

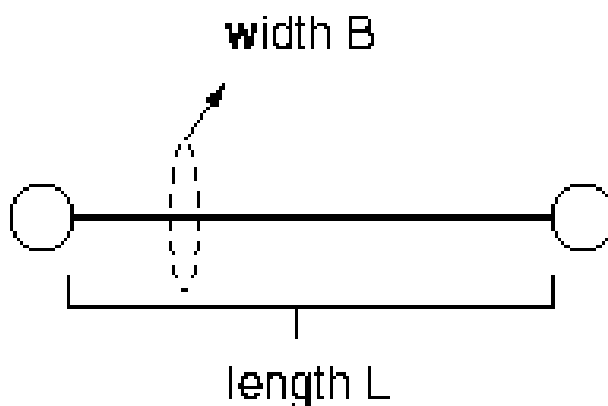
## LINK LAYER: BASIC TECHNIQUES

### Data Transmission

Link speed unit: bps

- abstraction
- ignore carrier frequency, coding etc.

Simplest case: point-to-point link



- wired or wireless

Interested in *completion time*:

→ time elapsed between sending/receiving first bit

→ i.e., how long will it take?

- Single bit:

→  $\approx L/SOL$  (lower bound)

→ latency (or propagation delay)

→ optical fiber, wireless: exact

- Multiple, say  $S$ , bits:

→  $\approx L/SOL + S/B$

→ latency + transmission time

Latency vs. transmission time: which dominates?

→ a lot to send, a little to send, ...

→ satellite, Zigbee, WLAN, broadband WAN

## Reliable Transmission

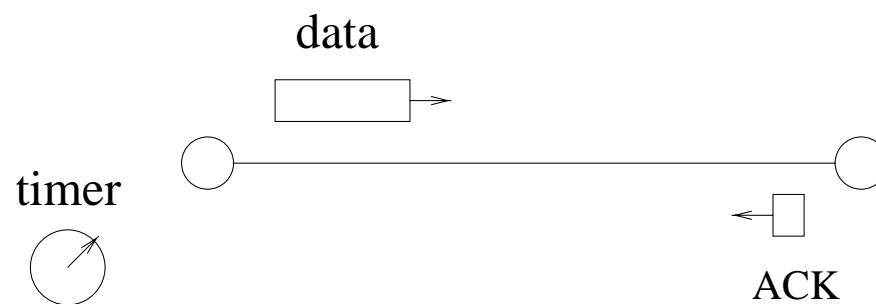
Main method: ARQ (Automatic Repeat reQuest)

- use retransmission
- used in both wired/wireless

- function duplication
  - link layer, transport layer, etc.
- alternative: FEC (forward error correction)
  - transmit redundant information
  - not assured
  - pros and cons?

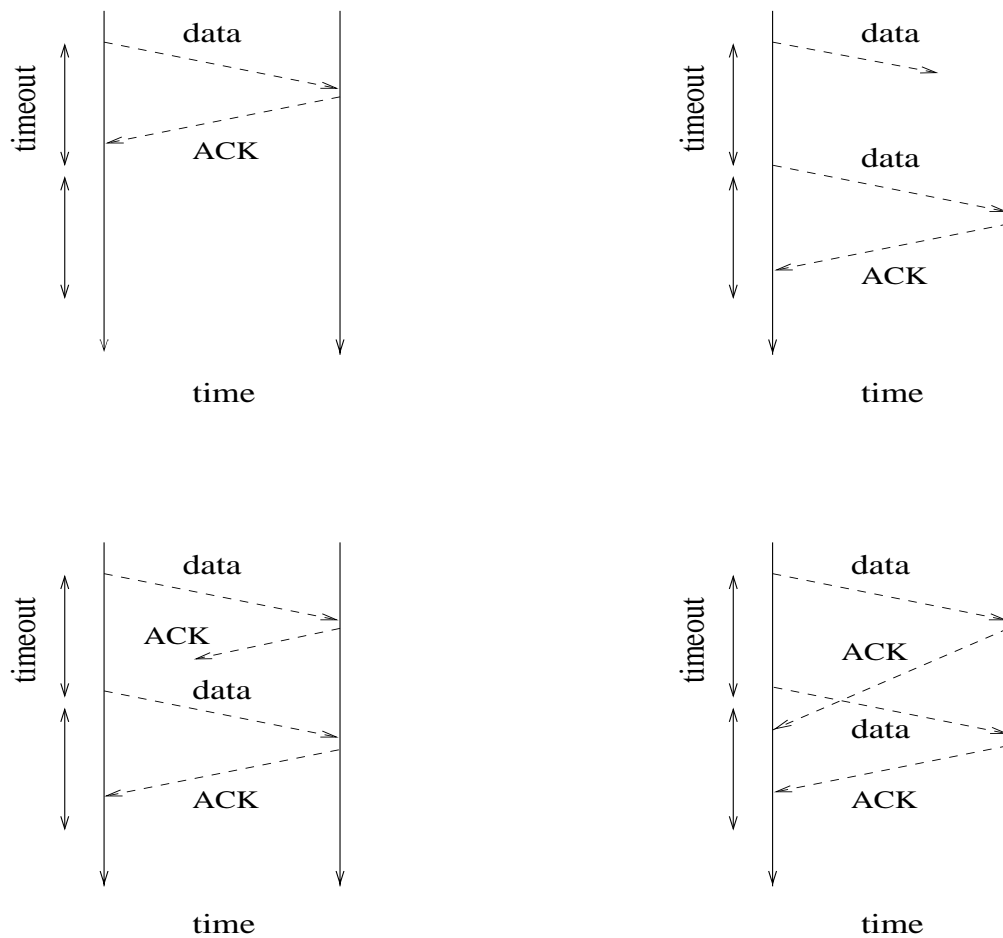
ARQ: three components

- timer
- acknowledgment (ACK)
- retransmit



Special case: stop-and-wait

Handle one packet (i.e., frame) at a time.



Issue of RTT (Round-Trip Time) & timer management:

- what is proper value of timer?
  - RTT estimation
- easier for single link
  - RTT is more well-behaved
- more difficult for multi-hop path in internetwork
  - latency + queueing effect

A “good” thing about stop-and-wait:

→ simple throughput formula

Stop-and-wait throughput (bps):

- RTT
- frame size (bits)

$$\longrightarrow \text{throughput} = \text{frame size} / \text{RTT}$$

Another important problem: not keeping the pipe full.

→ delay-bandwidth product

→ volume of data travelling on the link

High throughput: want to keep the pipe full

**Ex.:** Link BW 1.5 Mbps, 45 msec RTT

- if frame size 1 kB, then throughput:
  - $1024 \times 8 / 0.045 = 182 \text{ kbps}$
  - utilization: only  $182 \text{ kbps} / 1500 \text{ kbps} = 0.121$
- note: delay-bandwidth product
  - $1.5 \text{ Mbps} \times 45 \text{ msec} = 67.5 \text{ kb} \approx 8 \text{ kB}$

What happens to utilization if RTT increases to 90 msec?

What happens if bandwidth increases to 3 Mbps?

→ how to reduce bandwidth wastage?

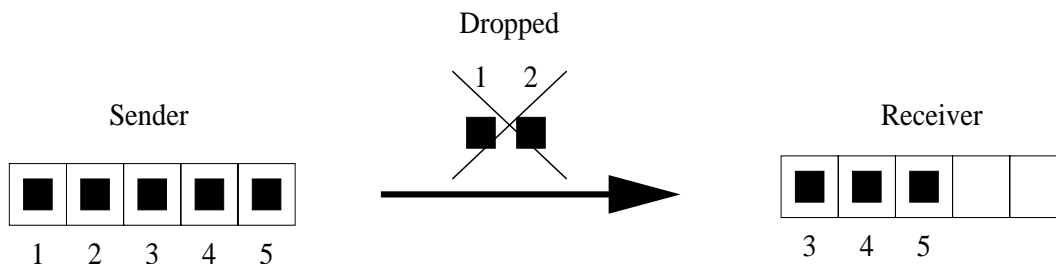


## Sliding Window Protocol

→ send block (i.e., window) of data

Issues:

- Shield application process from reliability management chore
  - exported semantics: continuous data stream
  - simple app abstraction: e.g., **read** system call
- Both sender and receiver have limited buffer capacity
  - task: plug holes & flush buffer



Simple solution when receiver has infinite buffer capacity:

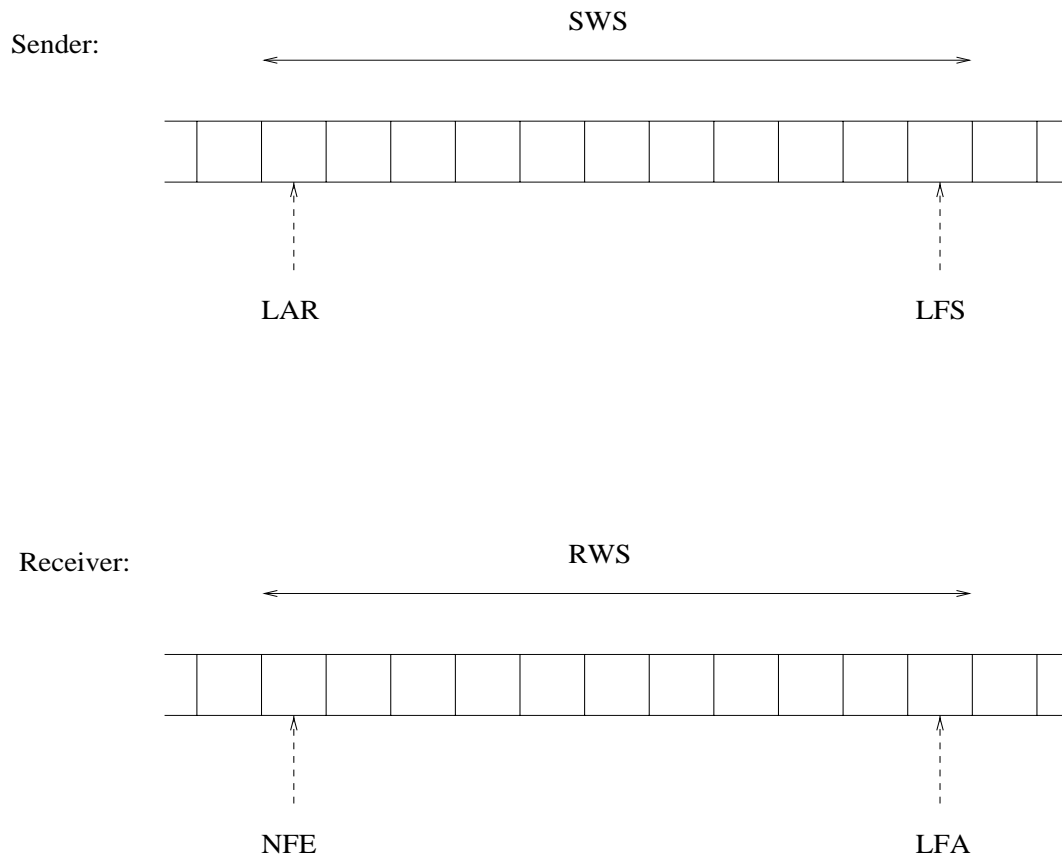
- sender keeps sending at maximum speed
- receiver informs sender of holes
  - “I’m missing this and that”
  - called negative ACK
- sender retransmits missing frames

Drawbacks?

What about positive ACK?

→ pros and cons

## Sliding window operation with positive ACK:



- *SWS*: Sender Window Size (sender buffer size)
- *RWS*: Receiver Window Size (receiver buffer size)
- *LAR*: Last ACK Received
- *LFS*: Last Frame Sent
- *NFE*: Next Frame Expected
- *LFA*: Last Frame Acceptable

Assign sequence numbers to frames.

→ IDs

Maintain invariants:

- $LFA - NFE + 1 \leq RWS$
- $LFS - LAR + 1 \leq SWS$

Sender:

- Receive ACK with sequence number  $X$
- Forwind LAR to  $X$
- Flush buffer up to (but not including) LAR
- Send up to  $SWS - (LFS - LAR + 1)$  frames
- Update LFS

Receiver:

- Receive packet with sequence number  $Y$
- Forward to (new) first hole & update NFE  
→ NFE need not be  $Y + 1$
- Send cumulative ACK (i.e., NFE)
- Flush buffer up to (but not including) NFE to application
- Update  $LFA \leftarrow NFE + RWS - 1$

Sequence number wrap-around problem:

$$\text{SWS} < (\text{MaxSeqNum} + 1)/2$$

- why?
- consider special case: stop-and-wait
- is sequence number needed?