Redes de Computadores

The Data Link Layer

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- *what are the main functions of the Data Link layer?*
- » What is the difference between a packet and a frame?
- » What techniques can be used to frame data? Why is stuffing important?
- » What is the relationship between BER and FER?
- » What techniques can be used to detect errors?
- » How does the Internet checksum work?
- » How does a CRC work and what are its error detection capabilities?
- » What is the purpose of ARQ and what are the common ARQ mechanisms?
- » Why are sequence numbers required in the STOP&Wait ARQ? What is its efficiency?
- » What is a sliding window?
- » What are the main differences between the Go Back N ARQ and STOP&WAIT ARQ? What is the efficiency of the former?
- » What are the differences between the Selective Reject ARQ and the Go Back N ARQ mechanisms? How does this efficiency evolve with FER?
- What framing, stuffing, error detection and ARQ mechanisms are used in Ethernet, PPP and WLAN?
- » What are the differences between End-to-End ARQ and Link-by-Link ARQ?
- » Where are the ARQ mechanisms in the TCP/IP reference model?

Data Link layer functions and services

Data Link Layer – Functions and Services

Main functions

- » Provide service interface to the network layer
- » Eliminate/reduce transmission errors
- » Regulate data flow
 Slow receivers not swamped by fast senders
- Services provided
 - » Unacknowledged connectionless service
 - » Acknowledged connectionless service
 - » Acknowledged connection-oriented service

Application layer

Transport layer

3 Network layer

5

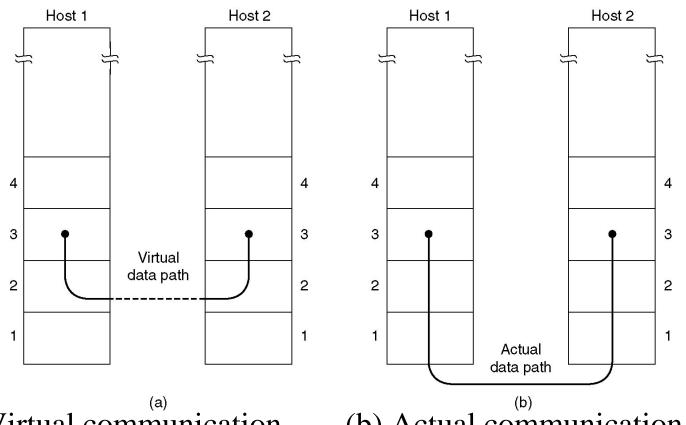
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Data Link layer

Physical layer

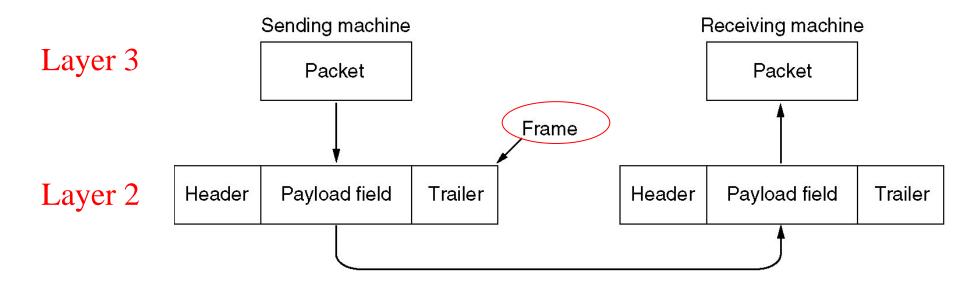
Services Provided to Network Layer



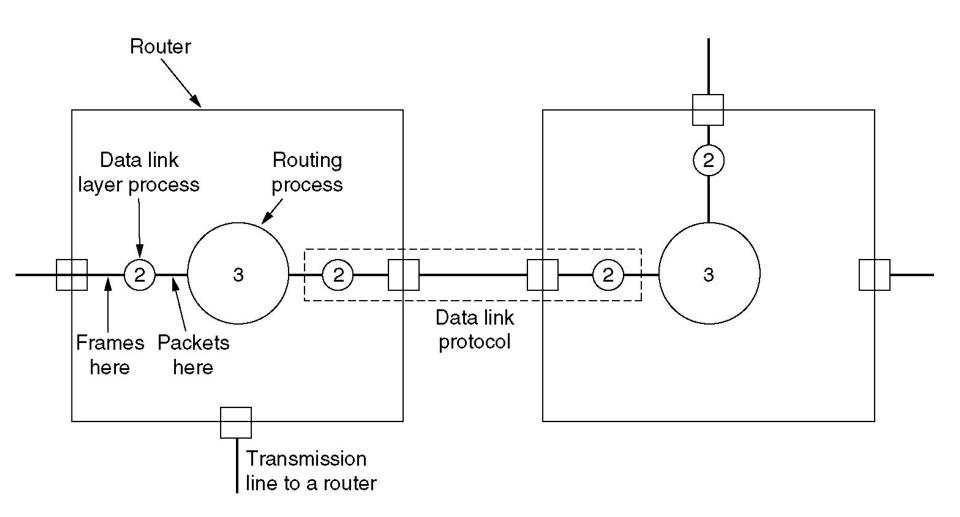
(a) Virtual communication

(b) Actual communication

Layer 3 Packets and Layer 2 Frames



Placement of the Data Link Protocol



Framing

To Think

- [Sender] \rightarrow ..1001101101101101101101101... \rightarrow [receiver]
 - » Where is the data? Where does the frame start and stop?

How to split this bit stream into frames (sets of bits)?

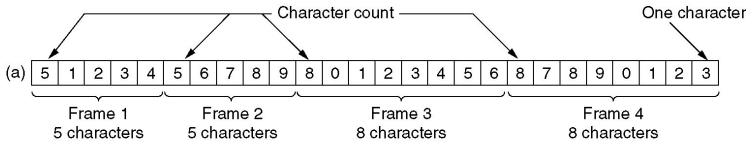
Framing

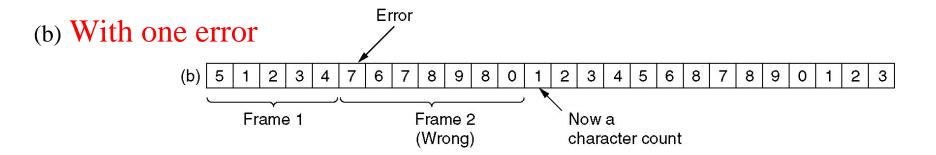
- [Sender] \rightarrow ..1001101101101101101101101... \rightarrow [receiver]
 - » Where is the data?
 - » Where does the frame start and stop?
- Three methods
 - » Character count
 - » Flag bytes with byte stuffing
 - » Start and ending flags, with bit stuffing

Framing – Character count

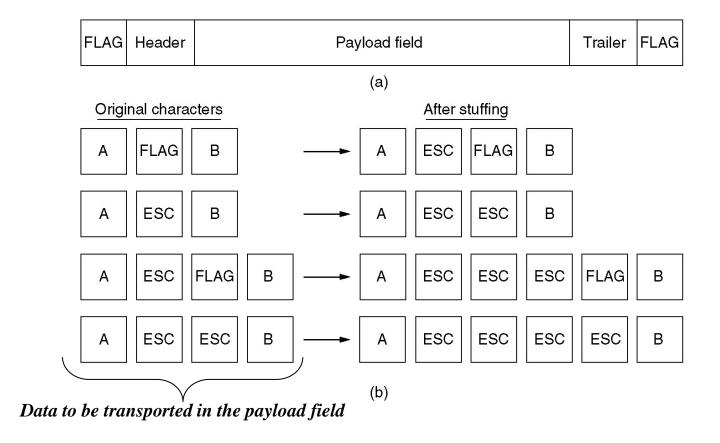
A stream of characters

(a) Without errors



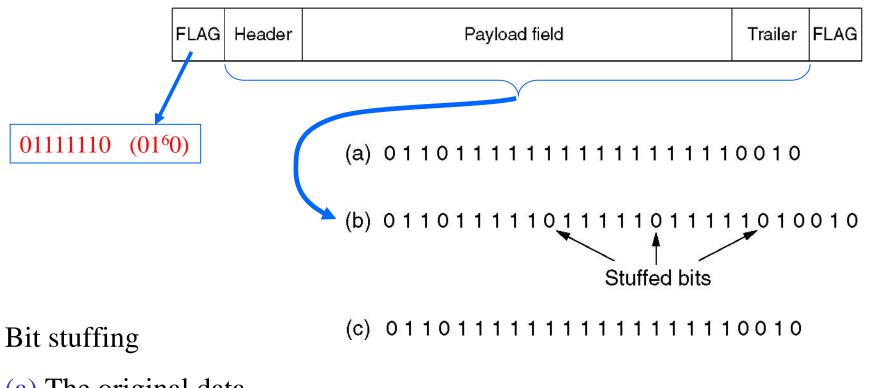


Framing - Flag bytes with byte stuffing



- (a) A frame delimited by flag bytes
- (b) Four examples of byte sequences before and after stuffing

Framing - Start and ending flags, with bit stuffing

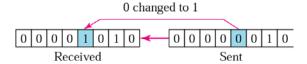


- (a) The original data
- (b) The data as it appears on the line: $1^5 \rightarrow 1^5 \bigcirc$
- (c) The data as stored in receiver's memory after destuffing: $01^{5}0 \rightarrow 01^{5}$

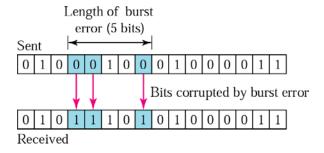
Error detection

Types of Errors

- Simple Error
 - » Random and independent from previous error



- Errors in burst
 - » Not independent; affect neighbour bits
 - » Burst lenght defined by the first and last bits in error



To Think

- Assume
 - p bit error probability (or Bit Error Ratio BER)
 - n frame length
 - Independent errors
 - FER: Frame Error Ratio
- Student A explains to Student B

why P[frame has no errors]= $(1-p)^n$

Student B explains to Student A

why P[frame has errors]= $1-(1-p)^n \Leftrightarrow FER = 1-(1-BER)^n$

Counting Errors

Assume

- p bit error probability (or Bit Error Ratio BER)
- n frame length
- Independent errors
- ♦ P[frame has no errors]= $(1-p)^n$ the n bits are good!
- ♦ P[frame has errors]= $1-(1-p)^n$ P[frame has errors]= Frame Error Ratio – FER
- P[1 bit received in error]= $\binom{n}{1}p(1-p)^{n-1}$
- P[i bits received in error]= $\binom{n}{i}p^i(1-p)^{n-i}$

```
p=10<sup>-7</sup> (good wired channel)

n=10<sup>4</sup> (~ Ethernet frame length)

P[FER]=1-(1-10<sup>-7</sup>)<sup>10<sup>4</sup></sup> \approx 10^{-3}
```

p=10⁻³ (wireless channel)
n=10⁴ (~ Ethernet frame length)
P[FER]=
$$1 - (1 - 10^{-3})^{10^4} \approx 1$$

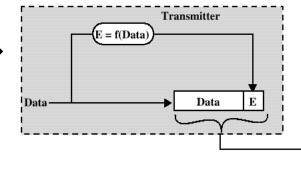
$$FER = 1 - (1 - BER)^n$$

Error Techniques

- Error techniques required!
 - » Detection (and correction)
- Effectiveness of error detection technique (code) characterized by
 - » Minimum distance of code: d
 - min number of bit errors undetected in a block of n bits
 - if fewer than d errors occur, errors are detected
 - » Burst detecting ability: B
 - max burst length of errors detected

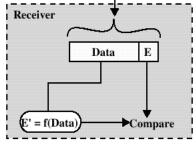
Error Detection Techniques

- Used by the receiver to determine if a packet contains errors
 - » If a packet is found to contain errors,
 the receiver may request the transmitter to re-send the packet
- ◆ Introducing redundancy →
 - » k → k+rk: data bits; r: redundancy bits



E, E' = error detecting codes f = error detecting code function

- Error detection techniques
 - » Parity check
 - » Cyclic Redundancy Check (CRC)
 - » ...



Simple Parity Check

- One parity bit added to every k information bits so that

 - » The total number of bit 1 is odd \rightarrow odd parity
- Detection of
 - » simple errors
 - » any number of odd errors in a block of k+1 bits
- Undetected
 - Even number of errors in a block
 n=k+1, block size
 p: bit error probability

$$P(un \det ected) = \sum_{i \text{ even}} \binom{n}{i} p^{i} (1-p)^{n-i}$$

Used in Character Oriented protocols

Bi-dimensional Parity

- Blocks represented in rows
 - » Parity bit per row; parity bit per column

Minimum code distance d=4

Any four errors in a rectangular configuration becomes undetectable

Internet Checksum

- ◆ The Internet (not layer 2) uses a checksum
 - » easily implementable in software
 - » 1's complement sum of 16 bit words
 - » Performance: d=2

- One's complement sum
 - » Mod-2 addition with carry-out
 - » Carry-out in the most-significant-bit is added to the least-significant bit
 - » Get one's complement of "one's complement sum"

```
1010011

0110110

carry-out ① 0001001

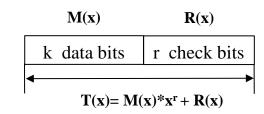
Carry wrap-around 0000001

0001010

One's complement = 1110101
```

Cyclic Redundancy Check

- Bit string represented as a polynomium
 - \rightarrow 110011 \rightarrow $x^5 + x^4 + x + 1$
- Module 2 operations
 - » Additions and subtractions identical to exclusive OR
 - » no carry, no borrow
- M(x); R(x); $T(x)=M(x) * x^r + R(x)$



- \bullet How to compute the check bits: R(x)?
 - » Choose a generator string G(x) of length r+1 bits
 - » Choose R(x) such that T(x) is a multiple of G(x): T(x)=A*G(x); R(x)=0
- $T(x) = M(x)x^r + R(x) = A*G(x)$

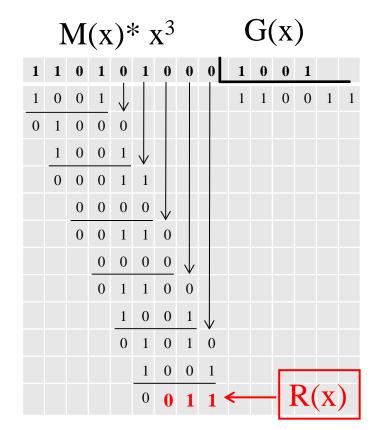
$$M(x)x^r = A*G(x) + R(x) \pmod{2}$$
 arithmetic

- $ightharpoonup R(x) = \text{remainder of } M(x)x^r / G(x)$
- Choice of G(x) is very important! ($G(x)=x^r+...+1$)

CRC - Generating R(x)

- $r=3, x^r=x^3; G(x)=x^3+1 (1001)$
- $M(x)=x^5+x^4+x^2+1$ (110101)
- $M(x)* x^3 = x^8 + x^7 + x^5 + x^3$ (110101000)
- R(x)= remainder of $M(x)x^r / G(x)$

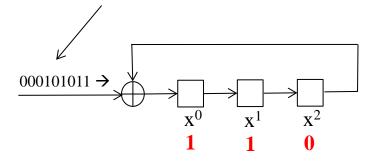
• R(x)=x+1 (011)



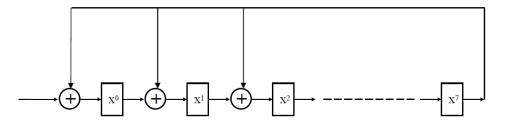
- Sent word
 - » $T(x)=M(x) * x^r + R(x) = x^8 + x^7 + x^5 + x^3 + x + 1 = 110101 011$

CRC - Generating R(x) with a Shift Register

- \bullet R(x) easily generated in hardware
- $G(x)=x^3+1$
 - $M(x) * x^3 = x^8 + x^7 + x^5 + x^3$ (110101000)
 - R(x)=x+1 (011)



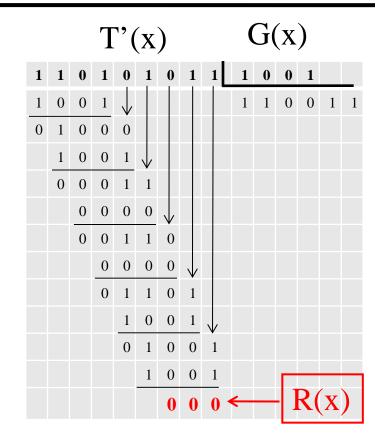
• $G(x)=x^8+x^2+x+1$



CRC – Checking at the Receiver

- Let T'(x) be the received word
 - » $T'(x) = x^8 + x^7 + x^5 + x^3 + x + 1$ (110101 011)

- Divide T'(x) by G(x)
 - » If remainder R(x)=0 \rightarrow no errors
 - » If remainder R(x) !=0
 - → errors have occurred



CRC - Performance

- For r check bits per frame the following can be detected
 - » All patterns of 1, 2, or 3 errors (d > 3)
 - » All bursts of errors of r or fewer bits
 - » All errors consisting of an odd number of inverted bits
- Common polynomials
 - » ITU-16: r=16, $G(x) = x^{16} + x^{12} + x^5 + 1$ (1000100000100001)
 - » ITU-32: r=32,

$$G(x) = x^{32} + x^{26} + x^{23} + x^{22} + x^{16} + x^{12} + x^{11} + x^{10} + x^{8} + x^{7} + x^{5} + x^{4} + x^{2} + x^{1} + 1$$

Automatic Repeat reQuest (ARQ)

Automatic Repeat ReQuest (ARQ)

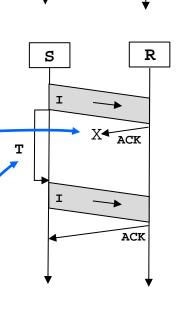
- ♦ When the receiver detects errors in a frame how to ask the sender to retransmit the frame?
- ARQ systems

Those which automatically request the retransmission of

- missing packets
- packets with errors
- ◆ Three common ARQ schemes
 - » Stop and Wait
 - » Go Back N
 - » Selective Repeat

Stop and Wait ARQ

- Sender
 - transmits Information frame I
 waits for positive confirmation ACK from receiver
- Receiver: receives I frame
 - » If I frame has no error → confirms with ACK
 - » If I frame has error → sends NACK
- Sender
 - » Receives **ACK** → proceeds and transmits new frame
 - » Receives NACK → retransmits frame I
- Problem
 - » What happens if I, ACK or NACK is lost?
 - → Timeout required!

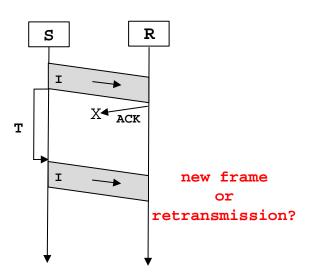


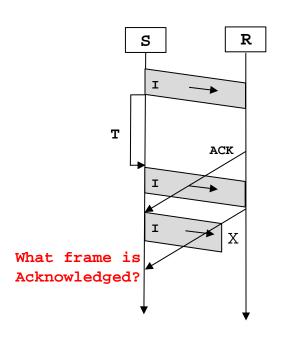
 \mathbf{R}

ACK

NACK

Stop and Wait ARQ – Sequence Numbers Required

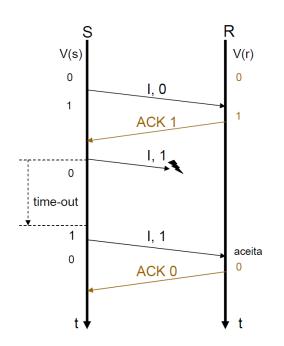


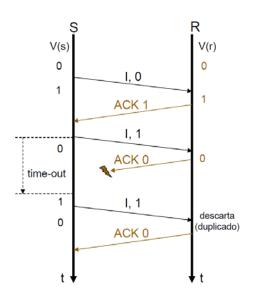


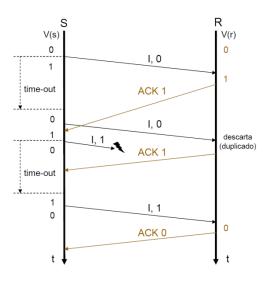
Solution

- » I frames numbered: I(0), I(1)
- » **ACK** frames numbered: **ACK(0)**, **ACK(1)**
- » ACK(i) indicates that receiver is waiting for frame I(i)
- » No **NACK** required
- » Module 2 numbers

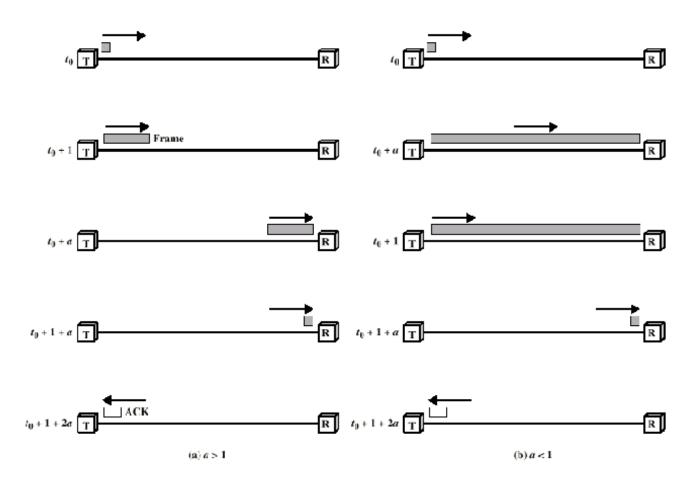
Stop and Wait ARQ – Examples







Stop and Wait - Efficiency



Stop-and-Wait Link Utilization (transmission time = 1; propagation time = a)

Stop and Wait – Efficiency Example

» WAN ATM

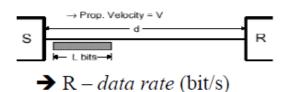
- d = 1000 km
- L = 424 bit, R = 155.52 Mbit/s
- $T_t = 2.7 \, \mu s$
- Fibra óptica \rightarrow 5 µs/km \rightarrow τ = 5 ms
- -a = 1852
- S = 1 / 3705 = 0.0003

» LAN

- d = 0.1 \sim 10 km
- L = 1000 bit, R = 10 Mbit/s
- $T_t = 100 \mu s$
- Cabo coaxial \rightarrow 4 µs/km \rightarrow $\tau = 0.4 \sim 40 \mu s$
- $-a = 0.004 \sim 0.4$
- S = 0.55 ~ 0.99 (e se R = 100 Mbit/s?)

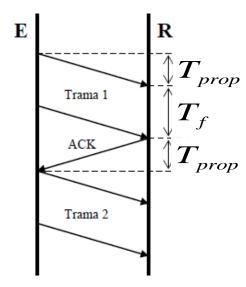
» Modem sobre linha telefónica

- d = 1000 m
- L = 1000 bit, R = 28.8 kbit/s
- $T_t = 34.7 \text{ ms}$
- UTP \rightarrow 5 μ s/km $\rightarrow \tau$ = 5 μ s
- $-a = 1.44*10^{-4}$
- S $\simeq 1.0$



$$T_{prop} = \tau = \frac{d}{V}$$

 $T_f = \frac{L}{R}$



$$a = rac{T_{prop}}{T_f}$$

$$S = \frac{T_f}{T_{prop} + T_f + T_{prop}} = \frac{1}{1 + 2a}$$

Stop and Wait ARQ – Efficiency with Errors

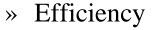
- » p_e frame error probability
- \rightarrow P[A=k]
 - probability of k Attempts required to transmit a frame with success

$$P[A=k] = p_e^{k-1}(1-p_e)$$

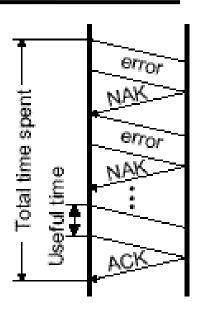


expected number of Attempts to transmit a frame with success

$$E[A] = \sum_{k=1}^{+\infty} k * P[A = k] = \frac{1}{1 - p_a}$$



$$S = \frac{T_f}{E[A](T_f + 2T_{prop})} = \frac{1}{E[A](1 + 2a)} = \frac{1 - p_e}{1 + 2a}$$



To Think

• Assume Sender and Receiver are separated by a large distance? How to improve the Efficiency of the STOP&Wait ARQ?

Go Back NARQ (Sliding Window)

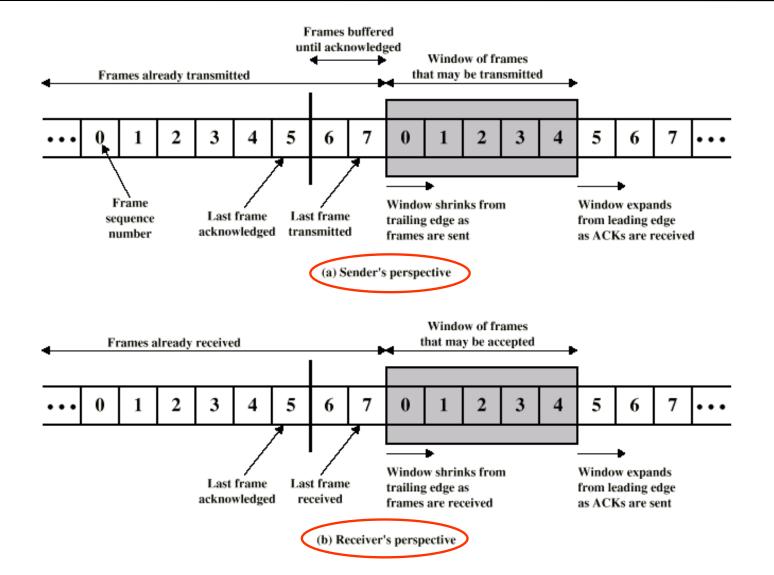
Stop and Wait

- » inefficient when $T_{prop} > T_f$ (a>1)
- » sends only one frame per Round-Trip Time (RTT= $2*T_{prop}+T_f$)

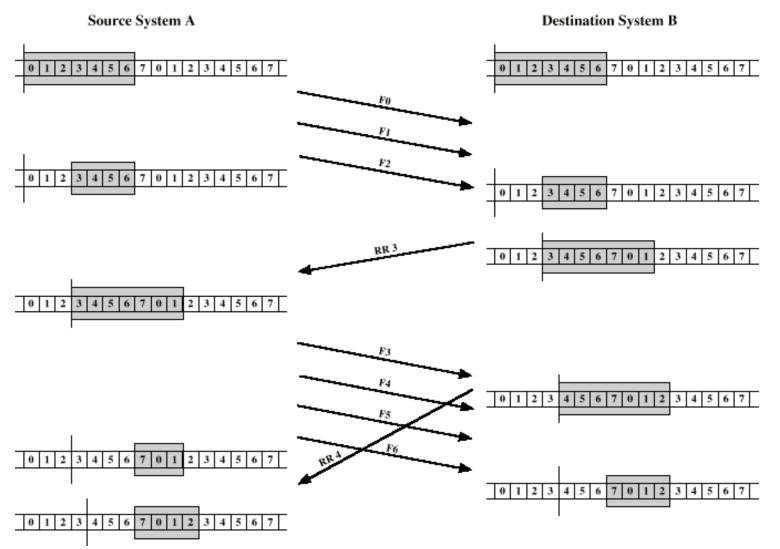
Go Back N

- » allows transmission of new packets before earlier ones are acknowledged
- » uses a Sliding Window mechanism
 - sender can send packets that are within a "window" (range) of packets
 - window advances as acknowledgements for earlier packets are received

Sliding Window - Model



Sliding Window - Example



Go Back NARQ – Basic Behaviour

Sender

- » may transmit up to W frames without receiving RR RR - Receiver Ready = ACK
- » I frames are numbered sequentially I(NS): I(0), I(1), I(2), ...
- » Cannot send I(NS=i+W) until it has received the RR(NR=i)

Receiver

- » does not accept frames out of sequence
- » sends RR(NR) to sender indicating
 - that all the packets up to NR-1 have been received in sequence
 - the sequence number, NR, of the next expected frame

Go Back N ARQ – Maximum Window, Extensions

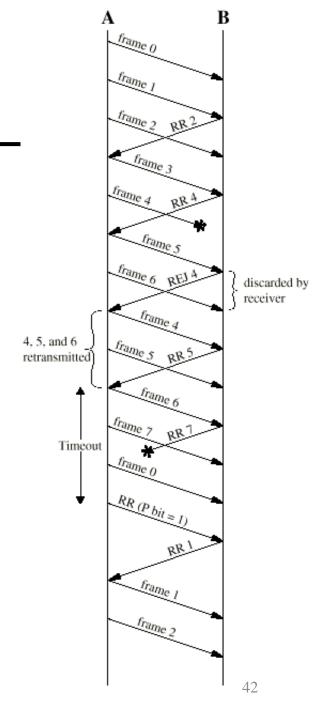
- Sequence numbers are represented module M
 - » NS, NR in {0, 1, ..., M-1}
- Maximum Window
 - $W = M-1=2^k-1$
 - » k is number of bits used to code sequence numbers

	— Header —					
F	•••	NS	NR	•••	Trailer	F

- Extensions to basic behaviour
 - » Piggybacking can be used for bidirectional flows
 - » RR information can be sent in the data packets of opposite direction

Go Back N ARQ – Behaviour under Errors

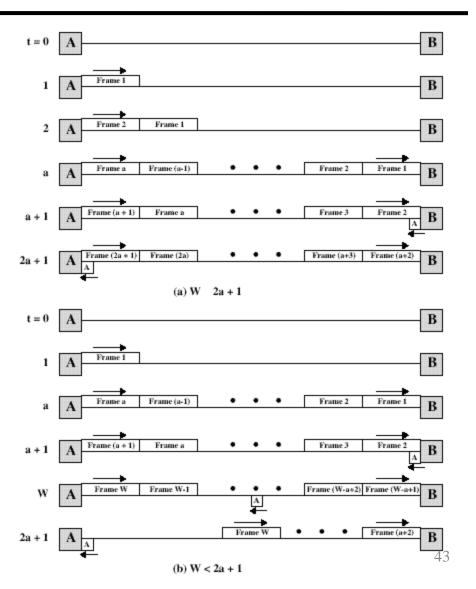
- Frame with errors
 is silently discard by the Receiver
- If Receiver receives Data frame out of sequence
 - » First out-of-sequence-frame?
 - Receiver sends REJ(NR)
 - NR indicates the next in-sequence frame expected
 - » Following out-of sequence-frames
 - Receiver discards them; no REJ sent
- When Sender receives REJ(NR=x), the Sender
 - » Goes-Back and retransmits I(x), I(x+1), ...
 - » Continues using Sliding Window mechanism
- If timeout occurs, the Sender
 - » requests the Receiver to send a RR message
 - » by sending a special message (RR command message)



Go Back N – Efficiency

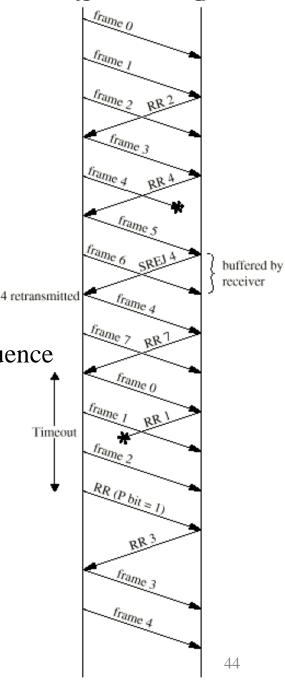
• If $W \ge 1 + 2a \implies S = 1$

• If $W < 1+2a \implies S = W/(1+2a)$



Selective Repeat ARQ

- Uses Sliding Window, but ...
- Receiver
 - » accepts out-of-sequence-frames
 - » confirms negatively, SREJ, a frame not arrived
 - » uses RR to confirm blocks of frames arrived in sequence
- Sender
 - » retransmits only the frames signaled by SREJ
- ◆ Adequate if W (a) is very large
- Maximum window size, $W = \frac{M}{2} = 2^{k-1}$



Go-Back-N and Selective Repeat ARQ – Efficiency under Errors

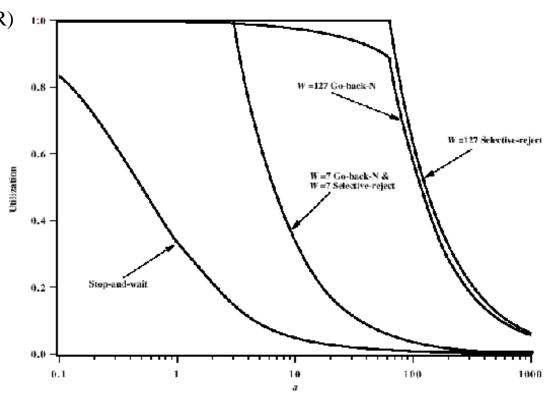
◆ Go-Back-N ARQ

p_e – frame error probability (ratio, FER) 1:0

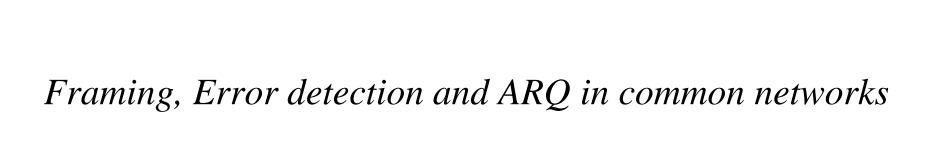
$$S = \begin{cases} \frac{1 - p_e}{1 + 2ap_e} & ,W \ge 1 + 2a\\ \frac{W(1 - p_e)}{(1 + 2a)(1 - p_e + Wp_e)} & ,W < 1 + 2a \end{cases}$$

♦ Selective Repeat ARQ

$$S = \begin{cases} 1 - p_e & ,W \ge 1 + 2a \\ \frac{W(1 - p_e)}{1 + 2a} & ,W < 1 + 2a \end{cases}$$



ARQ Utilization as a Function of $a (P = 10^{-3})$



Ethernet

- Framing
 - » Start of frame: preamble + SFD
 - » End of frame: end of signal transitions (Manchester code), length
- Error detection: FCS \rightarrow ITU-32, $G(x) = x^{32} + ... + 1$
- No ARQ
 - » Bit Error ratio (BER) very low
 - → Frame Error Ratio (FER) low
 - » CRC/FCS strong
 - Good detection of error frames
 - Frame detected with errors → discarded

p=10⁻⁷ (good wired channel)
n=10⁴ (~ Ethernet frame length)
P[frame has errors]=
$$1 - (1 - 10^{-7})^{10^4} \approx 10^{-3}$$

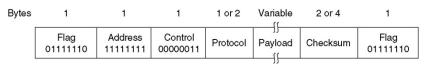
ITU-32: r=32, $G(x) = x^{32} + ... + 1$

SFD = Start of frame delimiter

DA = Destination address SA = Source address

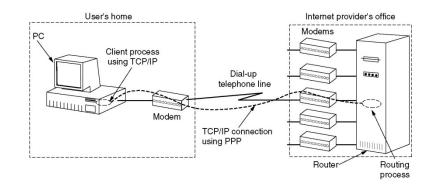
FCS = Frame check sequence

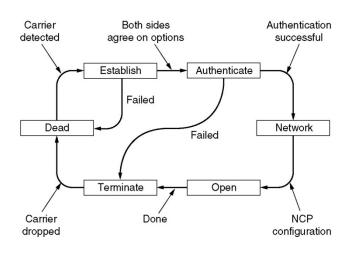
Point to Point Protocol



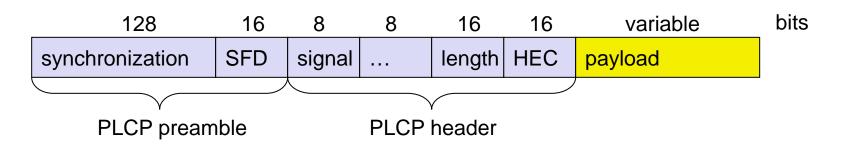
Byte stuffing

- ♦ Framing: Flags 0x7E
- Byte stuffing: ESC 0x7D
- ◆ Error detection can be negotiated
- No ARQ
- Two additional protocols
 - » LCP Link Control Protocol
 - » NCP Network Control Protocol

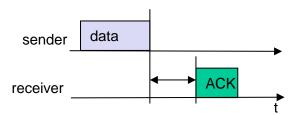




Wireless LAN



- Framing
 - » Synchronization: 0101010 ...
 - » SFD (Start Frame Delimiter \rightarrow 1111001110100000
 - » Length → Payload length in us
- HEC (Header Error Check)
 - \rightarrow ITU-16, $G(x) = x^{16} + x^{12} + x^5 + 1$
- Payload (data)
 - » Protected by strong codes
- ARQ
 - » modified version of Stop and Wait
- Signal: Payload bitrate (0A: 1 Mbit/s DBPSK; 14: 2 Mbit/s DQPSK)

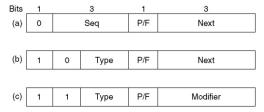


High-Level Data Link Control

HDLC, Data Link Control, bit oriented



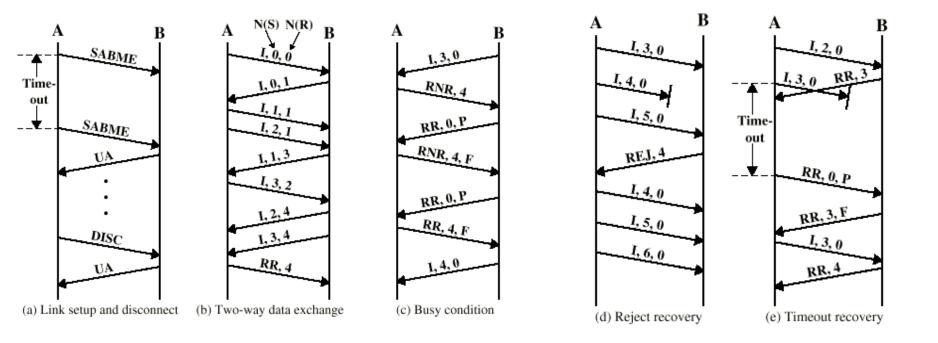
- Framing FLAGS
- Bit stuffing
- Error detection ITU-16
- Bit stuffing
- ◆ ARQ Selective Repeat ARQ
- Used as basis for many telecom networks
 - » GSM/GPRS/UMTS, Frame Relay
 - » LAP-x protocols



Control field of:

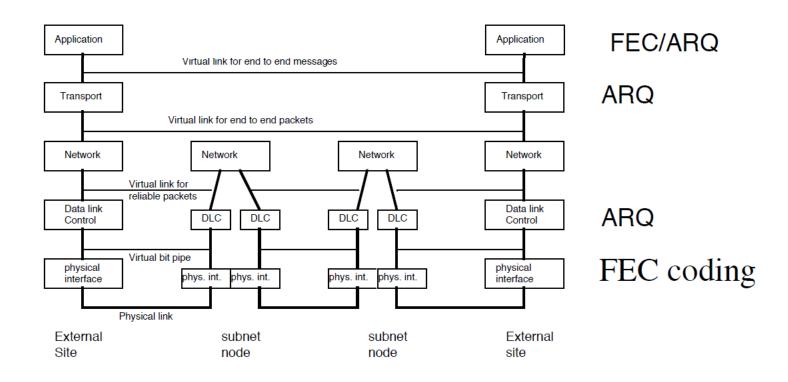
- a) An information frame.
- b) A supervisory frame.
- c) An unnumbered frame.

HDLC - Examples



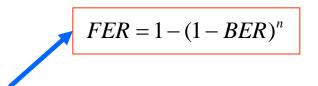
Reliability in the Protocol Stack

Reliability in the Protocol Stack



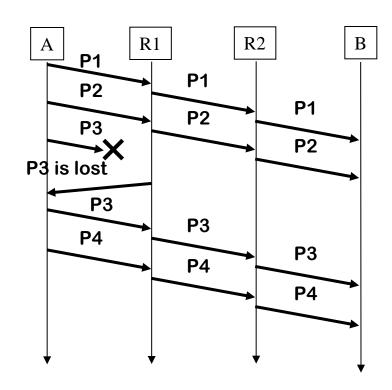
Reliability in the TCP/IP Reference Model

- ◆ The TCP/IP reference model assumes
 - » Every layer 2 offers an error free service to the upper layer
 - » Service Data Units are
 - delivered to upper layer without error,
 - or discarded

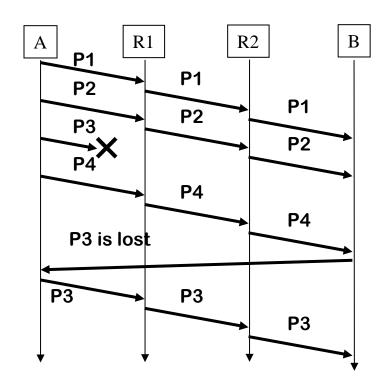


- The layered model transforms bit error in packet losses
 Therefore, packet losses must be repaired → ARQ solutions
- Two strategies can be used
 - » Link-by-Link ARQ
 - » End-to-end ARQ

Link-by-Link ARQ versus End-to-End ARQ



Link-by-Link ARQ (data link layer)

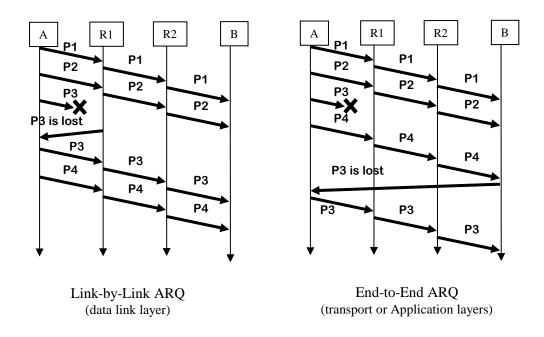


End-to-End ARQ (transport or Application layers)

To Think

Link-by-Link ARQ versus End-to-end ARQ

Compare these approaches from the points of view of delays and bitrates



Link-by-Link ARQ versus End-to-End ARQ

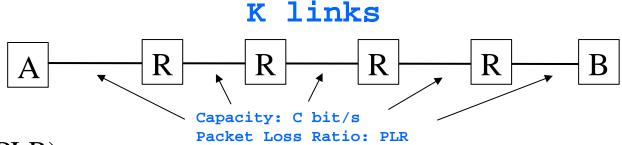
Link-by-Link ARQ

- » Repairs losses link by link
- » Requires network elements to
 - remember information about packet flows → high processing per frame/packet
 - store packets in case they have to be retransmitted → memory required

End-to-end ARQ

- » Low complexity
 - Intermediate network elements (switches, routers) become simple
- » Packets may follow different end to end paths
- » But, not acceptable when Packet Loss Ratio is high
- » Let's see why ...

End to End Capacity



- Packet Loss Ratio (PLR)
 - » Packet → layer 3; Frame → layer 2
 - » Let's assume PLR=FER (not considering losses in queues)
- ◆ Capacity of one link C₁=C*(1-PLR)
- End to End capacity
 - » using Link-by-Link ARQ: $C_{LL} = C_l = C^*(1-PLR)$
 - » Using End-to-End ARQ: $C_{EE} = C*(1-PLR)^{K}$
- ◆ End-to-end ARQ → <u>Inefficient when PLR is High</u>

k	PLR	$C_{\mathtt{EE}}$	$C_{\mathtt{LL}}$
10	0.05	√ 0.6*C	0.95 *C
10	0.0001	0.9990 *C	0.9999 *C

ARQ in the TCP/IP Reference Model

- The TCP/IP architecture assumes that
 - » The Data Link layer provides error-free packets to the network layer
 - » Data link layer provides a service with very low FER
 - » End-to-End ARQ is used, implemented at Transport or Application layers
- In the TCP/IP reference model, packet losses are repaired
 - » At the Data Link layer on lossy channels (e.g. wireless data links)
 - » At the end systems (transport layer or application layer)

Homework

- 1. Review slides
- 2. Read from Tanenbaum
 - » Chapter 3 (5th edition)
- 3. Read from Bertsekas&Gallager
 - » Sections 2.3, 2.4
- 4. Answer questions at moodle