Error and Flow Control

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Required reading: Garcia 5.2

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Error Control (1) Forward Error Correction (FEC)

Approaches

(2) Error Detection + Automatic Retrans. Req. (ARQ)

- not enough redundant info to enable error correction
 - case (a) <u>receiver detects no errors</u>
 - an ACK packet is sent back to sender
 - case (b) <u>receiver detects errors</u>
 - no ACK sent back to sender
 - sender retransmits frame after a 'time-out'



Error Control (cont.)

Challenges of ARQ-based Error Control

- send one frame at the time, wait for ACK
 - easy to implement, but inefficient in terms of channel usage
- send <u>multiple frames at once</u>
 - better channel usage, but more complex to implement sender must keep (all) <u>sent but unACKed</u> frame(s) in a buffer, as such frame(s) may have to be retransmitted



- Flow Control set of procedures used to restrict the amount of data that sender can send while waiting for acknowledgment
 - two main strategies
 - (1) Stop-and-Wait: sender waits until it receives ACK before sending next frame
 - (2) Sliding Window: sender can send W frames before waiting for ACKs

Error + Flow Control Techniques

- (1) Stop-and-Wait ARQ
- (2) Go-Back-N ARQ
- (3) Selective Repeat ARQ

Error Detection + ARQ (error detection with retransmissions) must be combined with methods that intelligently limit the number of 'outstanding' (unACKed) frames.

Fewer unACKed frames \Rightarrow fewer packets buffered at sender and receiver.

(1) Stop-and-Wait ARQ

Stop-and-Wait ARQ

Stop-and-Wait ARQ – simplest flow and error control mechanism

- sender sends an information frame to receiver
- sender, then, stops and waits for an ACK
- if no ACK arrives within <u>time-out</u>, sender will resend the frame, and again stop and wait
 - time-out period > roundtrip time
- abnormalities (and how to fix them)
 - lost acknowledgment
 - delayed acknowledgment



frame received correctly, but ACK undergoes errors / loss after time-out period, sender resends frame receiver receives the same frame twice

- <u>frames must be numbered</u> so that receiver can recognize and discard duplicate frames
 - sequence # are included in packet header



Delayed Acknowledgment (Premature Timeout)

- ACKs can be delayed due to problems with links or network congestion
 - time-out expires early, sender resends frame
 - when delayed ACK arrives, sender assumes that given ACK is for the last frame sent
- ACKs must be numbered to prevent gaps
 in delivered packet sequence





http://www.net-seal.net/animations.php?aid=37

Stop-and-Wait Efficiency

 t₀ = basic Stop-and-Wait delay – from time when frame is transmitted into channel until time when ACK arrives back to receiver, and another frame can be sent

$$t_{0} = 2 \cdot t_{prop} + 2 \cdot t_{proc} + t_{frame} + t_{ACK} = 2 \cdot t_{prop} + 2 \cdot t_{proc} + \frac{n_{f}}{R} + \frac{n_{ACK}}{R}$$

• R_{eff} = effective transmission (data) rate:

$$R_{eff} = \frac{number \text{ of info bits delivered to destination}}{total time required to deliver info bits} = \frac{n_{f} - n_{header}}{t_{0}}$$



- η_{sw} = transmission efficiency: ratio of actual and effective transmission (data) rate ideally, $\eta_{sw} \approx 1$
 - where do we lose channel efficiency, and how can $\eta_{sw} \rightarrow 1$ be achieved ?!



- (1) $\frac{n_{header}}{n_f}$ loss in efficiency due to (need for) header
- (2) $\frac{n_{ACK}}{n_f}$ loss in efficiency due to (need for) ACKs
- (3) $2(t_{prop} + t_{proc})R$ <u>bandwidth-delay product</u>
 - max number of bits in transit at any given time
 - in Stop-and-Wait ARQ delay-bandwidth product is a measure of lost opportunity in terms of transmitted bits

Bandwidth-delay product = 2*(t_{prop} + t_{proc})*R =

= capacity of the transmission pipe from the sender to the receiver and back.







Stop-and-Wait ARQ becomes inadequate when data is fragmented into small frames, such that $n_f / R = t_{frame}$ is small relative to t_{prop} .

Example [impact of delay-bandwidth product]



Efficiency	200 km	2000 km	20000 km	200000 km
	(t _{prop} = 1 ms)	(t _{prop} = 10 ms)	(t _{prop} = 100 ms)	(t _{prop} = 1 sec)
1 Mbps	10 ³	10 ⁴	10 ⁵	10 ⁶
	88%	49%	9%	1%
1 Gbps	10 ⁶	10 ⁷	10 ⁸	10 ⁹
	1%	0.1%	0.01%	0.001%

Stop-and-Wait does NOT work well for very high speeds or long propagation delays.

Stop-and-Wait Efficiency in **Channel with** Errors

- P_f = probability that transmitted frame has errors and need to be retransmitted
 - (1-P_f) probability of successful transmission and including • $\frac{1}{1-P_{e}}$ - <u>average</u> # of (re)transmission until first correct arrival

• total delay per frame: $t_0 \cdot (average \# of retrans.) = t_0 \cdot \frac{1}{1 - P_c}$



 P_{f} increases $\Rightarrow \eta_{sw}$ decreases



Probability that i transmission are needed to deliver frame successfully (i-1 transmission in error and the ith transmission is error free):

 $P[# of trans. in error = i-1] = (1-P_f) P_f^{i-1}$

$$\begin{split} \mathsf{E}[\text{\# of transmissions in error}] &= \sum_{i=1}^{\infty} (i-1) \cdot \mathsf{P}[\mathsf{n}_{\text{trans in error}} = i-1] = \sum_{i=1}^{\infty} (i-1) \cdot (1-\mathsf{P}_{\mathsf{f}}) \mathsf{P}_{\mathsf{f}}^{i-1} = \\ &= (1-\mathsf{P}_{\mathsf{f}}) \cdot \sum_{i=1}^{\infty} (i-1) \cdot \mathsf{P}_{\mathsf{f}}^{i-1} = (1-\mathsf{P}_{\mathsf{f}}) \cdot \sum_{n=1}^{\infty} \mathsf{n} \cdot \mathsf{P}_{\mathsf{f}}^{n} = \\ &= (1-\mathsf{P}_{\mathsf{f}}) \cdot \mathsf{P}_{\mathsf{f}} \cdot \left(\sum_{n=1}^{\infty} \mathsf{n} \cdot \mathsf{P}_{\mathsf{f}}^{n-1} \right) = (1-\mathsf{P}_{\mathsf{f}}) \cdot \mathsf{P}_{\mathsf{f}} \cdot \frac{1}{(1-\mathsf{P}_{\mathsf{f}})^2} = \\ &= \frac{\mathsf{P}_{\mathsf{f}}}{1-\mathsf{P}_{\mathsf{f}}} \end{split}$$

Total average delay per frame:

$$t_0 + time - out \cdot E[\# of transmiss in error] = t_0 + time - out \cdot \frac{P_f}{1 - P_f} \approx \frac{1}{1 - P_f} t_0$$

- **Piggybacking** Stop-and-Wait discussed so far was 'unidirectional'
 - in 'bidirectional' communications, both parties send & acknowledge data, i.e. both parties implement flow control
 - piggybacking method: outstanding ACKs are placed in the header of information frames
 - piggybacking can save bandwidth since the overhead from a data frame and an ACK frame (addresses, CRC, etc) can be combined into just one frame



(2) Go-Back-N ARQ

Go-Back-N ARQ – overcomes inefficiency of Stop-and-Wait ARQ – sender continues sending enough frames to keep channel busy while waiting for ACKs

- a window of W_s outstanding frames is allowed
- m-bit sequence numbers are used for both frames and ACKs, and $W_s = 2^{m}-1$



Assume: $W_s = 4$

- 1) sender sends frames one by one
- 2) frame 3 undergoes transmission error receiver ignores frame 3 and all subsequent frames
- 3) sender eventually reaches max number of outstanding frames, and takes following action:
 - go back N=W_s frames and retransmit all frames from 3 onwards

Go-Back-N ARQ (cont.)

Sender <u>Sliding</u> Window



after ACKs for frames 0 and 1 arrive and window slides

Receiver Sliding Window



- all frames are stored in a buffer, outstanding frames are enclosed in a window
 - frames to the left of the window are already ACKed and can be purged
 - frames to the right of the window cannot be sent until the window slides over them
 - whenever a new ACK arrives, the window <u>slides</u> to include new unsent frames
 - <u>once the window gets full</u> (max # of outstanding frames is reached), <u>entire window gets resent</u>
 - the size of receiver window is always 1
 - receiver is always looking for a specific frame to arrive in a specific order
 - any frame arriving out of order is discarded and needs to be resent

The complexity of the receiver in Go-Back-N is the same as that of Stop-and-Wait!!! Only the complexity of the transmitter increases.

Go-Back-N ARQ (cont.)

Problems with Go-Back-N (Go-Back-N with Timeout)

- Go-Back-N works correctly (retransmission of damaged frames gets triggered) as long as the sender has an unlimited supply of packets that need to be transmitted
 - but, in case when packets arrive sporadically, there may not be W_s-1 subsequent transmissions ⇒ window will not be exhausted, retransmissions will not be triggered
 - this problem can be resolved by modifying Go-Back-N such that:
 - 1) set a timer for each sent frame
 - 2) resend all outstanding frames either when window gets full or when the timer of first frame expires







Note:

- ACKs number always defines the number of the next expected frame !!!
- in Go-Back-N, receiver does not have to acknowledge each frame received it can send one <u>cumulative ACK</u> for several frames

Go-Back-N ARQ (cont.)

Sequence Numbers and Window Size

- m bits allotted within a header for seq. numbers
 ⇒ 2^m possible sequence numbers
 - how big should the sender window be!?
 - W > 2^m cannot be accepted multiple frames with same seq. # in the window ⇒ ambiguous ACKs
 - W = 2^m can still cause some ambiguity see below

• W = 2^m - 1 acceptable !!!





Go-Back-N • completely efficient if W_s is large enough to keep channel busy, and if channel is error free



 in case of error-prone channel, with P_f frame loss probability, time to deliver a frame is:

•
$$t_{frame}$$
 - if 1st transmission succeeds – prob. (1-P_f)
• t_{frame} + $\frac{1}{1-P_f}$ W_s · t_{frame} - if 1st transmission does NOT succeeds – prob. P_f

average # of frame/window (re)transmission until a successful transmission

• total <u>average time</u> required to transmit a frame:

$$\mathbf{t}_{\mathsf{GBN}} = (\mathbf{1} - \mathbf{P}_{\mathsf{f}}) \cdot \mathbf{t}_{\mathsf{frame}} + \mathbf{P}_{\mathsf{f}} \cdot \left(\mathbf{t}_{\mathsf{frame}} + \frac{\mathbf{1}}{\mathbf{1} - \mathbf{P}_{\mathsf{f}}} \cdot \mathbf{W}_{\mathsf{s}} \right) = \mathbf{t}_{\mathsf{frame}} + \frac{\mathbf{P}_{\mathsf{f}}}{\mathbf{1} - \mathbf{P}_{\mathsf{f}}} \cdot \mathbf{W}_{\mathsf{s}} \cdot \mathbf{t}_{\mathsf{frame}}$$

• transmission efficiency

$$\eta_{\text{GBN}} = \frac{\frac{n_{\text{f}} - n_{\text{header}}}{t_{\text{GBN}}}}{R} = \frac{1 - \frac{n_{\text{header}}}{n_{\text{f}}}}{1 + (W_{\text{s}} - 1)P_{\text{f}}} (1 - P_{\text{f}})$$
(**)

Go-Back-N ARQ (cont.)

What is total <u>average time</u> required to transmit a frame, assuming P_f?



1st attempt successful:

$$\mathbf{t}_{\mathsf{GBN}} = \mathbf{t}_{\mathsf{frame}}$$

2nd attempt successful:

$$\mathbf{t}_{\mathsf{GBN}} = \mathbf{t}_{\mathsf{frame}} + \mathbf{W}_{\mathsf{S}} \cdot \mathbf{t}_{\mathsf{frame}}$$

average case: $t_{\text{GBN}} = t_{\text{frame}} + \text{E}[\text{\# of transmissions in error}] \cdot W_{\text{s}} \cdot t_{\text{frame}}$

E[# of transmissions in error] =
$$\frac{P_f}{1-P_f}$$

$$\mathbf{t}_{\mathsf{GBN}} = \mathbf{t}_{\mathsf{frame}} + \frac{\mathbf{P}_{\mathsf{f}}}{\mathbf{1} - \mathbf{P}_{\mathsf{f}}} \mathbf{W}_{\mathsf{S}} \cdot \mathbf{t}_{\mathsf{frame}} \qquad \Longrightarrow \qquad \eta_{\mathsf{GBN}} = \frac{\frac{\mathbf{n}_{\mathsf{f}} - \mathbf{n}_{\mathsf{header}}}{\mathbf{t}_{\mathsf{GBN}}}}{\mathbf{R}} = \frac{1 - \frac{\mathbf{n}_{\mathsf{header}}}{\mathbf{n}_{\mathsf{f}}}}{1 + (\mathbf{W}_{\mathsf{s}} - 1)\mathbf{P}_{\mathsf{f}}} (1 - \mathbf{P}_{\mathsf{f}})$$

Example [Stop-and-Wait vs. Go-Back-N]

n_f = 1250 bytes = 10000 bits n_{ACK} = n_{header} = 25 bytes = 200 bits

Compare S&W with GBN efficiency for random bit errors with $p_b = 0$, 10⁻⁶, 10⁻⁵, 10⁻⁴ and bandwidth-delay product R*2*(t_{prop} + t_{proc}) = 1 Mbps * 100 ms = 100000 bits = 10 frames \rightarrow use $W_s = 11$.

Efficiency	p _b =0	p _b =10⁻ ⁶	p _b =10⁻⁵	р _b =10 ⁻⁴
S&W	8.9%	8.8%	8.0%	3.3%
GBN	98%	88.2%	45.4%	4.9%

- Go-Back-N provides significant improvement over Stop-and-Wait for large delaybandwidth product
- Go-Back-N becomes inefficient as error rate increases

(3) Selective Repeat ARQ

Selective Repeat ARQ • Go-Back-N is NOT suitable for 'noisy links' – in case of a lost/damaged frame a whole window of frames need to be resent

- excessive retransmissions use up the bandwidth and slow down transmission
- Selective Repeat ARQ overcomes the limitations of Go-Back-N by adding 2 new features
 - (1) receiver window > 1 frame, so that out-of-order but error-free frames can be accepted
 - (2) retransmission mechanism is modified only individual frames are retransmitted
- Selective Repeat ARQ is used in TCP !!!



Selective Repeat ARQ Operation



Receiver:

- window advances whenever next in-order frame arrives
- out-of-order frames are accepted only
 if their sequence numbers satisfy

 $R_{next} < R_{frame} < R_{next} + W_{s}$

 a <u>negative ACK</u> (NAK) <u>with sequence</u> <u>number R_{next}</u> is sent whenever an out-of-sequence frame is observed

Sender:

- window advances whenever an ACK arrives
- if a timer expires, the corresponding frame is resent, and the timer is reset
- whenever a NAK arrives, R_{next} frame is resent

Window Sizes (W_S and W_R)

- Window Sizes m bits allotted within a header for sequence numbers
 - \Rightarrow 2^m possible sequence numbers
 - how big should the windows be!?
 - W_S and W_R = 2^m-1 cannot be accepted due to possible ambiguity as shown below
 - W = 2^m/2 = 2^{m-1} acceptable !!!





window size $2^{m-1} = 2$

Selective Repeat Efficiency

- completely efficient if $\rm W_s$ is large enough to keep channel busy, and if channel is error free
 - of course, sequence number space must be 2X sequence sequence number space of Go-Back-N
- in case of error-prone channel, total <u>average time</u> required to transmit a frame:

$$t_{SR} = \frac{t_{frame}}{1 - P_f} = \frac{n_f}{R \cdot (1 - P_f)}$$

• transmission efficiency

$$\eta_{SR} = \frac{R_{eff}}{R} = \frac{\frac{n_f - n_{header}}{t_{SR}}}{R} = \left(1 - \frac{n_{header}}{n_f}\right) \cdot (1 - P_f) \quad (***)$$

What is total <u>average time</u> required to transmit a frame, assuming P_f?



Stop-and-Wait vs. Go-Back-N vs. Selective Repeat ³³

Performance Comparison

• assume n_{ACK} and n_{header} are negligible relative to n_f , and $\underbrace{\frac{2(t_{prop} + t_{proc})R}{n_f}}_{l} = L = W_s - 1$ W_s is for 1 less than the number of frames currently in transit

size of the "pipe" in multiples of frames

• efficiencies of three ARQ techniques are

$$\eta_{SW} = \frac{1}{1+L} \cdot (1-P_{f})$$

$$\eta_{GBN} = \frac{1}{1+LP_{f}} (1-P_{f})$$

$$\eta_{SR} = (1-P_{f})$$

$$\eta_{SR} = (1-P_{f})$$

- for 0 < P_f < 1, Selective Repeat provides best performance
- for $\rm P_f \rightarrow 0~Go\text{-}Back\text{-}N$ as good as Selective Repeat



Delay-Bandwidth product = 10, 100