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16.36 Communication Systems Engineering
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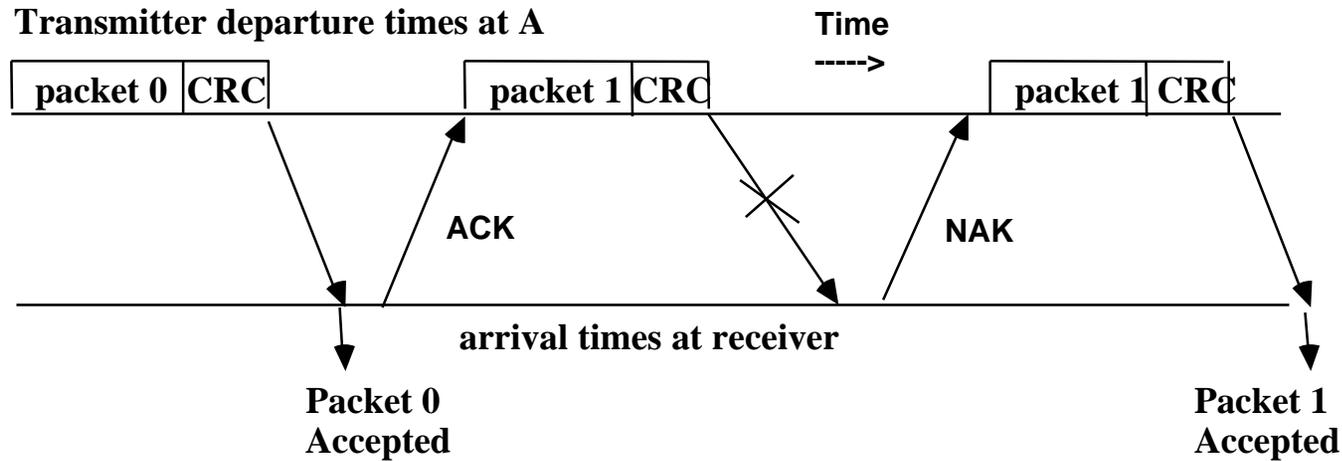
16.36: Communication Systems Engineering

ARQ Protocols: Stop & Wait

Automatic Repeat ReQuest (ARQ)

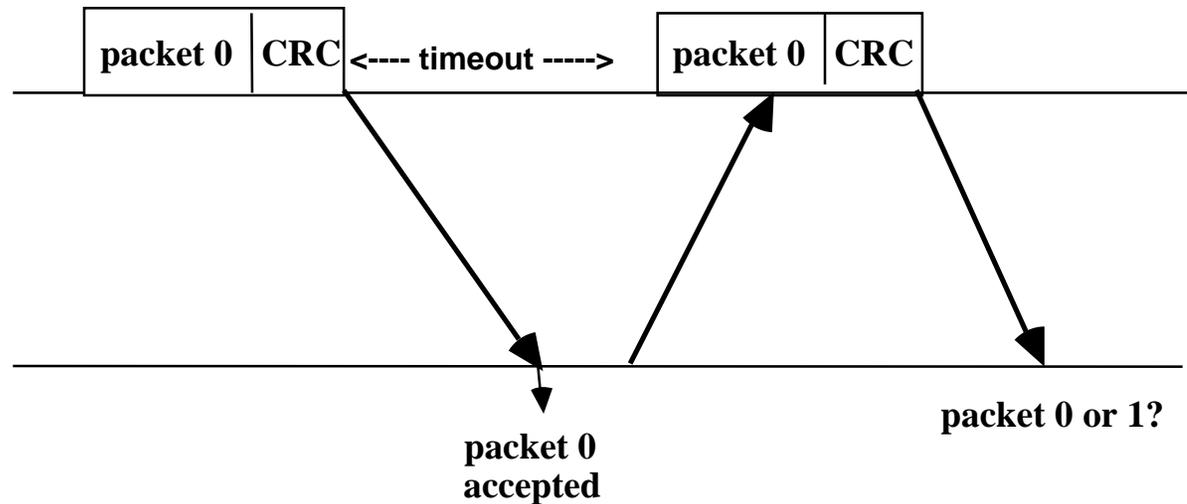
- **When the receiver detects errors in a packet, how does it let the transmitter know to re-send the corresponding packet?**
- **Systems which automatically request the retransmission of missing packets or packets with errors are called ARQ systems.**
- **Three common schemes**
 - **Stop & Wait**
 - **Go Back N**
 - **Selective Repeat**
- **Byzantine Army Problem**
 - **Byzantine failure: the failure to correctly execute a step in an algorithm**

Pure Stop and Wait Protocol



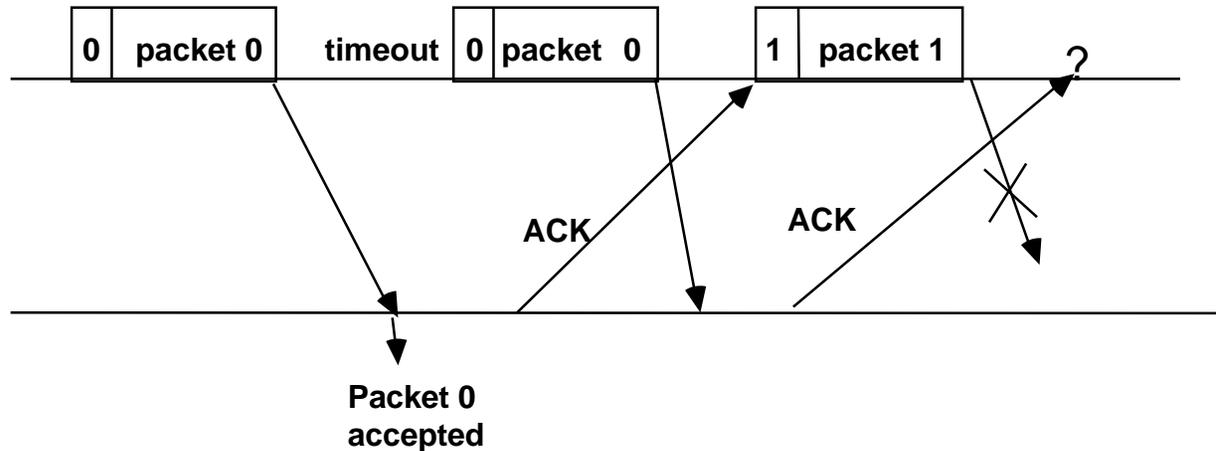
- **Problem: Lost Packets**
 - Sender will wait forever for an acknowledgement
- **Packet may be lost due to framing errors**
- **Solution: Use time-out (TO)**
 - Sender retransmits the packet after a timeout

The Use Of Timeouts For Lost Packets Requires Sequence Numbers



- **Problem:** Unless packets are numbered the receiver cannot tell which packet it received
- **Solution:** Use packet numbers (sequence numbers)

Request Numbers Are Required On ACKs To Distinguish Packet ACKed



- **REQUEST NUMBERS:**

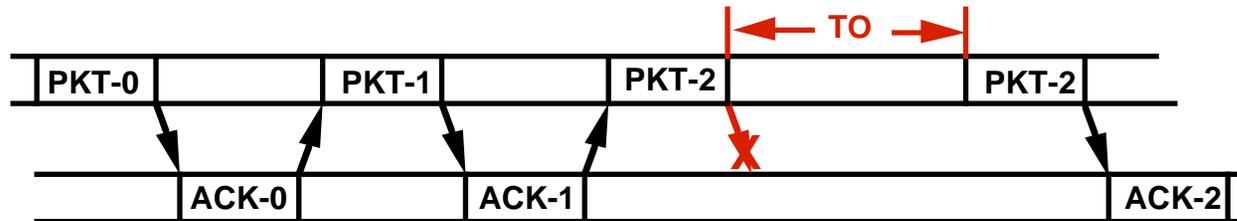
- Instead of sending "ack" or "nak", the receiver sends the number of the packet currently awaited.
- Sequence numbers and request numbers can be sent modulo 2.

This works correctly assuming that

- 1) Frames travel in order (FCFS) on links
- 2) The CRC never fails to detect errors
- 3) The system is correctly initialized.

The stop and wait protocol

- Original ARQ protocol
- Sender transmits one packet at a time and waits for an ACK
 - Receiver ACK's packets
 - Sender retransmits packet after a timeout



- Packet numbering
 - Sender numbers packets with sequence numbers (SN)
 - Receiver uses request numbers (RN) to ACK packets
 - $RN = j$ is the same as an ACK for packet $j-1$
- Note:
 - Transmitter idle while waiting for ACK
 - Efficiency limited by round trip delay time
 - Requires no storage of packets

Stop and Wait Protocol Algorithm at sender (node A)

(with initial condition SN=0)

- 1) Accept packet from higher layer when available;
assign number SN to it**
- 2) Transmit packet SN**
- 3) Wait for an error free packet from B**
 - i. if received and it contains $RN > SN$ in the
request # field, set SN to RN and go to 1**
 - ii. if not received within given time (TO), go to 2**

Stop and Wait

Algorithm at receiver (node B)

(with initial condition $RN=0$)

- 1) Whenever an error-free frame is received from A with a sequence # equal to RN, release received packet to higher layer and increment RN.**
- 2) At arbitrary times, but within bounded delay after receiving any error free frame from A, transmit a frame to A containing RN in the request # field.**

Correctness of Stop and Wait

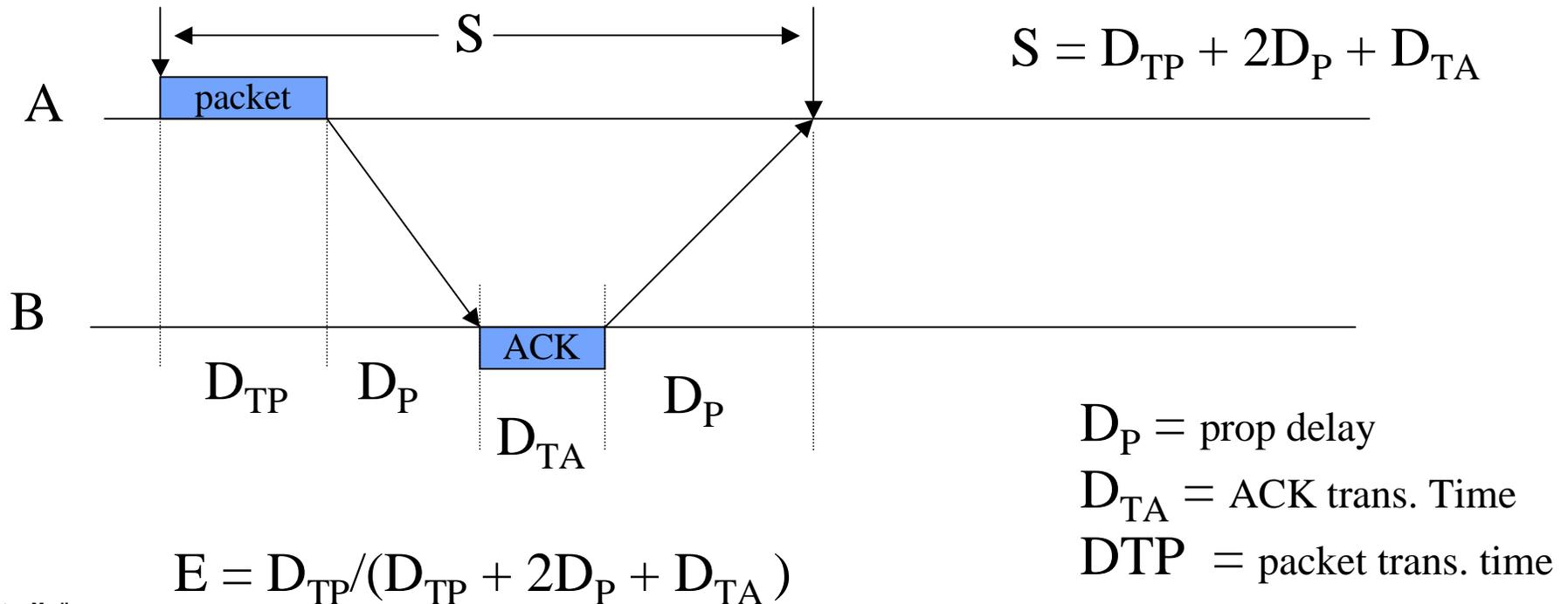
- **SAFETY:** show that no packet is ever released out of order or more than once
 - This is immediately obvious from the algorithm. We start with packet '0' and receiver does not increment its RN until '0' is received; upon which it can only accept packet '1', etc.
- **LIVENESS:** show that every packet is eventually released
 - The sender keeps sending a packet until it gets an ack, so eventually every packet is correctly received
- **Packet numbering:** packets are numbered modulo 2 (0 or 1)
 - Start with packet 0, then 1, then 0, then 1, etc...
 - This works because at any time, the received packet can only be either the packet that the receiver is waiting for (i.e., $SN = RN$) or the previous packet which has already been received (i.e., $SN = RN-1$). Mod 2 numbering can be used to distinguish between these packets.

Efficiency of stop and wait

Let S = total time between the transmission of a packet and reception of its ACK

D_{TP} = transmission time of the packet

Efficiency (no errors) = D_{TP}/S



Stop and wait in the presence of errors

Let P = the probability of an error in the transmission of a packet or in its acknowledgment

$$S = D_{TP} + 2D_P + D_{TA}$$

TO = the timeout interval

X = the amount of time that it takes to transmit a packet and receive its ACK. This time accounts for retransmissions due to errors

$$E[X] = S + TO * P / (1 - P),$$

$$\text{Efficiency} = D_{TP} / E[X]$$

Where,

TO = D_{TP} in a full duplex system

TO = S in a half duplex system