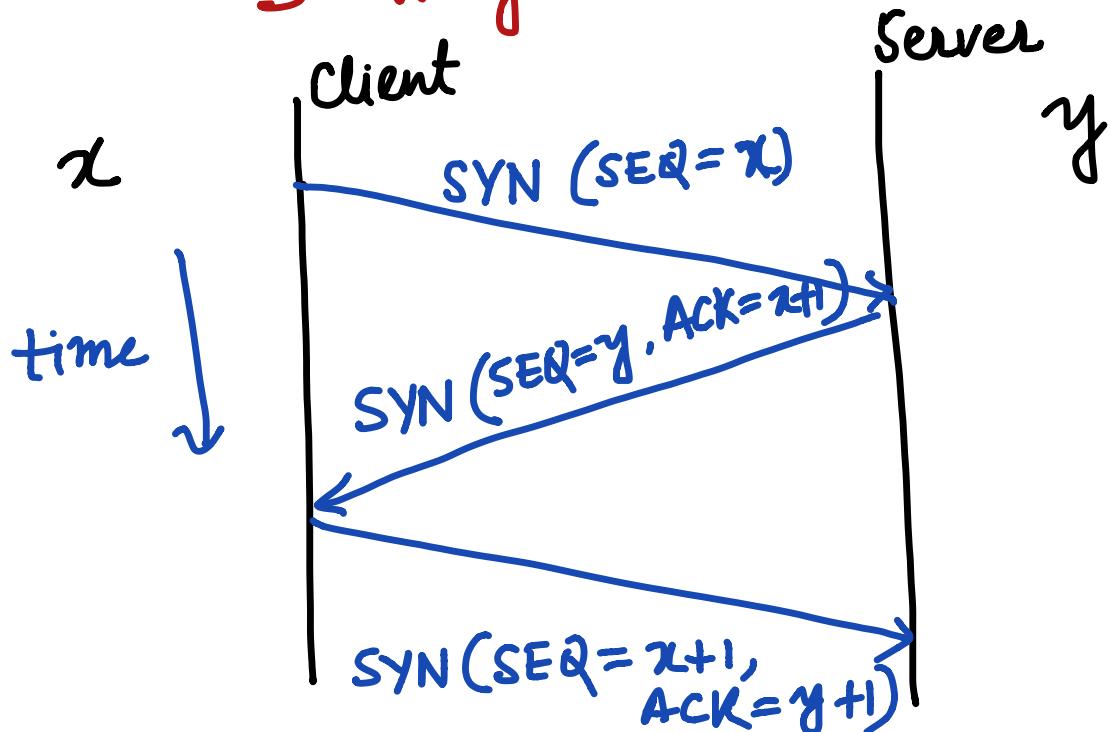
### Step 1

Agreement on the parameters  
e.g., max segment size

— SIGNALING

## TRANSMISSION CONTROL PROTOCOL

### 3-way handshake



Both the parties pick an arbitrary sequence number before setting up a connection

- SEQ, ACK are fields of TCP header
- SEQ numbers can be arbitrary

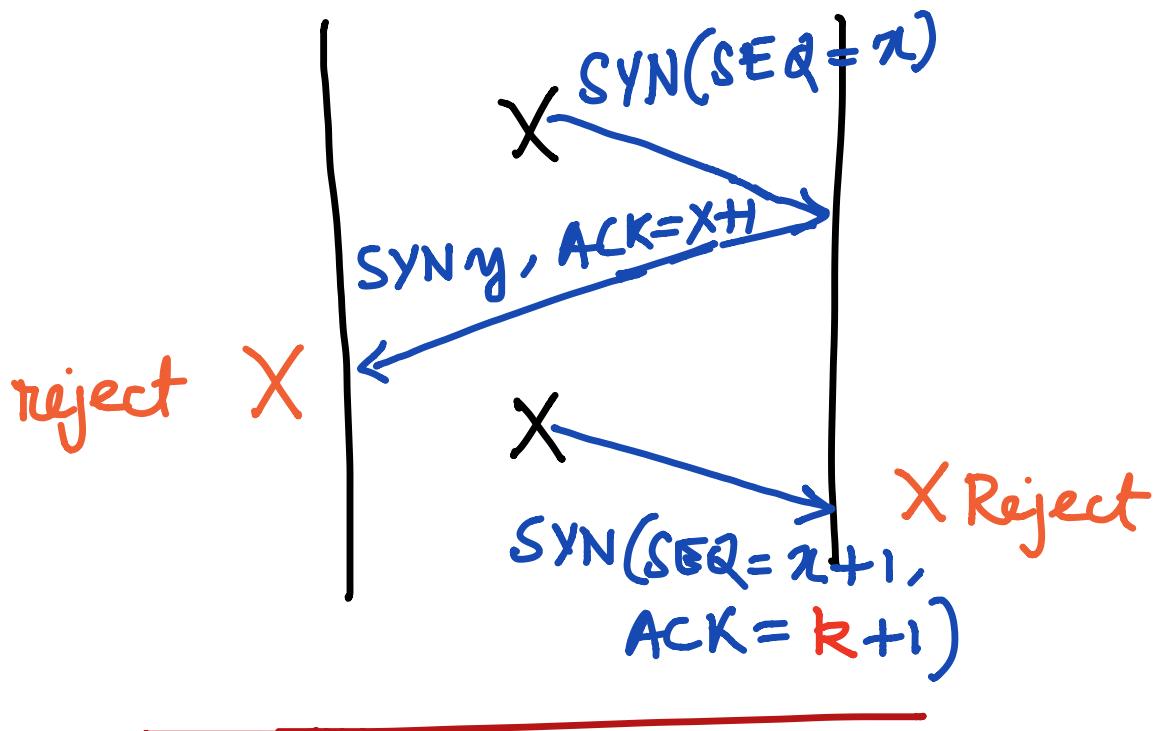
## Robustness

Easy case - any of the packets lost in transit

after a timeout, duplicate packet is sent

- lost packet doesn't reappear

Harder case - lost packets reappear



Also, both sides can open a TCP connection

need to keep track of the sequence and ack numbers

- finally after 3-way handshake both connections with diff SEQ & ACK #'s are established.
- 

## Connection Release

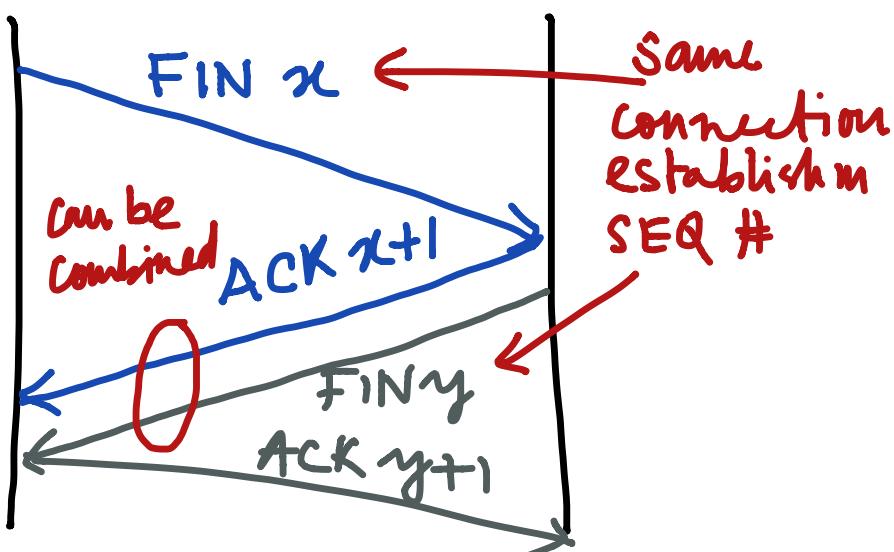
Why? need to successfully deliver all data before tear down

- also clean up the states.

orderly manner

## TCP symmetric release

party initiating the teardown



## TCP sliding window algorithm

Widely known for guaranteeing reliability.

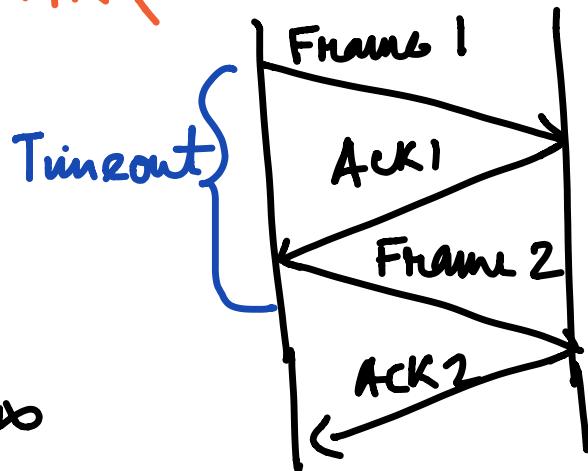
- modified stop-and-wait ARQ

Recap: Stop-Wait ARQ

Good for small network

delay is negligible

but the wastage is too high for larger NW.



Recall the numerical example from last time.

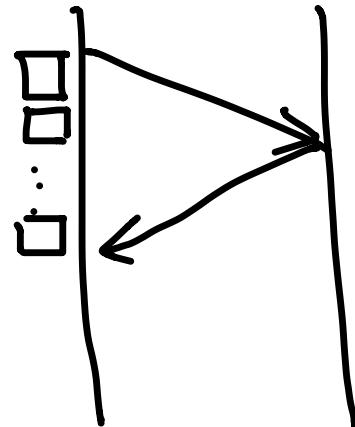
## Sliding window, or frames/segment

- allows  $W$  packets to be outstanding
- can send  $W$  packets in one RTT
- improves BW usage
- needs  $W = 2BD$  to fill network path

Ex.  $R = 1 \text{ Mbps}$ ,  $D = 50 \text{ ms}$

$W = 100 \text{ K bits} \sim 10 \text{ packets}$   
if a packet is of  $10 \text{ K bits}$

if  $R = 10 \text{ Mbps}$   
 $100 \text{ packets}$   
Changing the window.



---

### More detail:

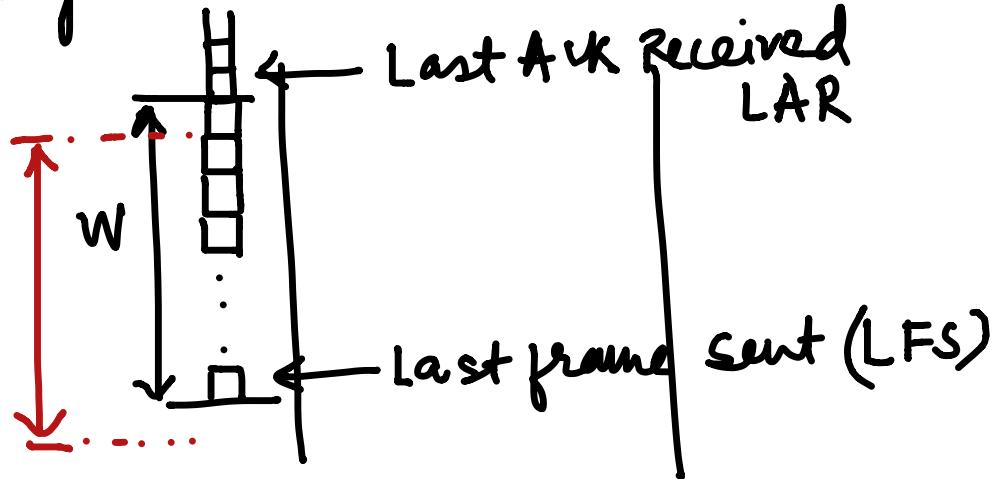
many variations of sliding window available depending on

- Buffer size, ACK, Retransmission

Go back N

Selective repeat

Sliding window - Sender side



IF  $LFS - LAR = W$

WAIT until next ACK

ELSE IF  $LFS - LAR < W$

SEND one packet

WHEN ACK RECEIVED

SLIDE WINDOW

Receiver Side

Go-back-N

Last ACK sent

(LAS)

packet seq

Waiting for the next  
one buffer

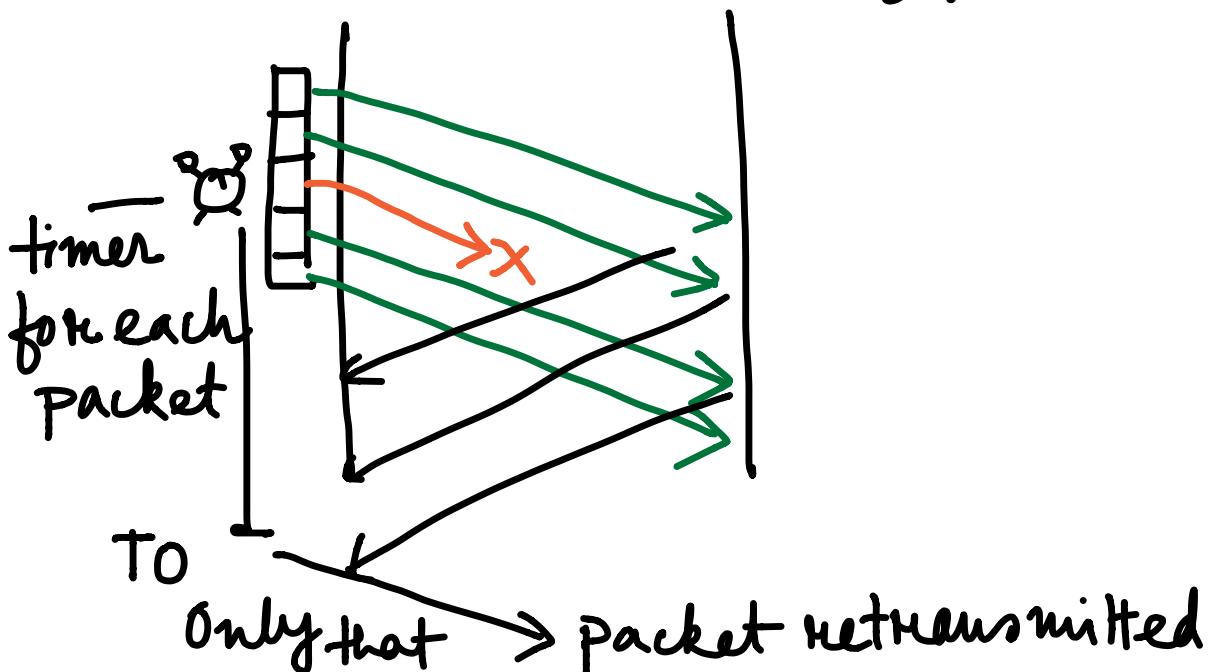
This protocol discards any out of order packets

Q: If one packet was lost, but later ones went successfully?

Go-back-n will repeat sending them all - hence the name (from LAR+1)

### Selective Repeat

- larger buffer - W (same as window)
- keeps track of all packets received
- sends ACKs accordingly



Seq numbers needed

Stop and Wait - only 1 bit

Selective Repeat -  $W$  for current window packets and  $W$  for past packets' ACKs

$$= 2W$$

Go-back-N =  $W + 1$

only for one outstanding ACK