COMP 431

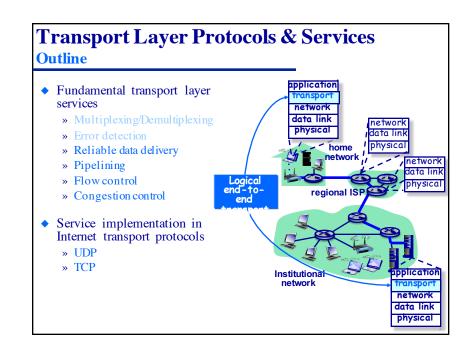
Internet Services & Protocols

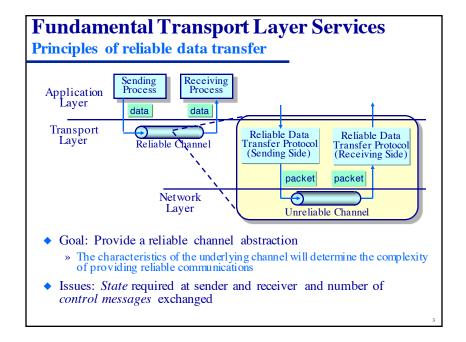
The Transport Layer

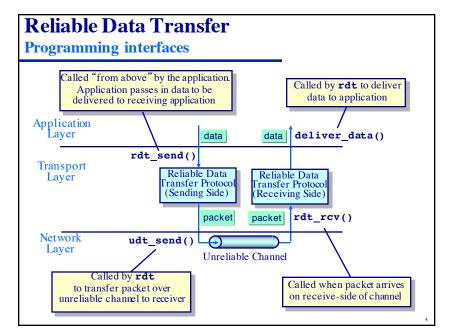
Principles of Reliable Data Delivery

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RDT reliable data transfer UDT unreliable data transfer

Reliable Data Transfer

Protocol specification method

- Use finite state machines to specify sender and receiver algorithms
 - » When in a given state, the next state (and actions) are uniquely determined by the next event

event causing state transition actions taken on state transition



Reliable Data Transfer Protocol 1.0

Reliable transfer over a reliable channel

- The underlying channel is assumed to be perfectly reliable
 - » No bit errors
 - » No loss of packets
- Sender state machine
- Receiver state machine

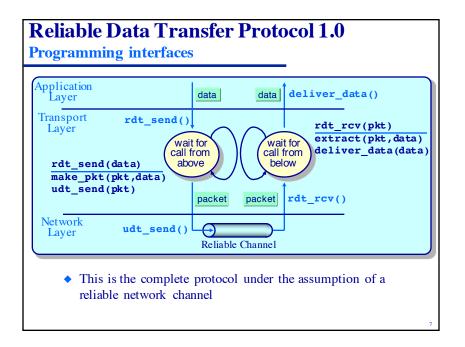


rdt_send(data) make_pkt(pkt,data) udt_send(pkt)



rdt_rcv(pkt) extract(pkt,data) deliver_data(data)

6



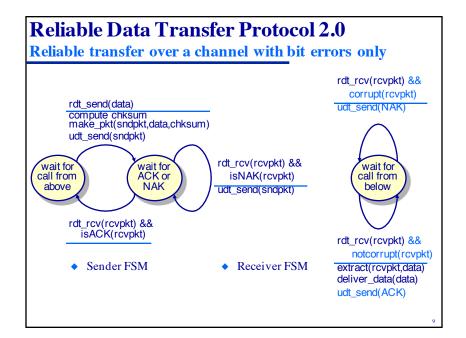
Reliable Data Transfer Protocol 2.0

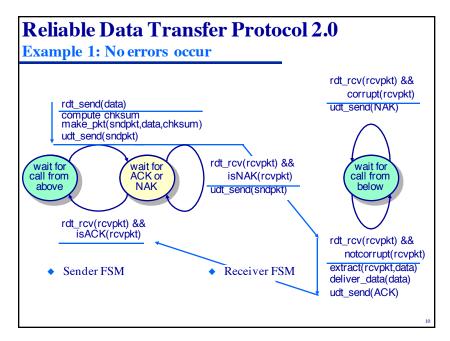
Reliable transfer over a channel with bit errors

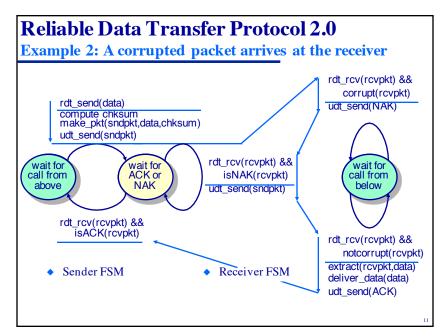
- ◆ Now assume the underlying channel may "flip" random bits in a packet
- ◆ How to detect errors?
- ◆ How to recover from errors:
 - » acknowledgements (ACKs) the receiver explicitly tells the sender that a packet was received OK
 - » negative acknowledgements (NAKs) the receiver explicitly tells the sender that a packet had errors
 - » Sender retransmits packet on receipt of NAK
- New mechanisms to deal with bit errors:
 - » Error detection
 - » Control messages (ACK, NAK) from a receiver to the sender
 - » Retransmission

Detect errors with checksum

Sender would have to store a packet until it gets an ack

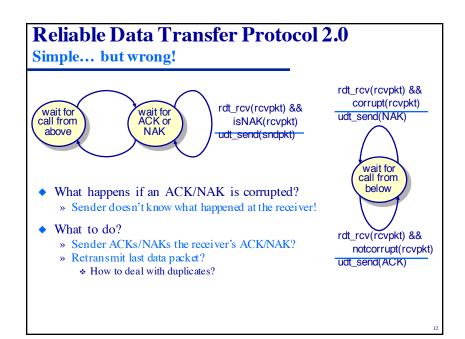






Sender and receiver both calculate checksum look up CRC(cyclic redundancy checksum) online

What if the ack is corrupted?

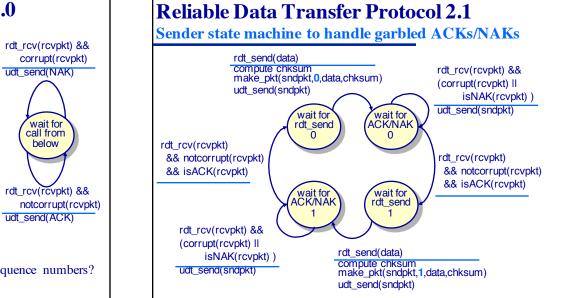


Reliable Data Transfer Protocol 2.0

Simple... but wrong!



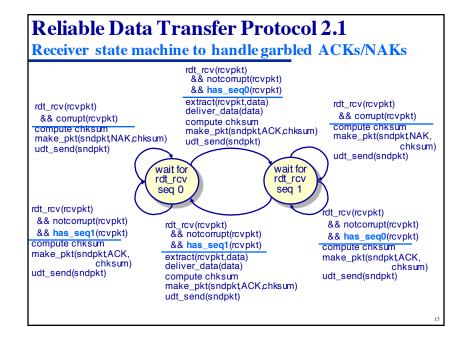
- Deal with corrupted ACKs/NAKs by retransmission of data packets
- Sender will add a sequence number to each packet to allow the receiver to detect duplicate packets
 - » Receiver's transport layer discards duplicate packets
- ♦ How much space to reserve in a header field for sequence numbers? need 2 sequence numbers, or 1 bit



Midterm

1st question: compare and contrast 2 definition

example: compare circuit switching and packet switching



Reliable Data Transfer Protocol 2.1

Discussion (Handling garbled ACKs/NAKs)

Sequence number added to header

» Two sequence numbers suffice

Must check if received ACK/NAK is corrupted

Number of states doubles

- » State encodes whether current packet has sequence number 0 or 1
- Sender issues

Must check if received packet is duplicate

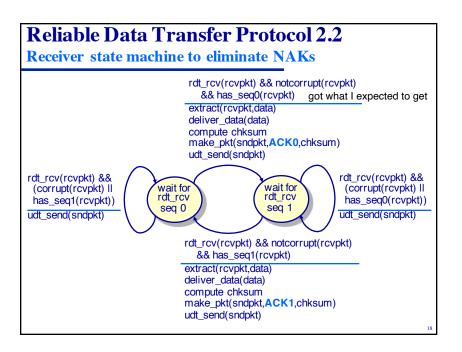
» State encodes whether expected packet sequence number is 0 or 1

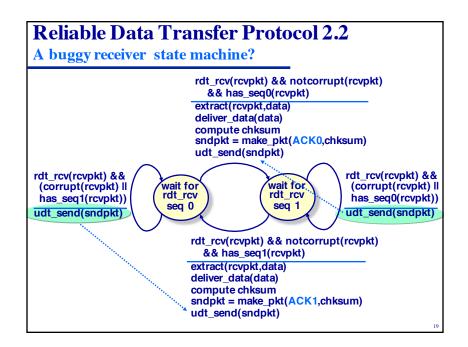
Note: receiver can *not* know if its last ACK/NAK received OK at sender

Receiver issues

16

Reliable Data Transfer Protocol 2.2 A NAK-free protocol • Instead of NAKing, receiver sends ACK for last packet rdt send(data) compute chksum received OK make_pkt(sndpkt,0,data,chksum) » Receiver must include the sequence udt_send(sndpkt) number of packet being ACKed in rdt_rcv(rcvpkt) && (corrupt(rcvpkt) || wait for isACK(rcvpkt,1)) wait for ACK0 rdt_send 0 udt_send(sndpkt) • Receipt of duplicate ACKs at sender is equivalent to a NAK rdt_rcv(rcvpkt) » Sender retransmits current packet && notcorrupt(rcvpkt) && isACK(rcvpkt,0) wait for rdt_send 1 **Sender FSM**





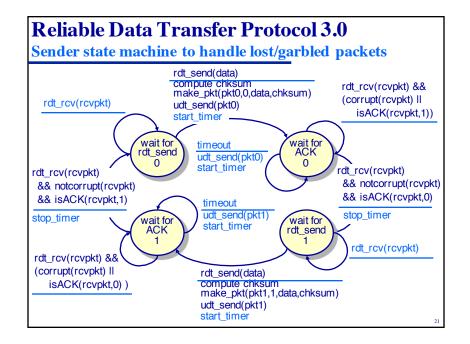
Reliable Data Transfer Protocol 3.0

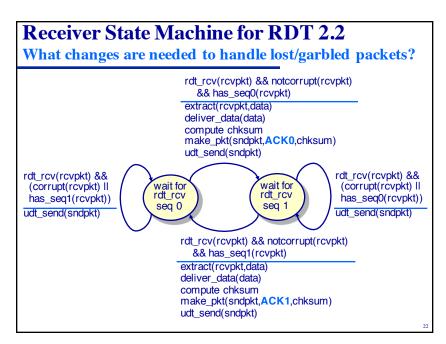
Dealing with channels with errors and loss

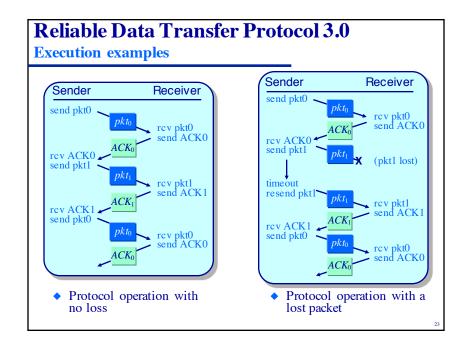
- Now assume the underlying channel can also **lose** packets
- New problem: How to detect loss?
- » Are checksums, ACKs, sequence numbers, retransmissions enough?
 add a timer for how long to wait before concluding that the ack isnt arriving
 Approach: sender waits "reasonable" amount of time and
- retransmits if no ACK received in this time
 - » Requires the use of a countdown timer
- What if packet (or ACK) just delayed beyond its timer?
 - » Retransmission will be duplicate...
 - » But use of sequence numbers already handles this!

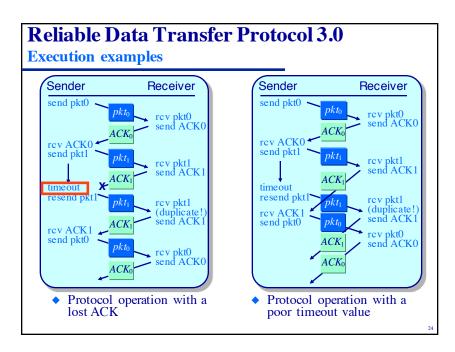
How long should you wait?

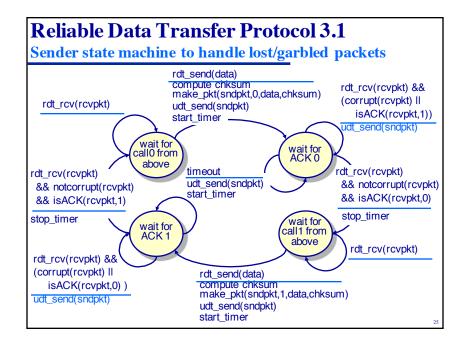
- a little longer than the typical RTT
- 3 way handshake gives a sample of the RTT (round trip time)

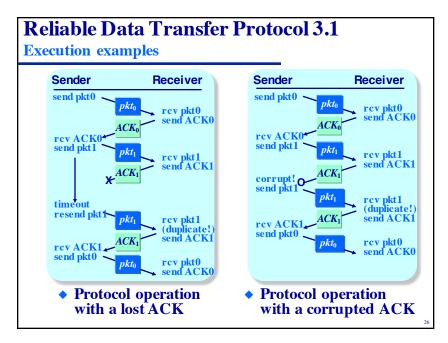


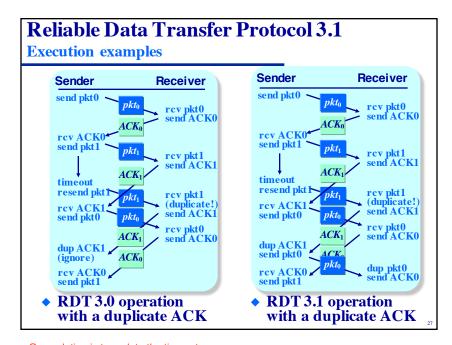












Reliable Data Delivery Summary Application Layer deliver_data() rdt_send() wait for ACK/NAK Transport Layer wait for rdt_send 0 wait for CK/NAK wait for rdt_send **Network** udt_send() rdt rcv() Layer **Unreliable Channel** • Reliable delivery requires a distributed algorithm for detecting and repairing packet errors and loss » Reliability achieved at the expense of considerable overhead and complexity

One solution is to update the timeout

adding congestion to the network but reliable data transfer is still achieved

RTT isn't measured for any packet that has to be sent again