



Data Link Control Layer, Error Detection, Error Correction, and Framing

EEE 538, WEEK 2

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Error Detection Techniques



- Used by the receiver to determine if a packet contains errors
- If a packet is found to contain errors the receiver requests the transmitter to re-send the packet
- Error Detection Techniques
 - Parity Check
 - Single bit
 - Horizontal and vertical redundancy check
 - Cyclic Redundancy Check (CRC)
- Assume initially that the receiving DLC module knows where frames begin and end





Effectiveness of an Error Detection Technique



- **minimum distance of code (d)** (min # bit errors undetected) The minimum distance of a code is the smallest number of errors that can map one codeword onto another. If fewer than d errors occur they will always be detected. Even more than d errors will often be detected (but not always!)
- **burst detecting ability (B)** (max burst length always detected)
- **probability of random bit pattern mistaken as error free** (good estimate if # errors in a frame $\gg d$ or B)





Parity Check Codes



K data bits

L parity check bits

- Each parity check is a modulo 2 sum of some of the data bits
- Example
 - $C1 = x1 + x2 + x3$
 - $C2 = x2 + x3$
- Internet checksum





Single Parity Check Code



- The check bit is 1 if frame contains odd number of 1's; otherwise it is 0
 - 1011011 -> 1011011 1
 - 1100110 -> 1100110 0
- Thus, encoded frame contains even number of 1's
- Receiver counts number of ones in frame
 - An even number of 1's is interpreted as no errors
 - An odd number of 1's means that an error must have occurred. A single error (or an odd number of errors) can be detected. An even number of errors cannot be detected. Nothing can be corrected
 - Probability of undetected error (independent errors)

$$P(\text{undetected}) = \sum_{i \text{ even}} \binom{N}{i} p^i (1-p)^{N-i} \quad \begin{array}{l} N = \text{packet size} \\ p = \text{error prob.} \end{array}$$



Horizontal and Vertical Parity



1	0	0	1	0	1	0	1		
0	1	1	1	0	1	0	0		
1	1	1	0	0	0	1	0		
1	0	0	0	1	1	1	0		
0	0	1	1	0	0	1	1		
1	0	1	1	1	1	1	0		

Horizontal checks

1	0	0	1	0	1	0	1		
0	1	1	1	0	1	0	0		
1	1	1	0	0	0	1	0		
1	0	0	0	1	1	1	0		
0	0	1	1	0	0	1	1		
1	0	1	1	1	1	1	0		

Vertical checks

- The data is viewed as a rectangular array (i.e., a sequence of words)
- Minimum distance=4, any 4 errors in a rectangular configuration is undetected





Cyclic Redundancy Check



M

C

K data bits	L parity check bits
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$$T = M 2^L + C$$

T: transmitted data, codeword

M: message, information bits

C: parity bits





Physical Layer Error Characteristics



- Most Physical Layers (communications channels) are not well described by a simple BER parameter
- Most physical error processes tend to create a mix of random & bursts of errors
- A channel with a BER of 10^{-7} and a average burst size of 1000 bits is very different from one with independent random errors
- Example: For an average frame length of 10^4 bits
 - random channel: $E[\text{Frame error rate}] \sim 10^{-3}$
 - burst channel: $E[\text{Frame error rate}] \sim 10^{-6}$
- Best to characterize a channel by its Frame Error Rate
- This is a difficult problem for real systems





Automatic Repeat Request ARQ

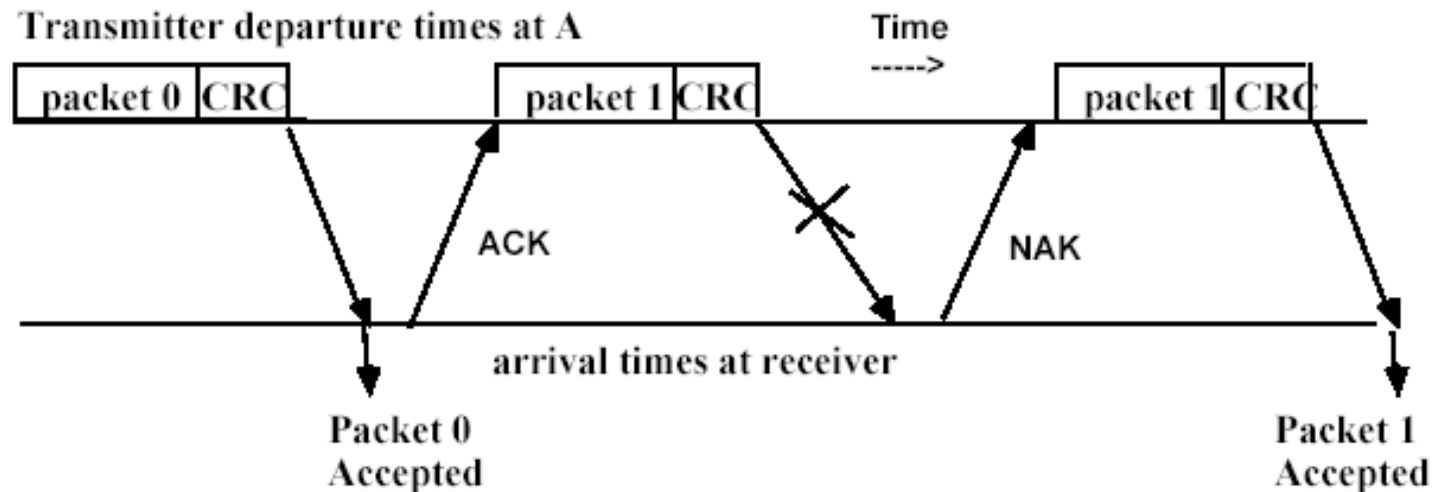


- When the receiver detects errors in a packet, how does it let the transmitter know to re-send the corresponding packet?
- Systems which automatically request the retransmission of missing packets or packets with errors are called ARQ systems.
- Three common schemes
 - Stop & Wait
 - Go Back N
 - Selective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





Pure Stop and Wait

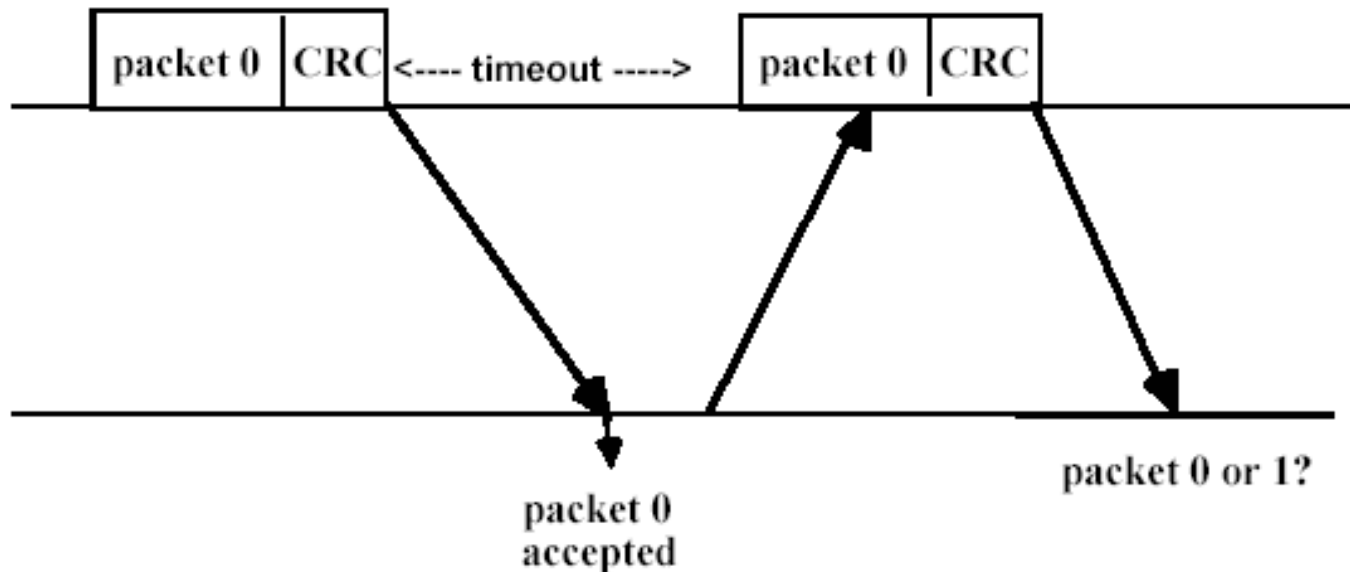


- Problem: Lost packets
- Sender will wait forever for an acknowledgement
- Packet may be lost due to framing errors
- Solution: Use time-out (TO)
- Sender retransmits the packet after a timeout





Sequence Numbers

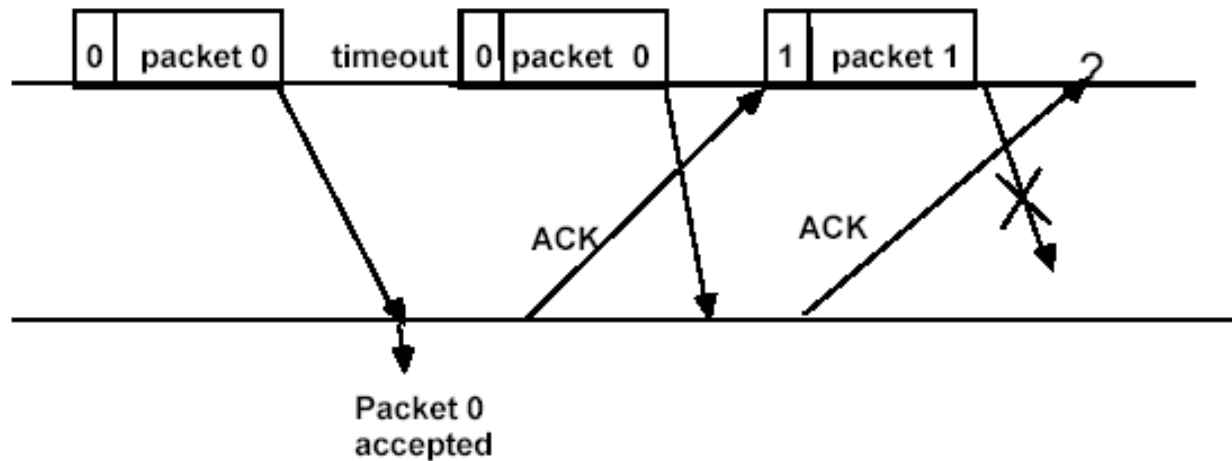


- Problem: Unless packets are numbered the receiver cannot tell which packet it received
- Solution: Use packet numbers (sequence numbers)





Request Numbers are Required on ACKs



- As opposed to sending an ACK or a NAK, the receiver sends the number of the packet currently awaited
- Sequence numbers and request numbers can be sent modulo 2, avoiding the need to use large sequence numbers





Algorithm at Sender

- Initial condition $SN = 0$
- 1) Accept packet from higher layer when available; assign number SN to it
- 2) Transmit packet SN in frame with sequence # SN
- 3) Wait for an error free frame 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, go to 2

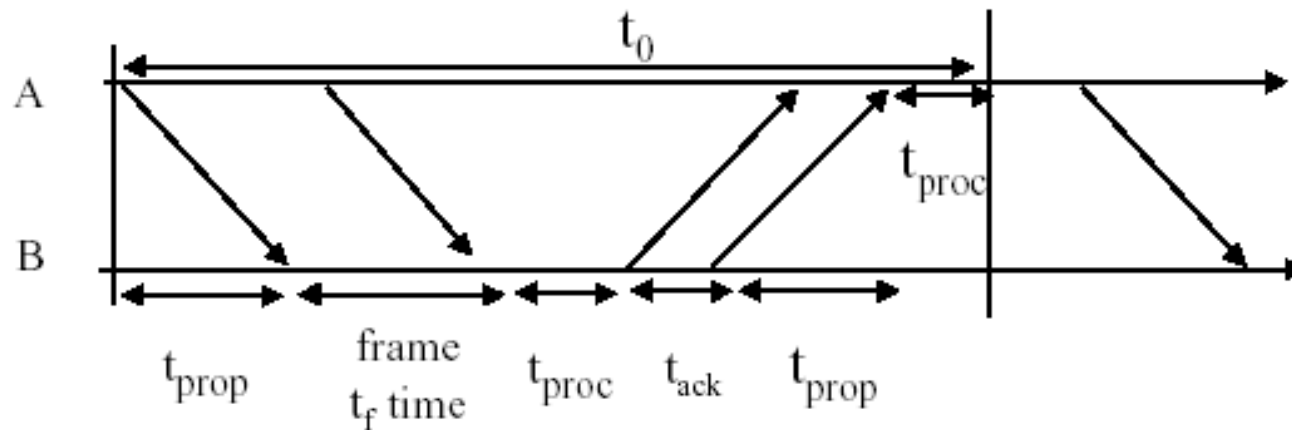


Algorithm at Receiver

- 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
- 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 # RN



Efficiency of Stop and Wait



t_{prop} = propagation delay

t_{proc} = processing delay

n_f = # bits in a frame (assume constant)

n_a = # bits in ACK/NAK frame

R = bit rate

Time to send frame (in absence of errors) and receive ACK:

$$t_0 = 2(t_{\text{prop}} + t_{\text{proc}}) + (n_f + n_a)/R$$



Stop and Wait



Effective transmission rate

$$R_{eff}^0 = \frac{\text{number of information bits delivered to destination}}{\text{total time required to deliver the information bits}} = \frac{n_f - n_o}{t_0},$$

Transmission efficiency

$$\eta_0 = \frac{\frac{n_f - n_o}{t_0}}{R} = \frac{1 - \frac{n_o}{n_f}}{1 + \frac{n_a}{n_f} + \frac{2(t_{prop} + t_{proc})R}{n_f}}.$$

Delay-Bandwidth Product





Stop and Wait



nframe = 8192
noverhead = 64
nack = 64

	R bps	3.00E+04	1.50E+06	4.50E+07	2.40E+09
tprop+tproc					
0.005		9.50E-01	3.50E-01	1.77E-02	3.39E-04
0.05		7.22E-01	5.14E-02	1.80E-03	3.39E-05
0.5		2.12E-01	5.39E-03	1.81E-04	3.39E-06

nframe= 524288

	R bps	3.00E+04	1.50E+06	4.50E+07	2.40E+09
tprop+tproc					
0.005		9.99E-01	9.72E-01	5.38E-01	2.14E-02
0.05		9.94E-01	7.77E-01	1.04E-01	2.18E-03
0.5		9.46E-01	2.59E-01	1.15E-02	2.18E-04

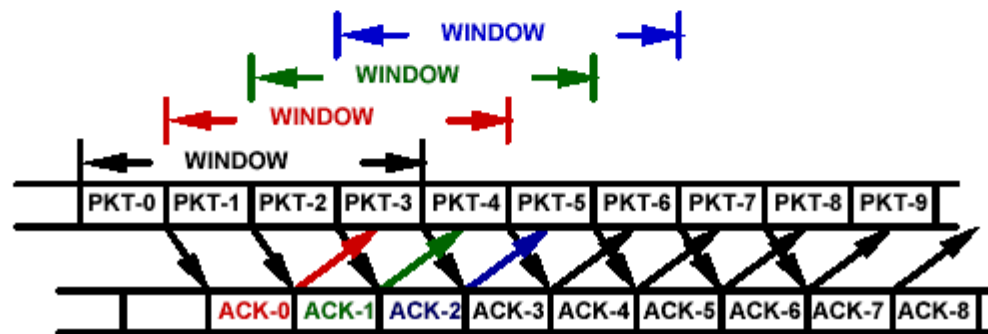




Go-back N



- Stop and Wait is inefficient when propagation delay is larger than the packet transmission time
 - Can only send one packet per round-trip time
- Go Back N allows the transmission of new packets before earlier ones are acknowledged
- Go back N uses a window mechanism where the sender can send packets that are within a “window” (range) of packets
 - The window advances as acknowledgements for earlier packets are received





Features of Go-back N



- Window size = N
 - Sender cannot send packet $i+N$ until it has received the ACK for packet i
- Receiver operates just like in Stop and Wait
 - Receive packets in order
 - Receiver cannot accept packet out of sequence
 - Send $RN = i + 1 \Rightarrow$ ACK for all packets up to and including i
- Use of piggybacking
 - When traffic is bi-directional RN's are piggybacked on packets going in the other direction
 - Each packet contains a SN field indicating that packet's sequence number and a RN field acknowledging packets in the other direction

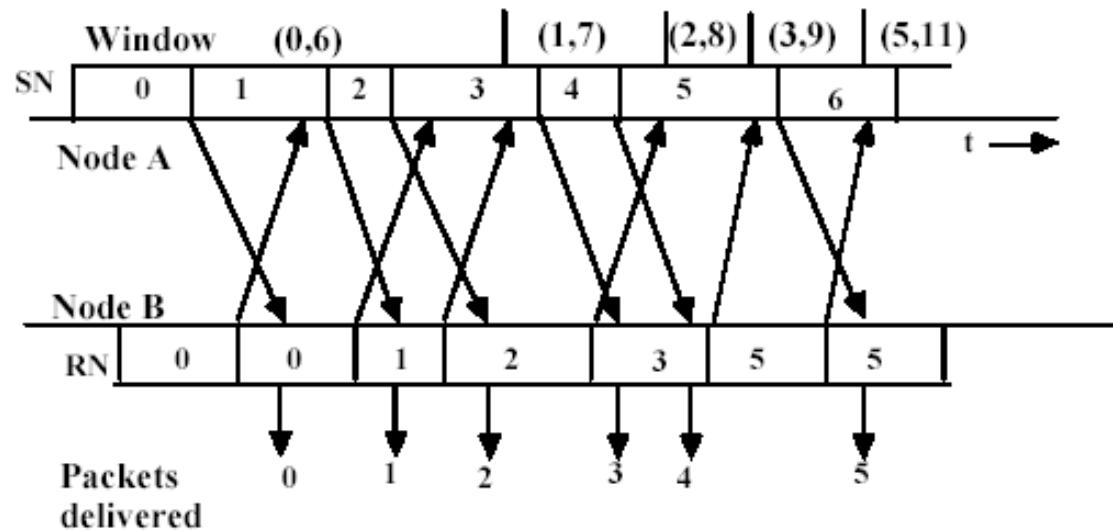
<--Frame Header ----->

	SN	RN		Packet	CRC
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Example Go-back 7

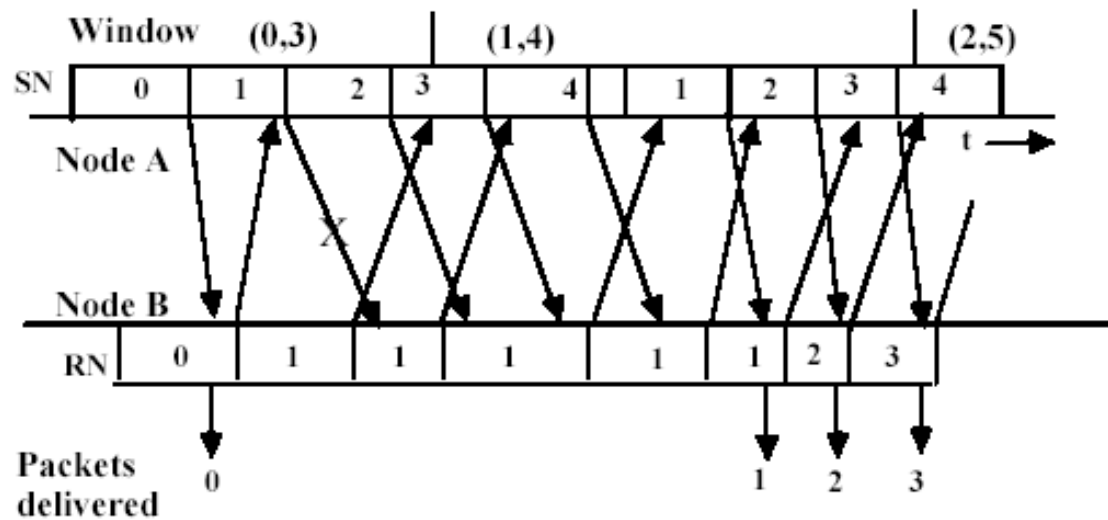


- Note that packet RN-1 must be accepted at B before a frame containing request RN can start transmission at B





Errors

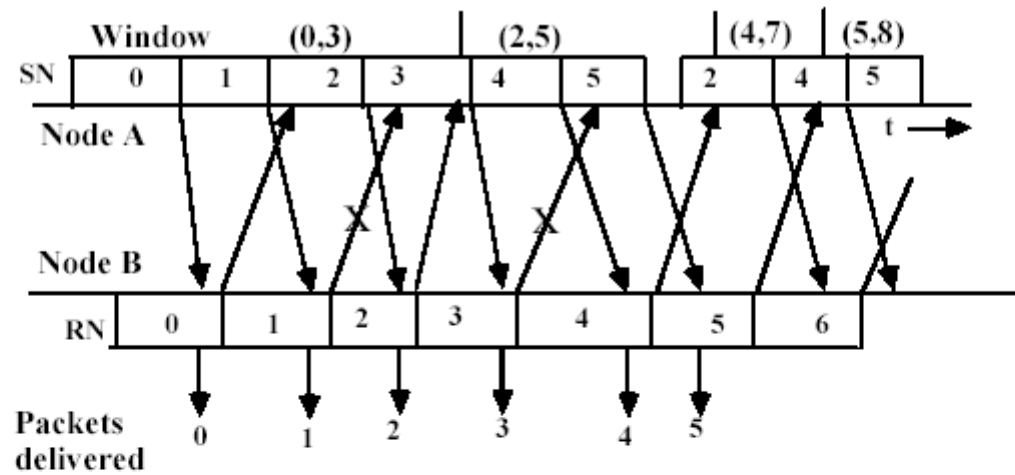


- Note that the timeout value here is take to be the time to send a full window of packets
- Note that entire window has to be retransmitted after an error





Example



- When an error occurs in the reverse direction the ACK may still arrive in time. This is the case here where the packet from B to A with RN=2 arrives in time to prevent retransmission of packet 0
- Packet 2 is retransmitted because RN = 4 did not arrive in time, however it did arrive in time to prevent retransmission of packet 3
 - Was retransmission of packet 4 and 5 really necessary?

Strictly no because the window allows transmission of packets 6 and 7 before further retransmissions. However, this is implementation dependent



Selective Repeat ARQ



- Selective Repeat attempts to retransmit only those packets that are actually lost (due to errors)
 - Receiver must be able to accept packets out of order
 - Since receiver must release packets to higher layer in order, the receiver must be able to buffer some packets
- Retransmission requests
 - Implicit

The receiver acknowledges every good packet, packets that are not ACKed before a time-out are assumed lost or in error

Notice that this approach must be used to be sure that every packet is eventually received
 - Explicit

An explicit NAK (selective reject) can request retransmission of just one packet

This approach can expedite the retransmission but is not strictly needed
 - One or both approaches are used in practice

