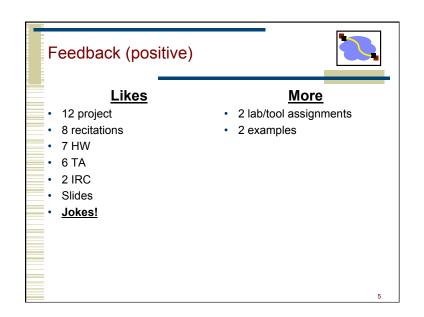


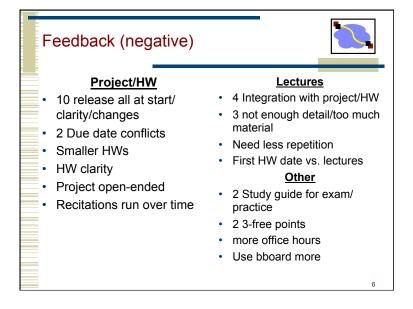
Announcements

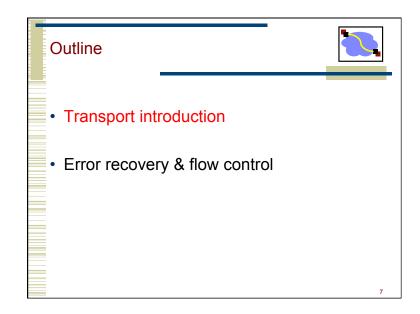


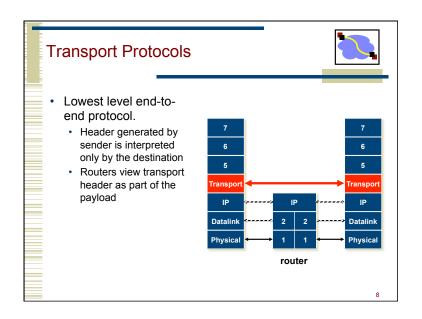
- Mid-semester grades
 - Based on (ckpt 1 & ckpt2) + midterm + HW1 + HW2
 - NOTE: GRADES DO NOT REFLECT LATE PENALTIES!
 - 25.4% of class
 - If you got a D+,D, or F → must meet with Roger or me
- Worry but don't panic
 - 74.6% of class grade remains!
 - Grading is likely a bit harsher than end-of-semester
- Project 1 grades coming soon lots of handin issues

A B C D E Perfect 15 24 9 14 17 Average 12.8 18.4 6.3 11 12.5 Crypto Bridging & Routing	What was hard					
Average 12.8 18.4 6.3 11 12.5 Crypto Bridging &		Α	В	С	D	E
Crypto Bridging &	Perfect	15	24	9	14	17
&	Average	12.8	18.4	6.3	11	12.5
				Crypto		&









Functionality Split



- · Network provides best-effort delivery
- · End-systems implement many functions
 - Reliability
 - · In-order delivery
 - Demultiplexing
 - Message boundaries
 - · Connection abstraction
 - · Congestion control

• ...

Transport Protocols



- UDP provides just integrity and demux
- TCP adds...
 - Connection-oriented
 - Reliable
 - Ordered
 - · Point-to-point
 - · Byte-stream
 - Full duplex
 - · Flow and congestion controlled

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UDP: User Datagram Protocol [RFC 768]



- "No frills," "bare bones" Internet transport protocol
- "Best effort" service, UDP segments may be:
 - Lost
 - Delivered out of order to app
- Connectionless:
 - No handshaking between UDP sender, receiver
 - Each UDP segment handled independently of others

Why is there a UDP?

- No connection establishment (which can add delay)
- Simple: no connection state at sender, receiver
- · Small header
- No congestion control: UDP can blast away as fast as desired

UDP, cont. Often used for 32 bits streaming multimedia apps Source port # Dest port # Length, in bytes of UDP Checksum Loss tolerant **→**Length segment, Rate sensitive including Other UDP uses header (why?): Application · DNS, SNMP data Reliable transfer (message) over UDP Must be at application layer **UDP** segment format Application-specific error recovery

UDP Checksum



Goal: detect "errors" (e.g., flipped bits) in transmitted segment – optional use!

Sender:

- Treat segment contents as sequence of 16-bit integers
- Checksum: addition (1's complement sum) of segment contents (and parts of IP header)
- Sender puts checksum value into UDP checksum field

Receiver:

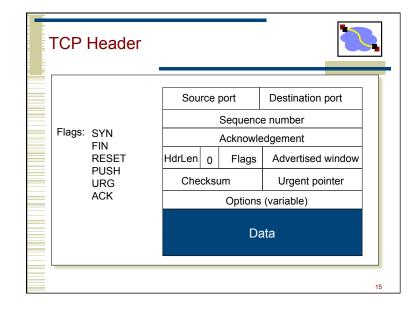
- Compute checksum of received segment
- Check if computed checksum equals checksum field value:
 - · NO error detected
 - YES no error detected But maybe errors nonethless?

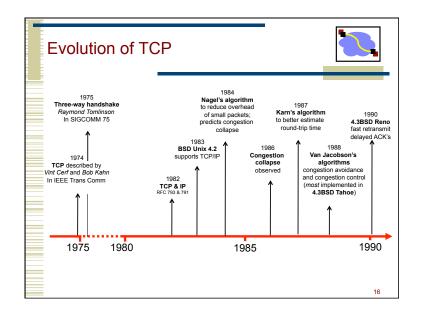
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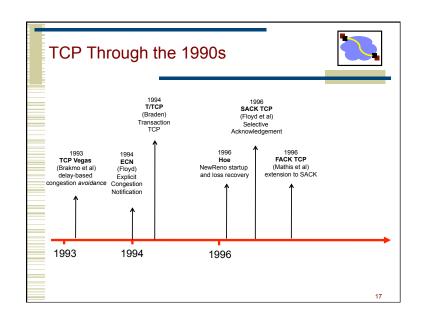
High-Level TCP Characteristics

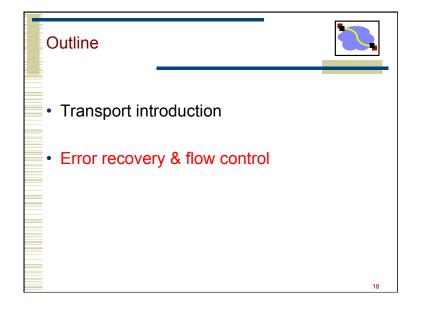


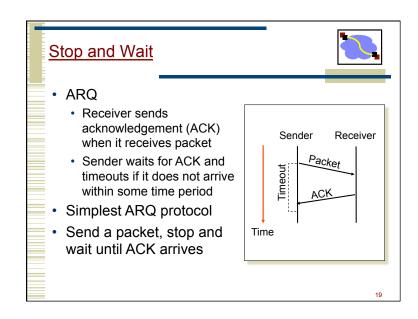
- · Protocol implemented entirely at the ends
 - Fate sharing
- Protocol has evolved over time and will continue to do so
 - · Nearly impossible to change the header
 - · Use options to add information to the header
 - · These do change sometimes
 - · Change processing at endpoints
 - · Backward compatibility is what makes it TCP

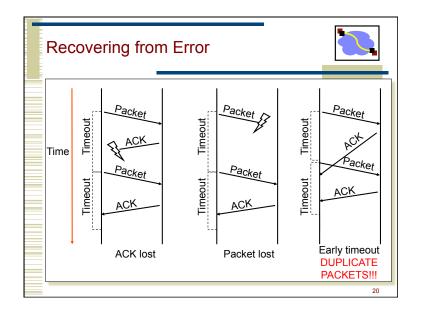












Problems with Stop and Wait



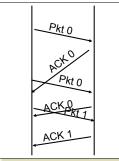
- · How to recognize a duplicate
- Performance
 - · Can only send one packet per round trip

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How to Recognize Resends?



- · Use sequence numbers
 - · both packets and acks
- Sequence # in packet is finite
 - → How big should it be?
 - For stop and wait?
- One bit won't send seq #1 until received ACK for seq #0

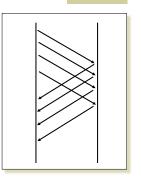


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How to Keep the Pipe Full?



- Send multiple packets without waiting for first to be acked
 - Number of pkts in flight = window
- Reliable, unordered delivery
 - · Several parallel stop & waits
 - Send new packet after each ack
 - Sender keeps list of unack'ed packets; resends after timeout
 - · Receiver same as stop & wait
- How large a window is needed?
 - Suppose 10Mbps link, 4ms delay, 500byte pkts
 - <u>1</u>? <u>10</u>? <u>20</u>?
 - Round trip delay * bandwidth = capacity of pipe

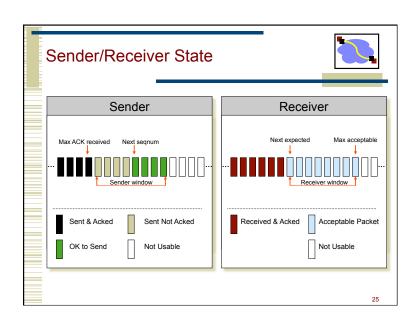


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Sliding Window



- Reliable, ordered delivery
- Receiver has to hold onto a packet until all prior packets have arrived
 - · Why might this be difficult for just parallel stop & wait?
 - Sender must prevent buffer overflow at receiver
- · Circular buffer at sender and receiver
 - Packets in transit ≤ buffer size
 - Advance when sender and receiver agree packets at beginning have been received



Sequence Numbers



- · How large do sequence numbers need to be?
 - · Must be able to detect wrap-around
 - · Depends on sender/receiver window size
- E.g.
 - Max seq = 7, send win=recv win=7
 - · If pkts 0..6 are sent succesfully and all acks lost
 - Receiver expects 7,0..5, sender retransmits old 0..6!!!
- Max sequence must be ≥ send window + recv window

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Window Sliding - Common Case



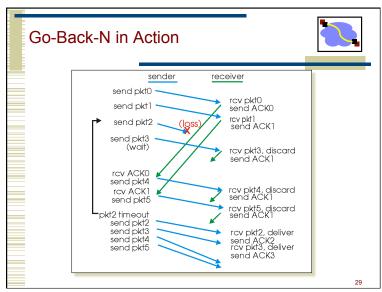
- On reception of new ACK (i.e. ACK for something that was not acked earlier)
 - · Increase sequence of max ACK received
 - · Send next packet
- On reception of new in-order data packet (next expected)
 - · Hand packet to application
 - Send cumulative ACK acknowledges reception of all packets up to sequence number
 - Increase sequence of max acceptable packet

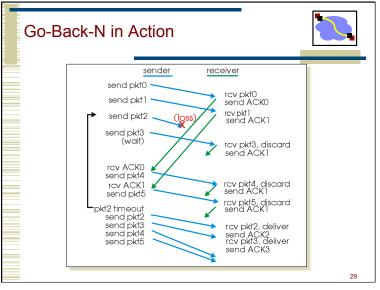
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Loss Recovery



- On reception of out-of-order packet
 - · Send nothing (wait for source to timeout)
 - Cumulative ACK (helps source identify loss)
- Timeout (Go-Back-N recovery)
 - Set timer upon transmission of packet
 - Retransmit all unacknowledged packets
- · Performance during loss recovery
 - No longer have an entire window in transit
 - · Can have much more clever loss recovery





Selective Repeat: Sender, Receiver Windows send base nextseanum already usable, not ack'ed yet sent sent, not not usable yet ack'ed window size ____ (a) sender view of sequence numbers out of order acceptable (buffered) but already ack'ed not usable Expected, not vet received window size____ rcv_base (b) receiver view of sequence numbers

Selective Repeat



- Receiver individually acknowledges all correctly received pkts
 - · Buffers packets, as needed, for eventual in-order delivery to upper layer
- Sender only resends packets for which ACK not received
 - · Sender timer for each unACKed packet
- Sender window
 - · N consecutive seg #'s
 - · Again limits seq #s of sent, unACKed packets

Important Lessons



- Transport service
 - UDP → mostly just IP service
 - TCP → congestion controlled, reliable, byte stream
- Types of ARQ protocols
 - Stop-and-wait → slow, simple
 - Go-back-n → can keep link utilized (except w/ losses)
 - Selective repeat → efficient loss recovery
- Sliding window flow control
 - · Addresses buffering issues and keeps link utilized

