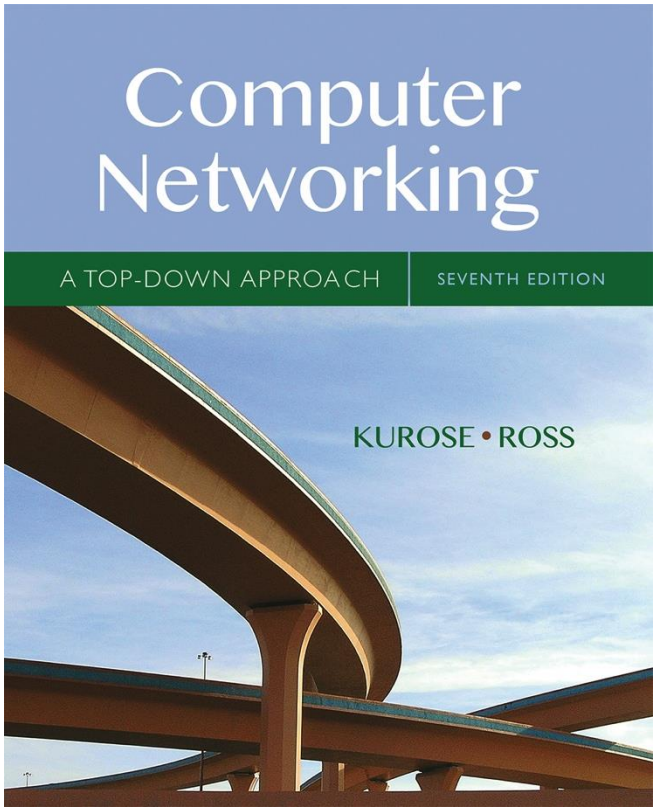
Chapter 3
Transport Layer
Part 2



Computer Networking: A Top Down Approach

7th edition
Jim Kurose, Keith Ross
Pearson/Addison Wesley
April 2016

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Thanks and enjoy! JFK/KWR

Chapter 3 outline

3.1 transport-layer services

3.2 multiplexing and demultiplexing

3.3 connectionless transport: UDP

3.4 principles of reliable data transfer

3.5 connection-oriented transport: TCP

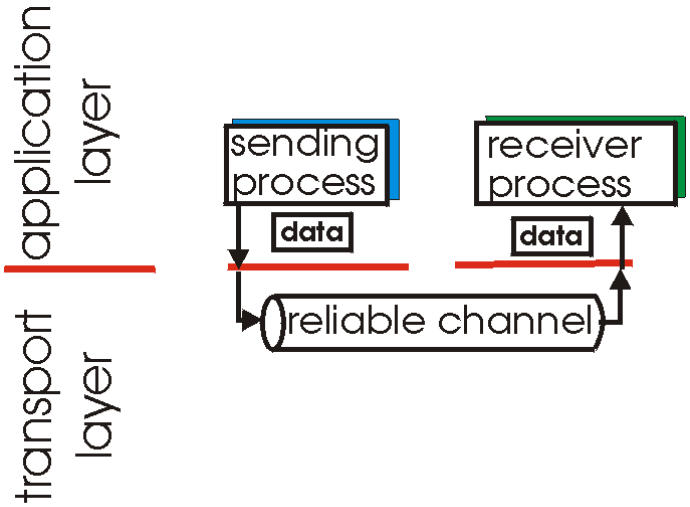
- segment structure
- reliable data transfer
- flow control
- connection management

3.6 principles of congestion control

3.7 TCP congestion control

Principles of reliable data transfer

- important in application, transport, link layers
  - top-10 list of important networking topics!

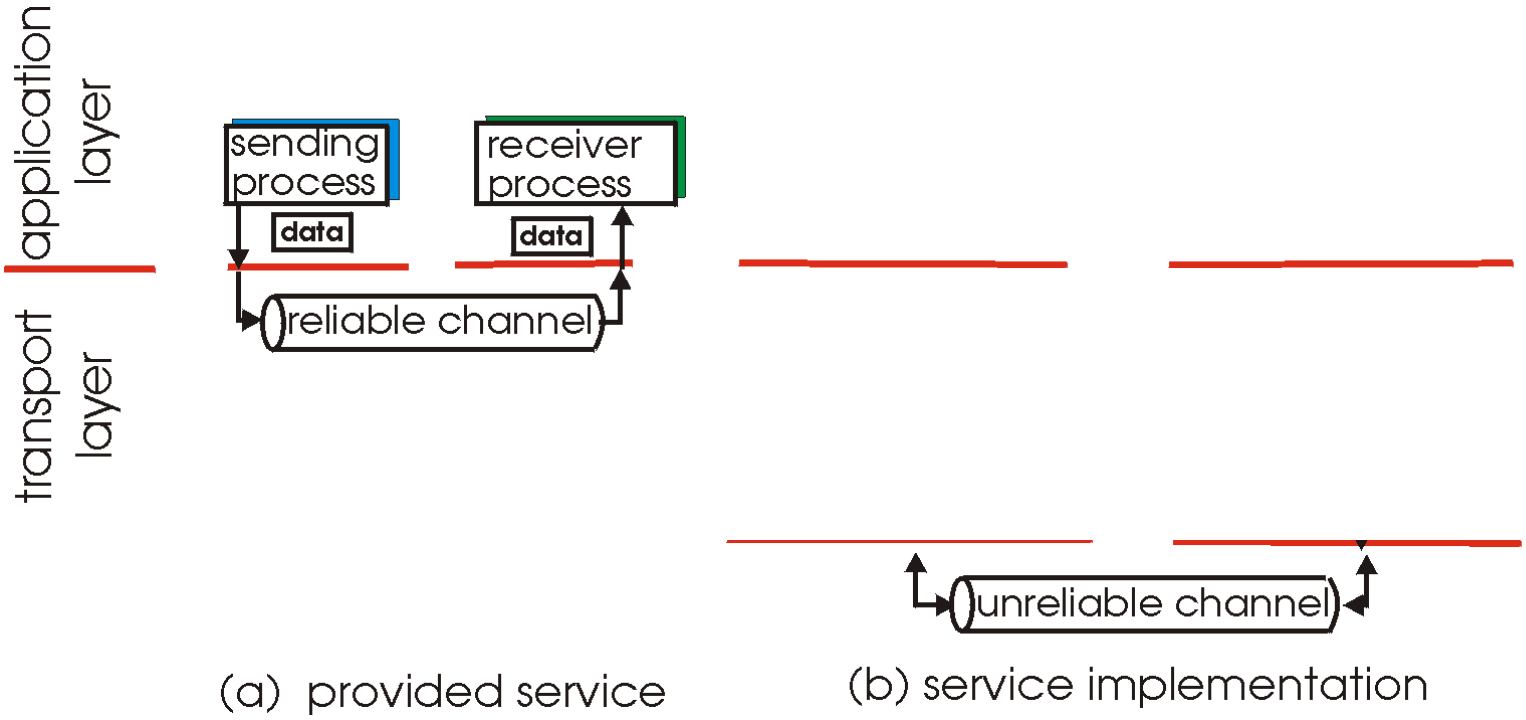


(a) provided service

- characteristics of unreliable channel will determine complexity of reliable data transfer protocol (rdt)

Principles of reliable data transfer

- important in application, transport, link layers
  - top-10 list of important networking topics!



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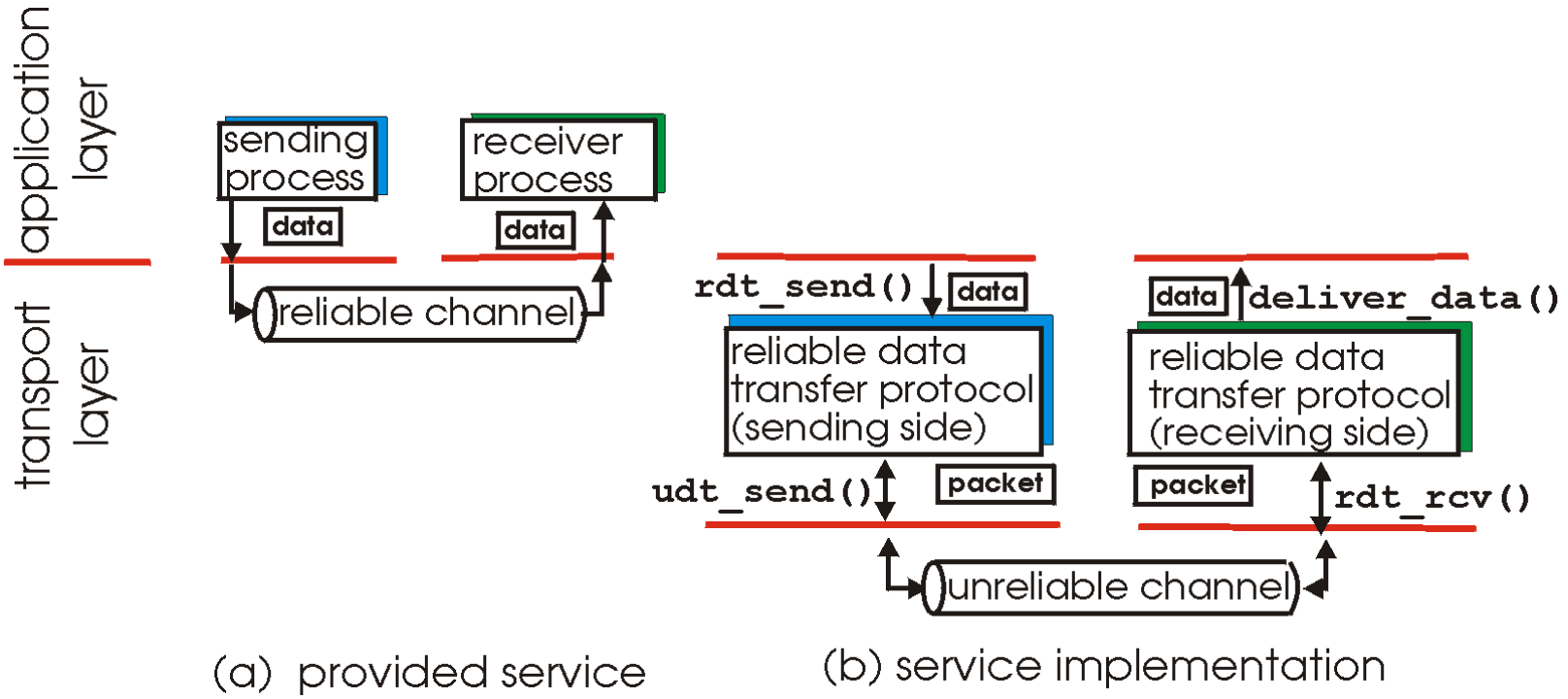
Principles of reliable data transfer

- important in application, transport, link layers
  - top-10 list of important networking topics!

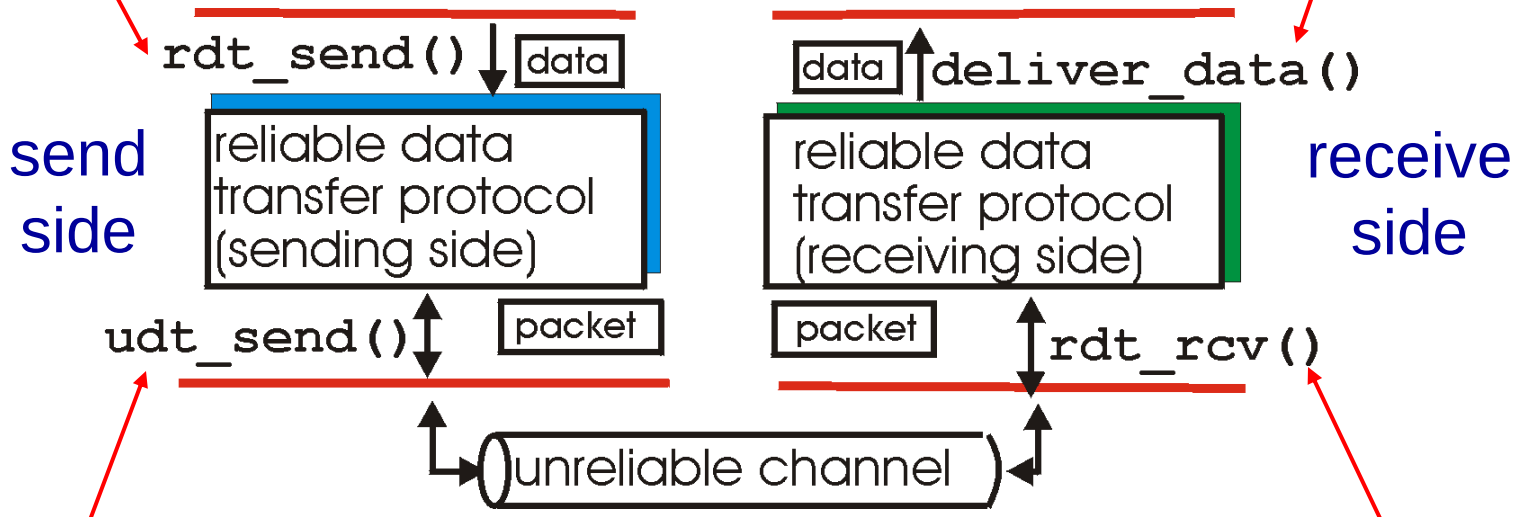


- characteristics of unreliable channel will determine complexity of reliable data transfer protocol (rdt)

Reliable data transfer: getting started

`rdt_send()`: called from above, (e.g., by app.). Passed data to deliver to receiver upper layer

`deliver_data()`: called by **rdt** to deliver data to upper



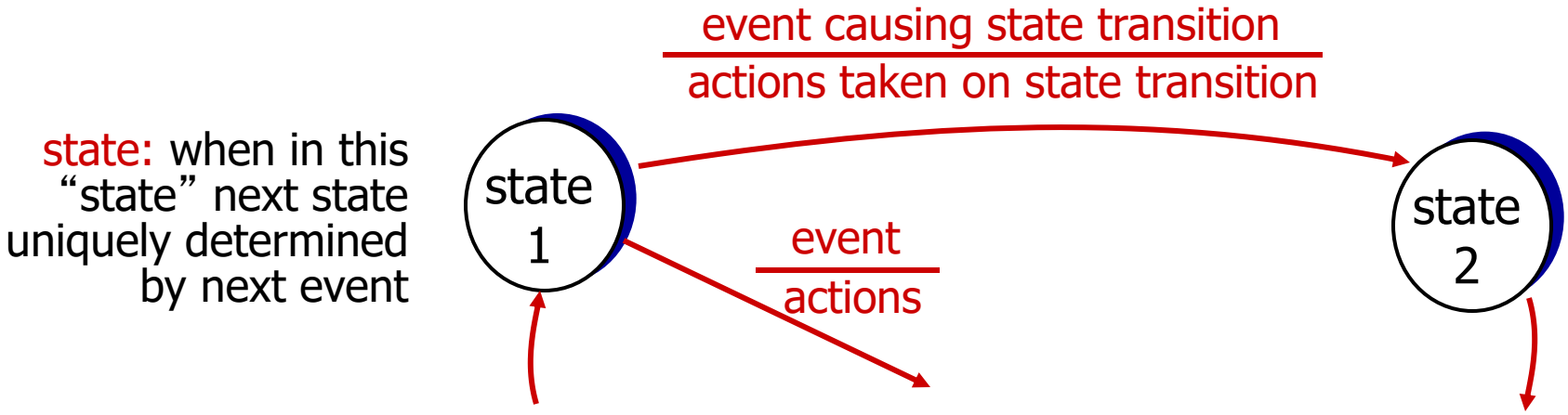
`udt_send()`: called by rdt, to transfer packet over unreliable channel to receiver

`rdt_rcv()`: called when packet arrives on rcv-side of channel

Reliable data transfer: getting started

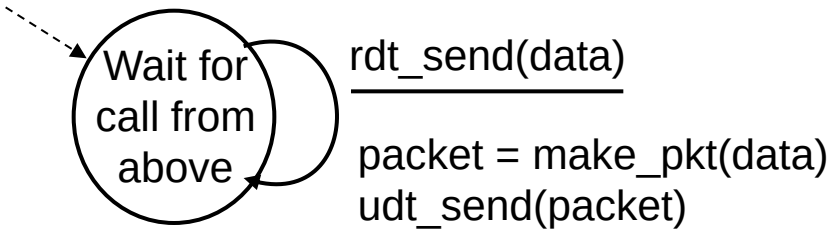
we'll:

- incrementally develop sender, receiver sides of reliable data transfer protocol (rdt)
- consider only unidirectional data transfer
  - but ***control info*** will flow on both directions!
- use finite state machines (FSM) to specify sender, receiver

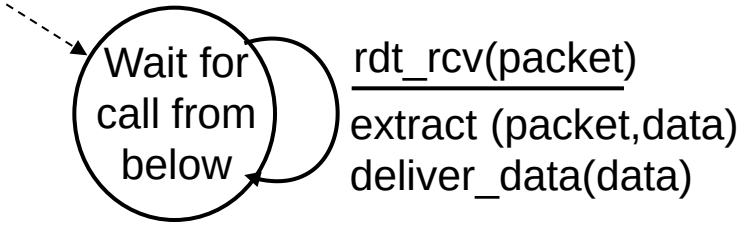


rdt1.0: reliable transfer over a reliable channel

- underlying channel perfectly reliable
  - no bit errors
  - no loss of packets
- separate FSMs for sender, receiver:
  - sender sends data into underlying channel
  - receiver reads data from underlying channel

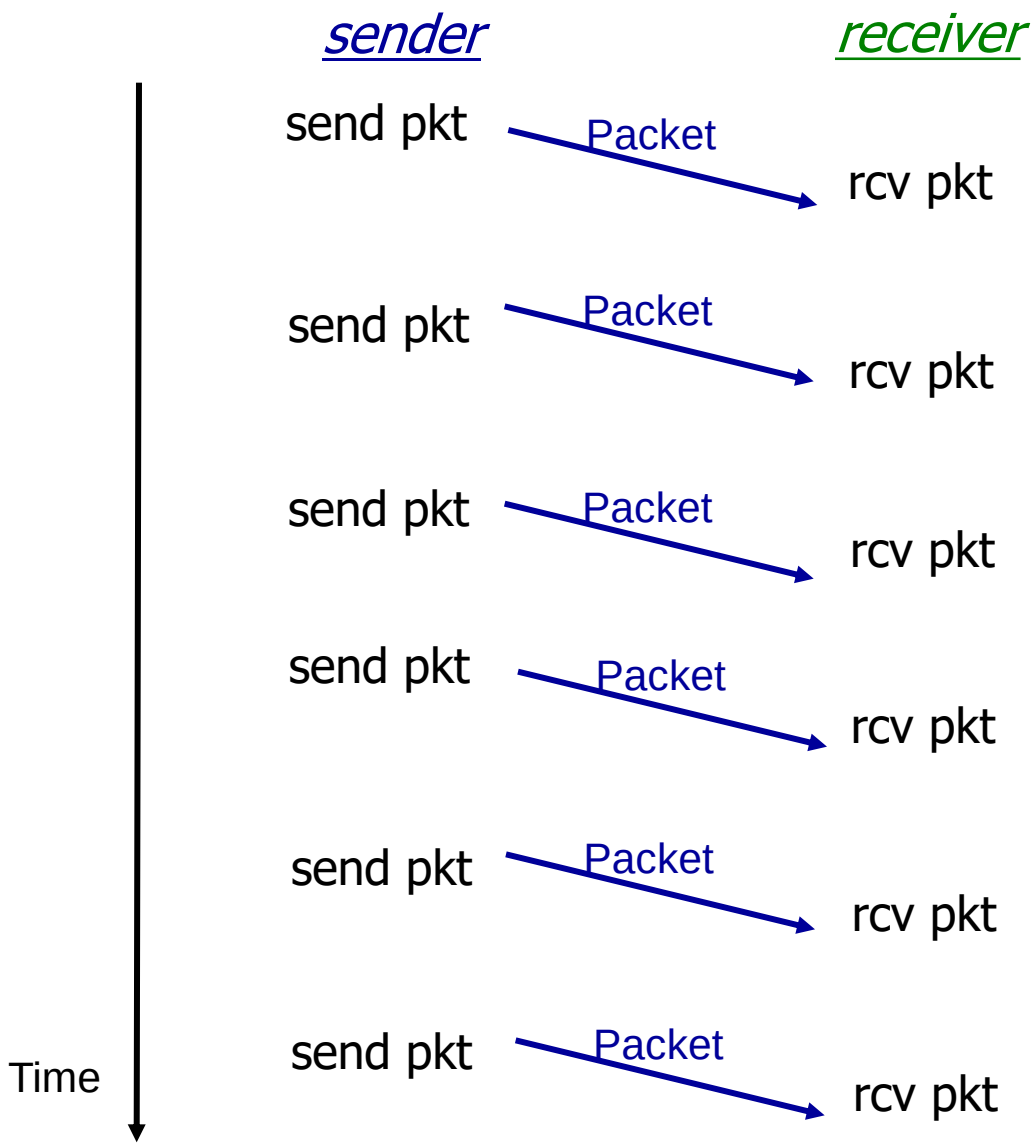


sender



receiver

rdt1.0: reliable transfer over a reliable channel



rdt2.0: channel with bit errors

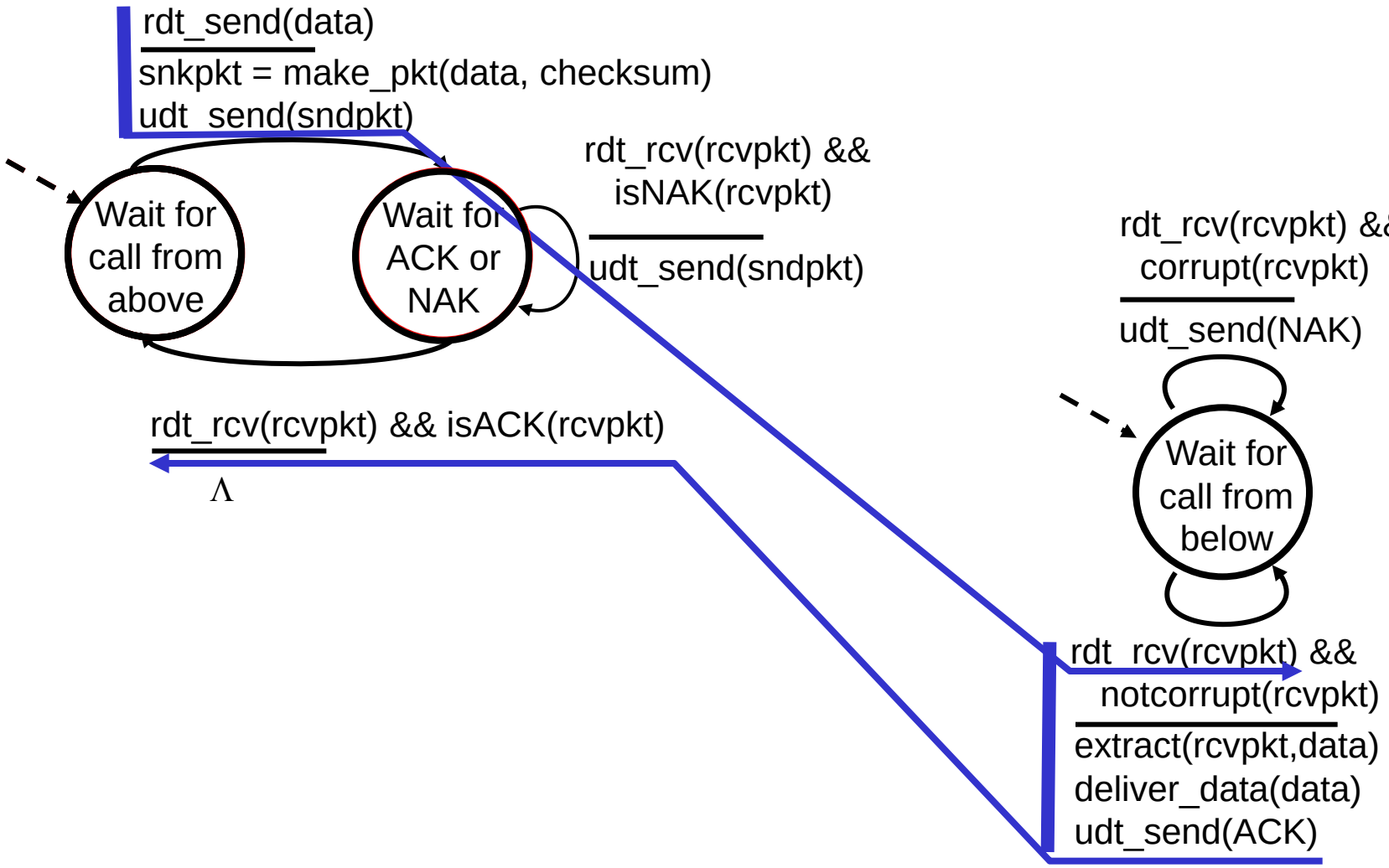
- underlying channel may flip bits in packet
  - checksum to detect bit errors
- *the* question: how to recover from errors:

How do humans recover from "errors" during conversation?

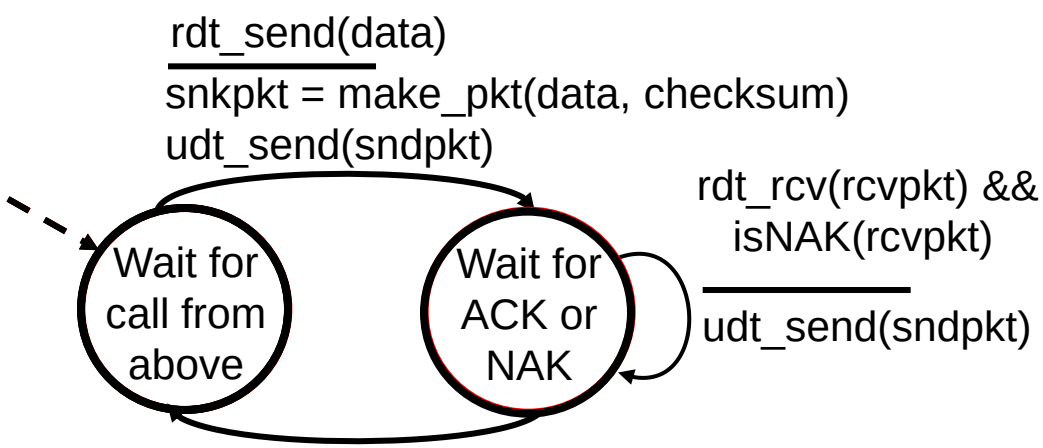
rdt2.0: channel with bit errors

- underlying channel may flip bits in packet
  - checksum to detect bit errors
- *the* question: how to recover from errors:
  - *acknowledgements (ACKs):* receiver explicitly tells sender that pkt received OK
  - *negative acknowledgements (NAKs):* receiver explicitly tells sender that pkt had errors
  - sender retransmits pkt on receipt of NAK
- new mechanisms in `rdt2.0` (beyond `rdt1.0`):
  - error detection
  - feedback: control msgs (ACK,NAK) from receiver to sender

rdt2.0: operation with no errors in data

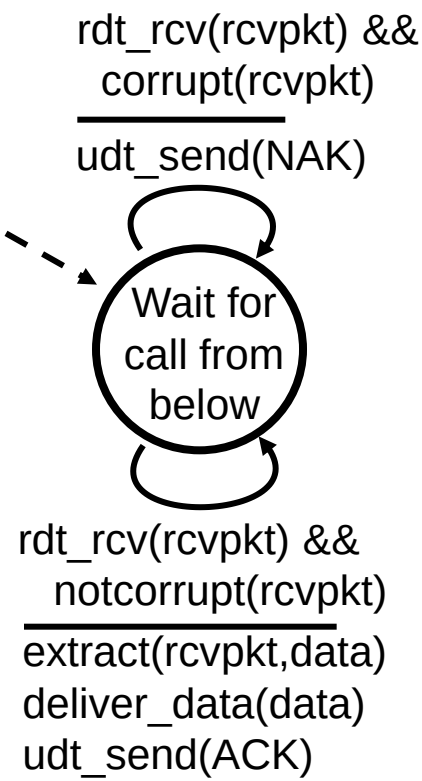


rdt2.0: error in data and no error scenarios

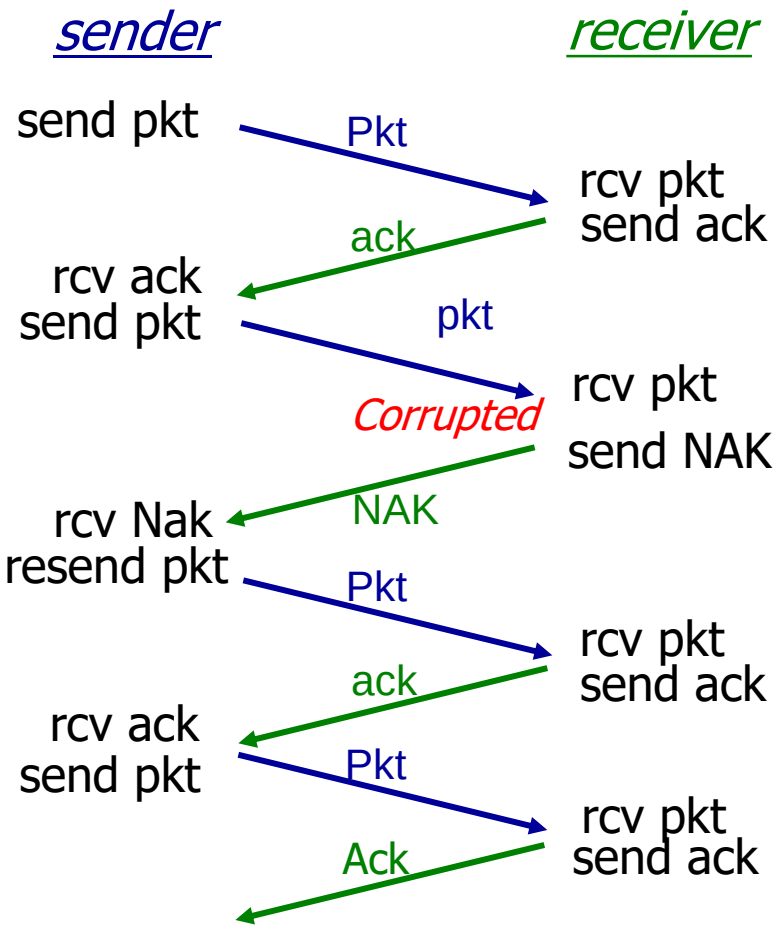


sender

receiver



rdt2.0 in action (Assuming ACK/NAK Not Corrupted)



Only Data Packet corrupted

rdt2.0 has a fatal flaw!

what happens if ACK/NAK corrupted?

- sender doesn't know what happened at receiver!
- can't just retransmit: possible duplicate

Solution for handling duplicates:

- sender retransmits current pkt if ACK/NAK corrupted
- sender adds *sequence number* to each data pkt
- receiver discards (doesn't deliver up) duplicate pkt

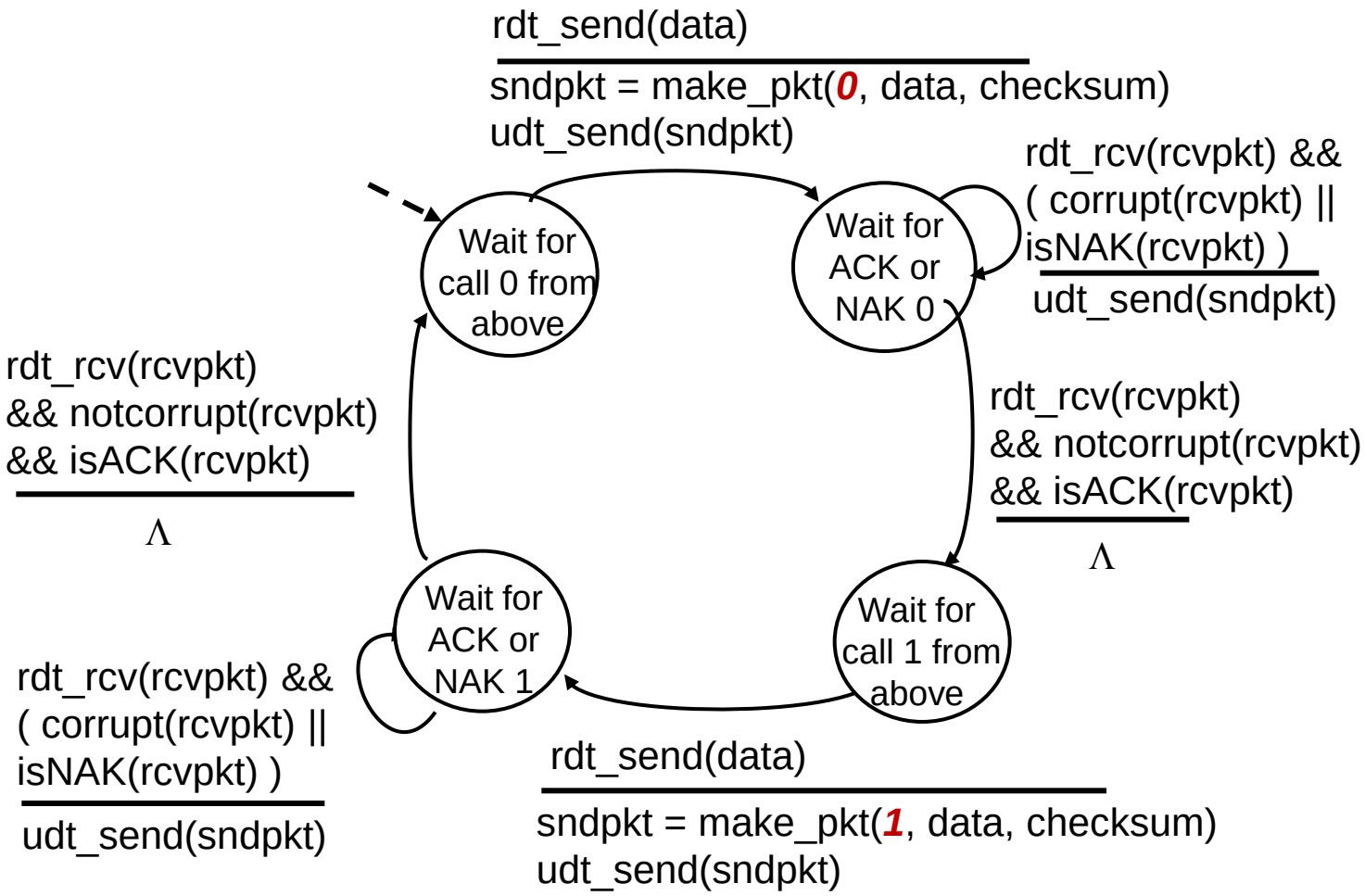
Problem: duplicate

Receiver doesn't know whether received pkt is a retransmit or a new pkt

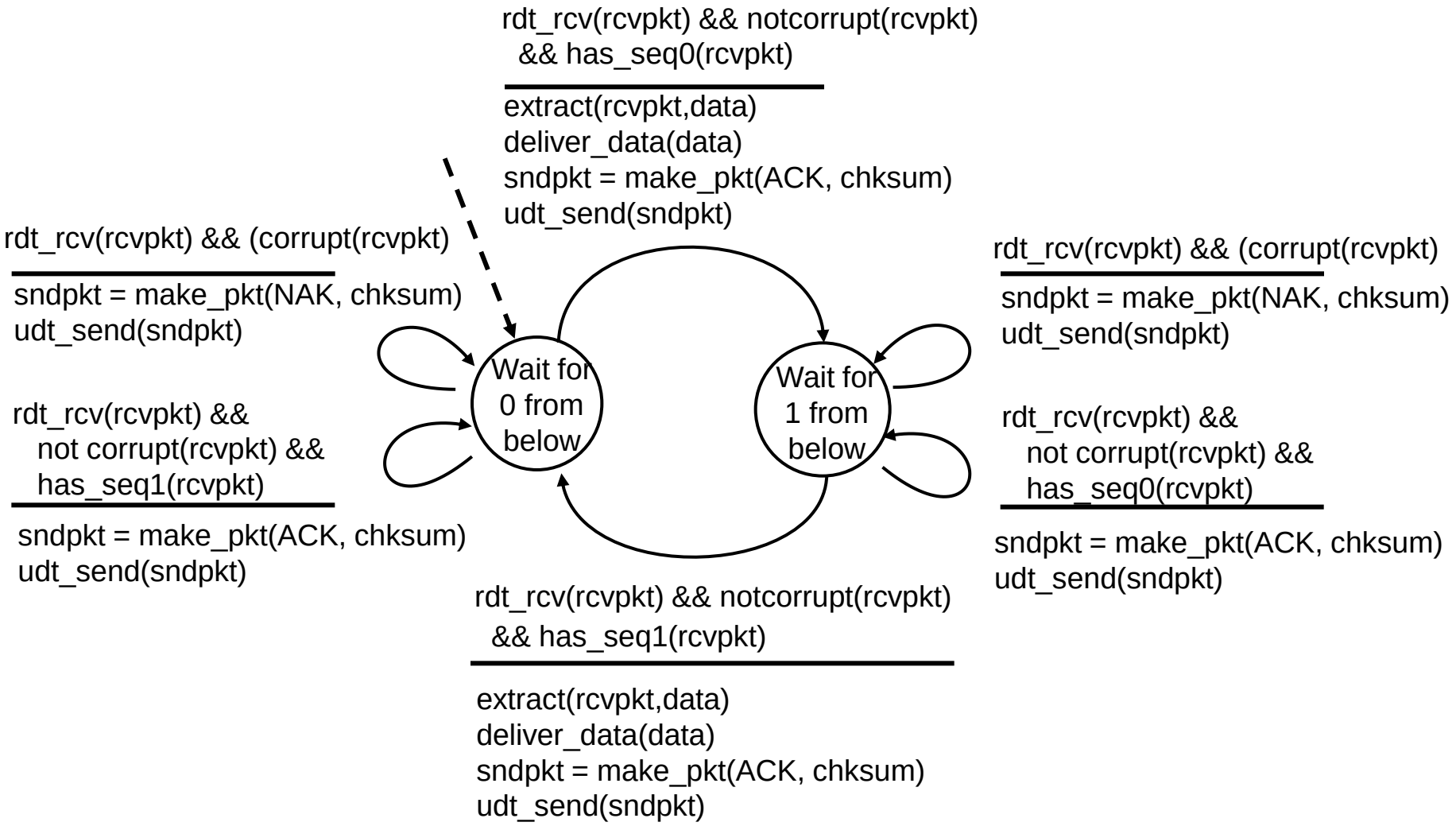
stop and wait

sender sends one packet, then waits for receiver response

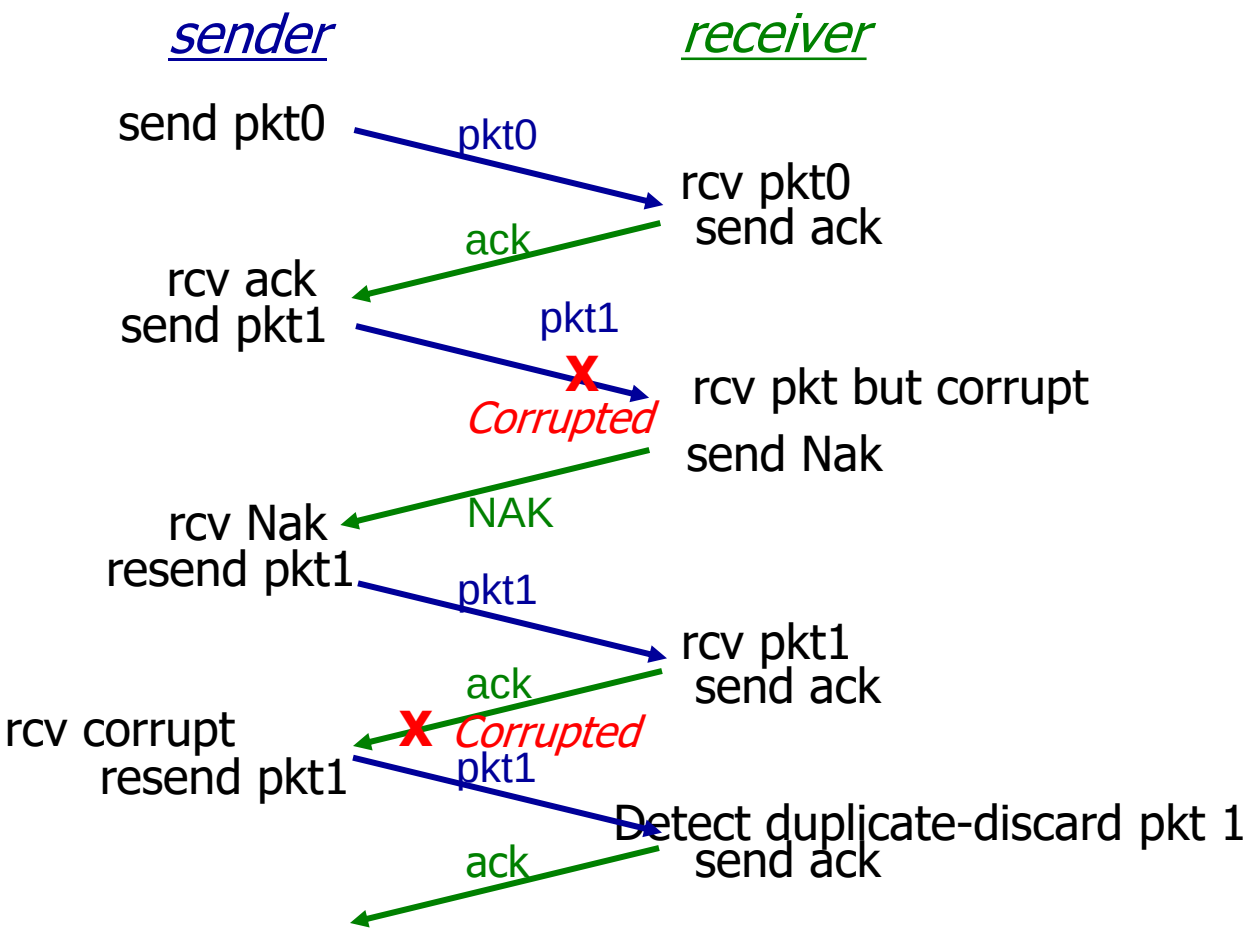
rdt2.1: sender, handles garbled ACK/NAKs



rdt2.1: receiver, handles garbled ACK/NAKs



rdt2.1 (Each DataPacket has a sequence number)



Packet and ack/nak corruption

rdt2.1: discussion

sender:

- seq # added to pkt
- two seq. #'s (0,1) will suffice.
- must check if received ACK/NAK corrupted
- twice as many states

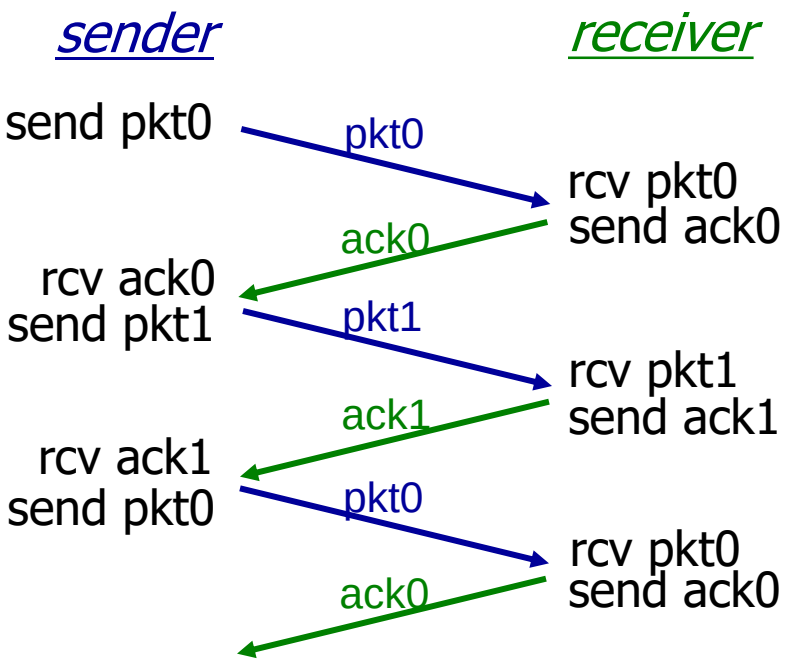
receiver:

- must check if received packet is duplicate
  - state indicates whether 0 or 1 is expected pkt seq #

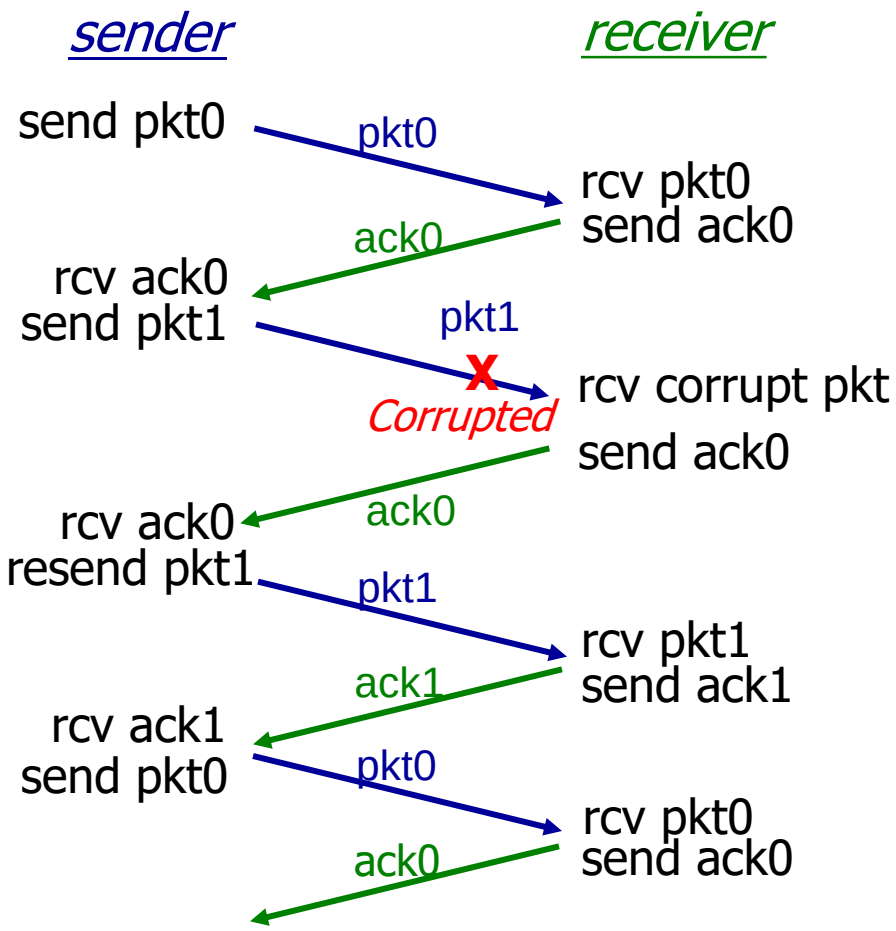
rdt2.2: a NAK-free protocol

- No changes in channel characteristics:
  - Channel introduces errors to the data and control packets
- same functionality as rdt2.1, using ACKs only
- instead of NAK, receiver sends ACK for last pkt received OK
  - receiver must *explicitly* include seq # of pkt being ACKed
- duplicate ACK at sender results in same action as NAK: *retransmit current pkt*
- NAK-free protocol puts foundation for more advanced 'reliable data transfer protocols' in the upcoming slides.

rdt2.2 in action (Add seq# to ACK/NAK too)

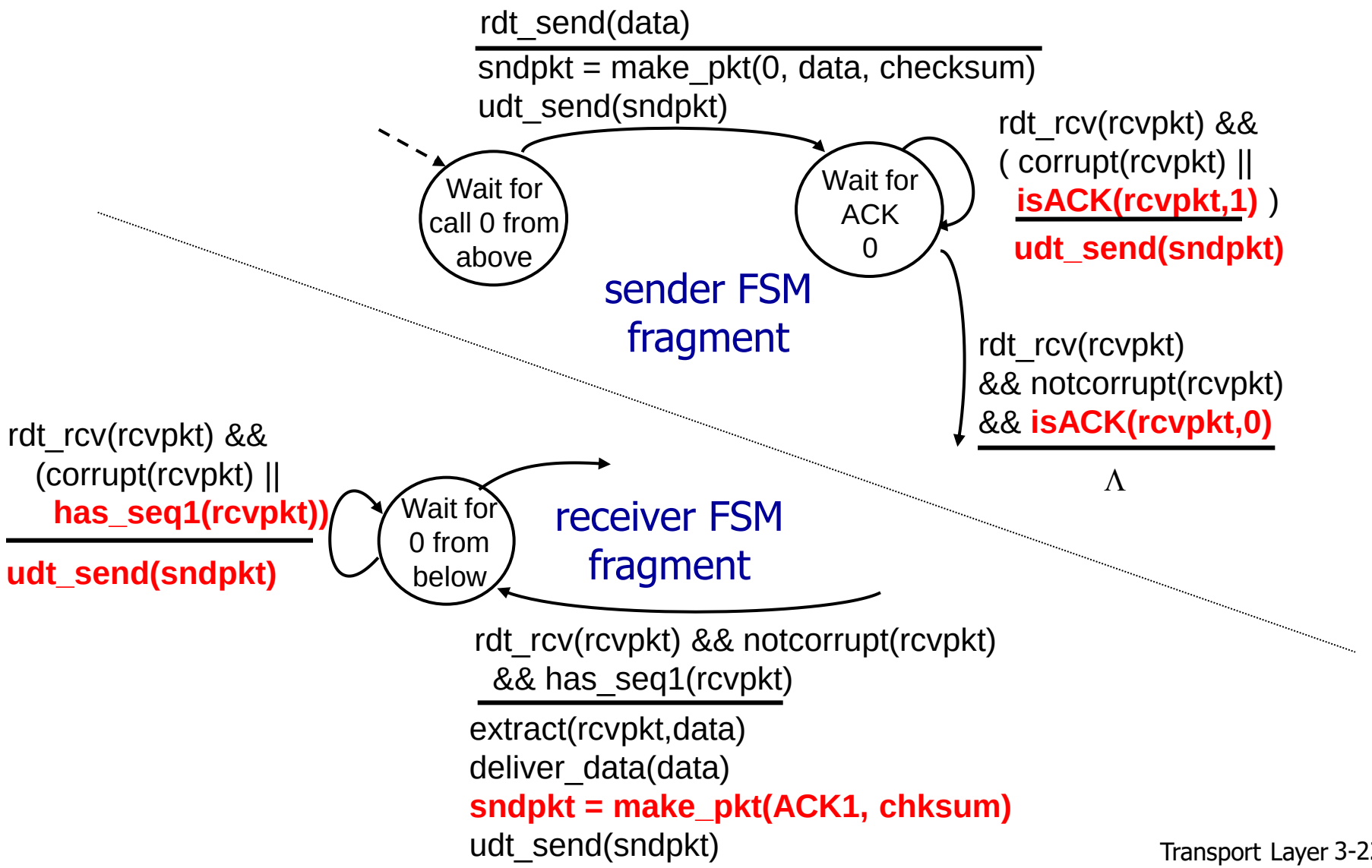


(a) no corruption

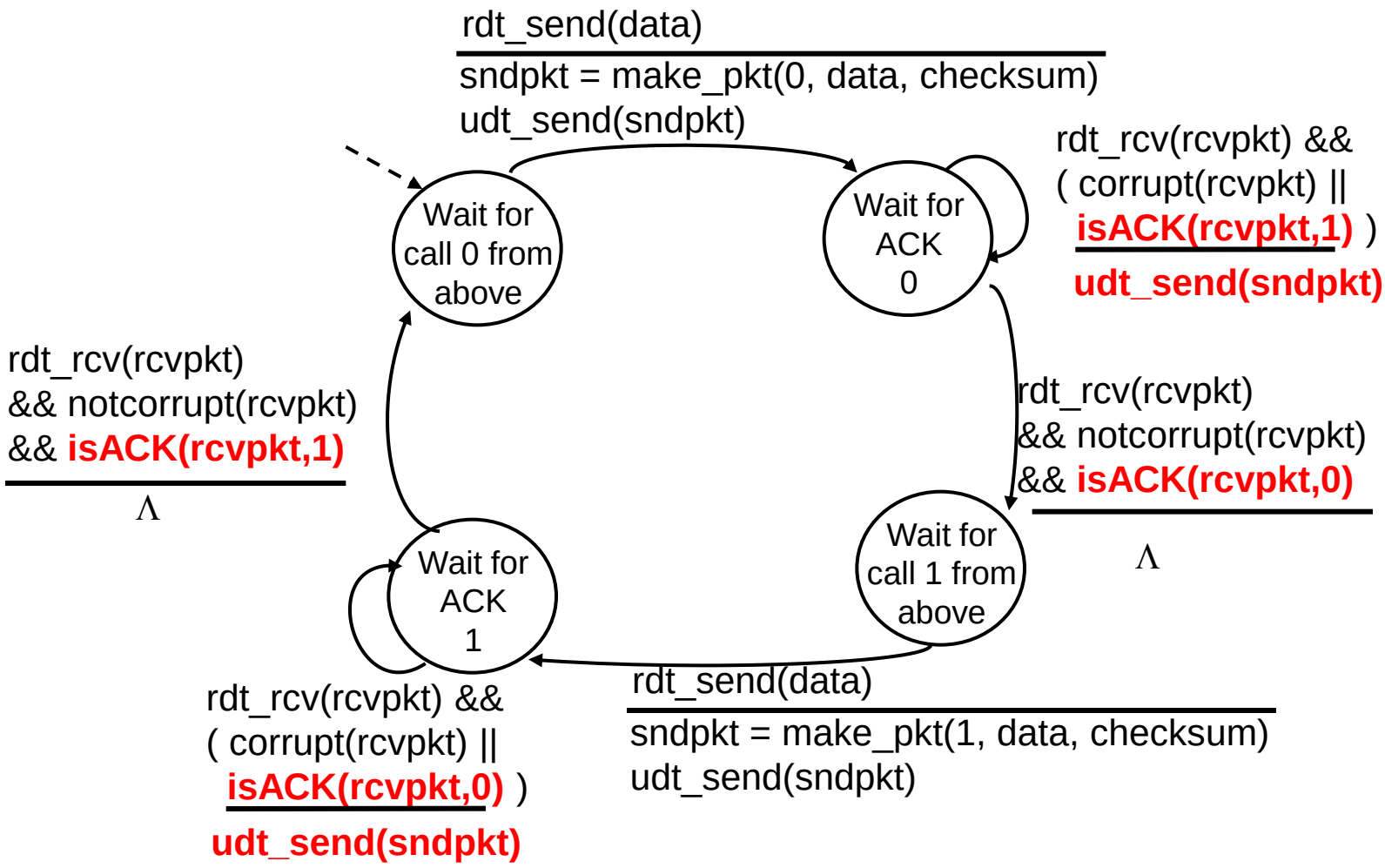


(b) Packet corruption

rdt2.2: sender, receiver fragments



rdt2.2: sender (full FSM)



rdt3.0: channels with errors *and* loss

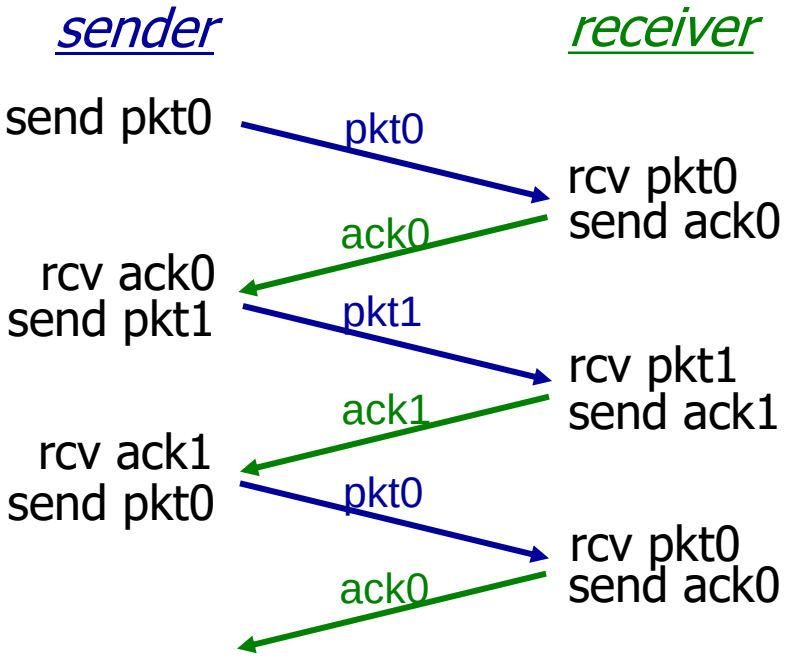
new assumption (or maybe relaxing one assumption of channel not being lossy anymore): underlying channel can also lose packets (data, ACKs)

- checksum, seq. #, ACKs, retransmissions will be of help … but not enough

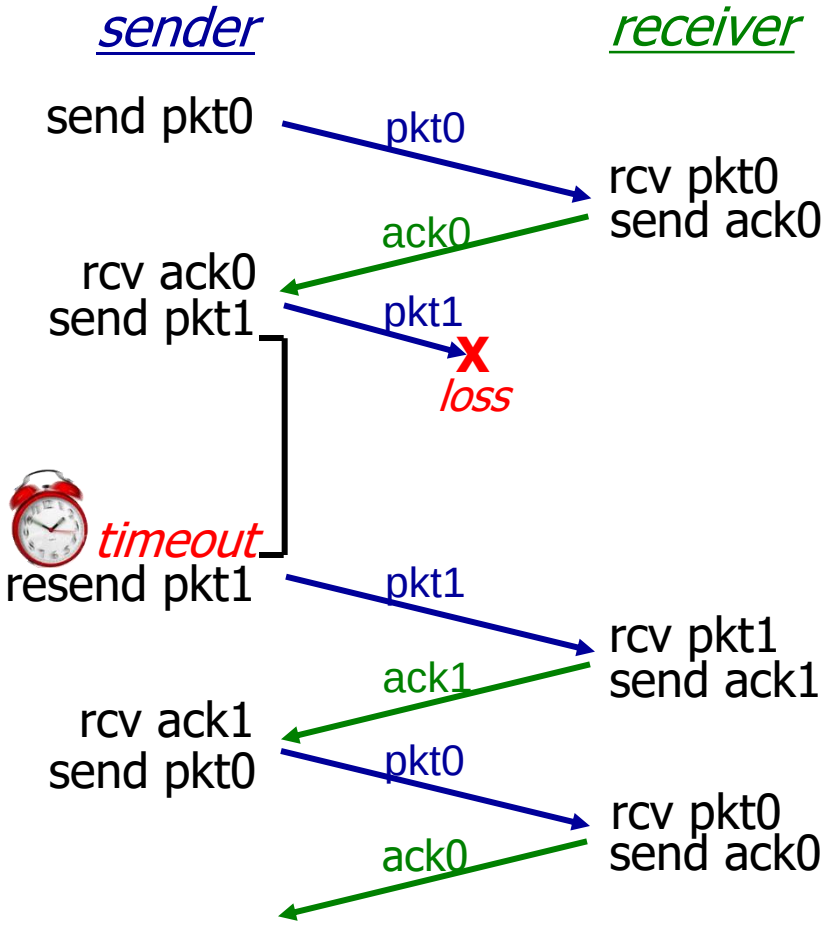
approach: sender waits "reasonable" amount of time for ACK

- retransmits if no ACK received in this time
- if pkt (or ACK) just delayed (not lost):
  - retransmission will be duplicate, but seq. #'s already handles this
  - receiver must specify seq # of pkt being ACKed
- requires countdown timer

rdt3.0 (no loss vs packet loss)

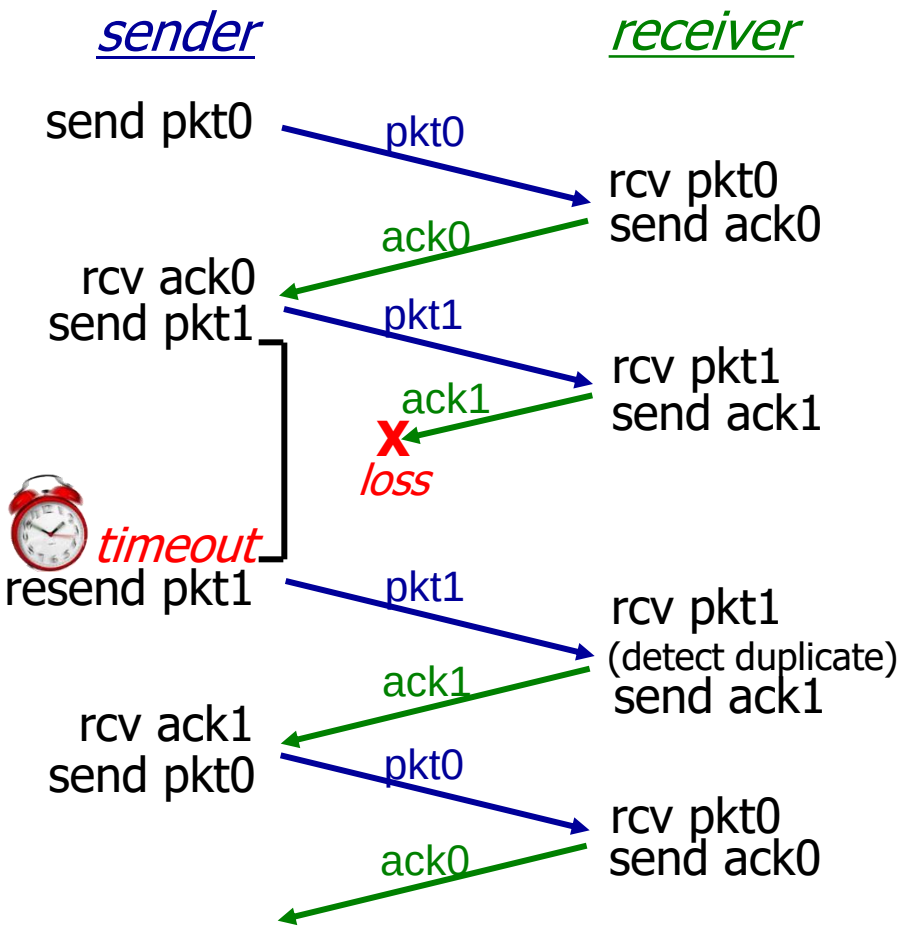


(a) no loss

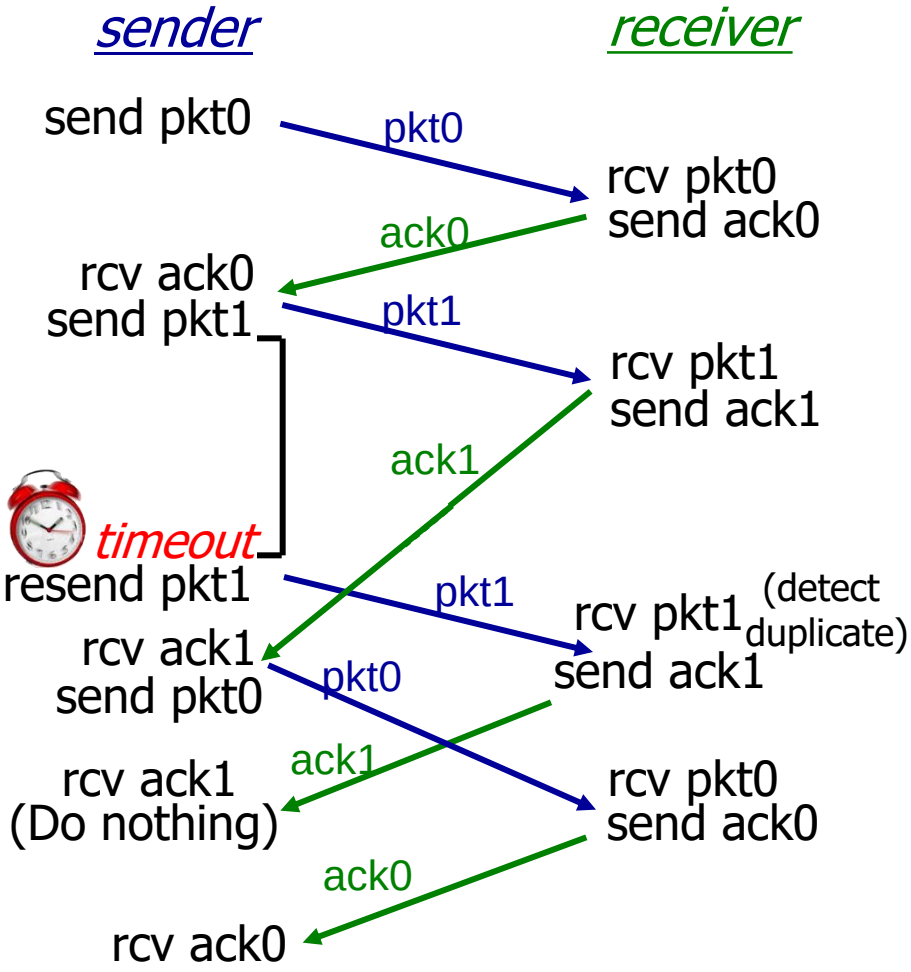


(b) packet loss

rdt3.0 (ACK loss vs ACK delayed)

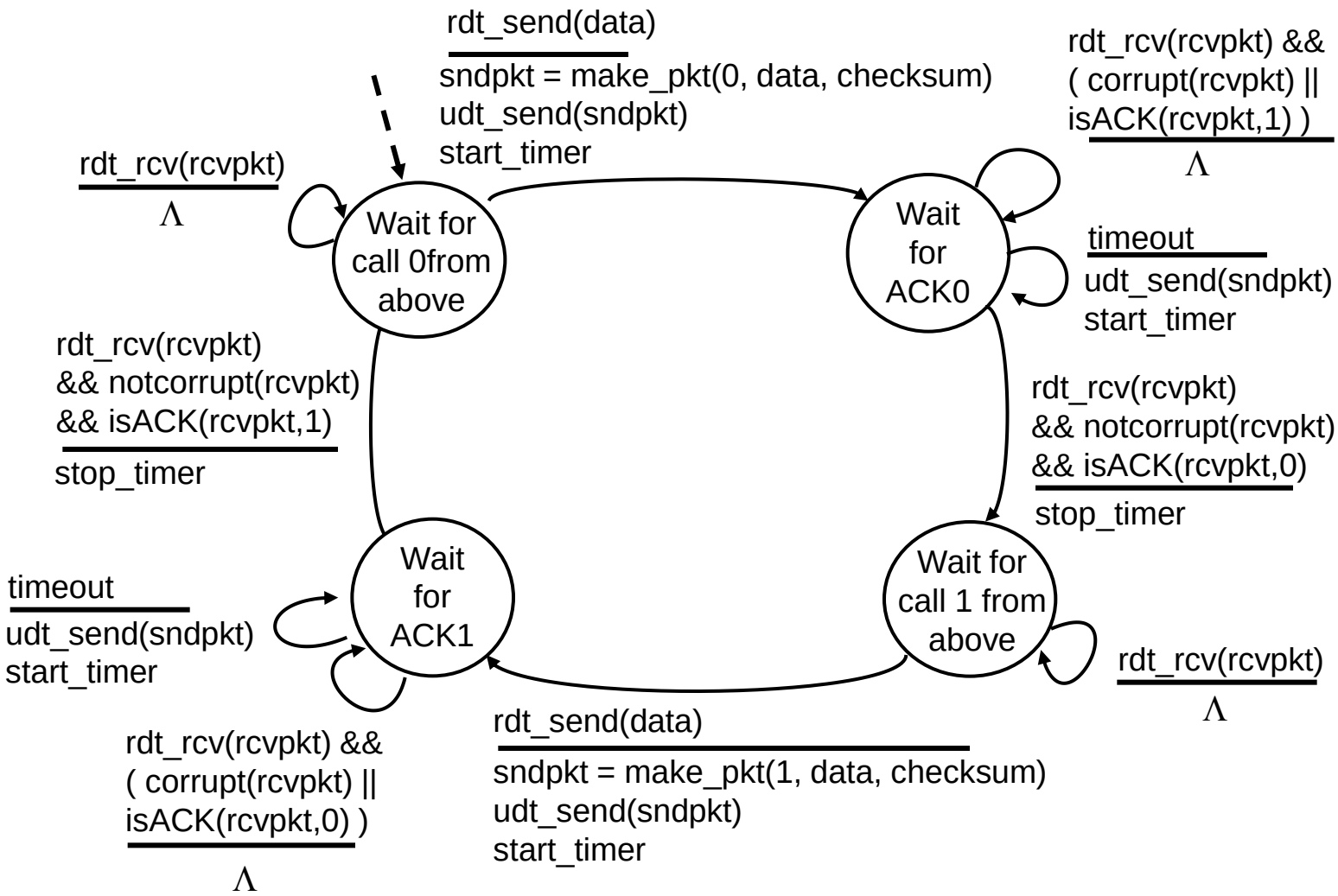


(c) ACK loss



(d) premature timeout/ delayed ACK

rdt3.0 sender



Performance of rdt3.0

- rdt3.0 is correct, but performance stinks
- e.g.: 1 Gbps link, 15 ms prop. delay, 8000 bit packet:

$$D_{trans} = \frac{L}{R} = \frac{8000\ bits}{10^9\ bits/sec} = 8\ microsecs = 0.008msec$$

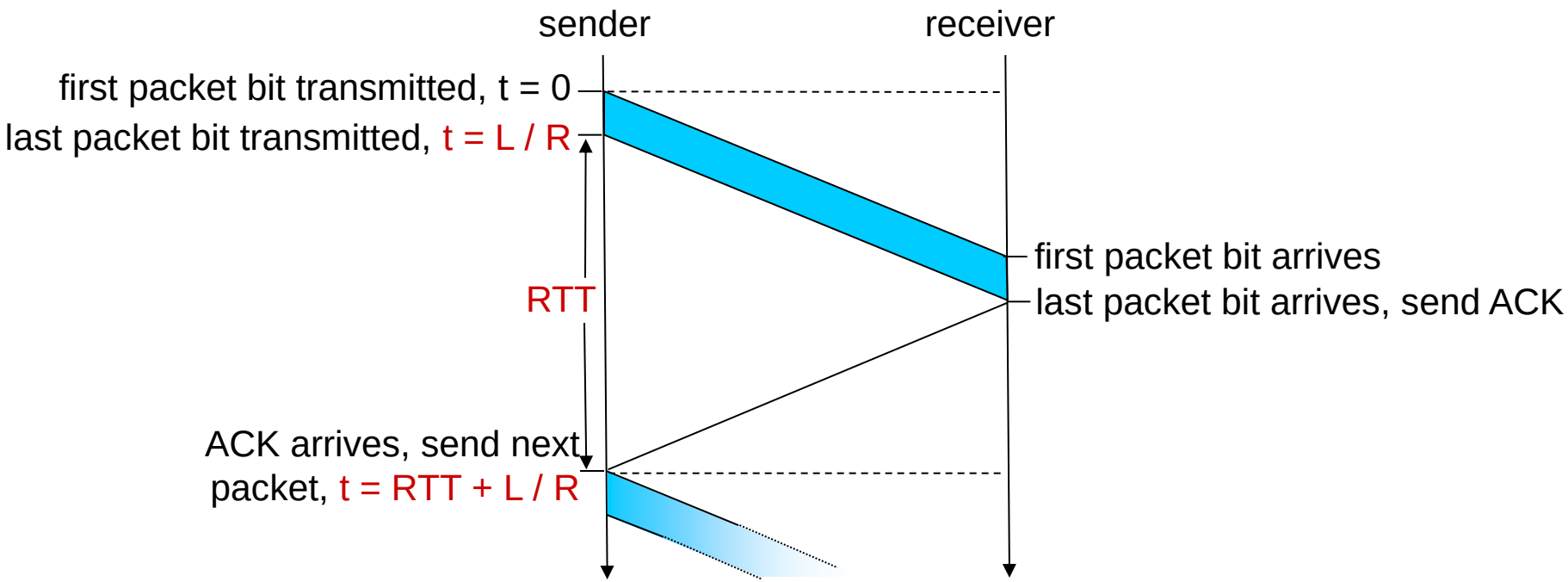
- Performance can be found using (1) utilization and (2) Throughput
  - U_{sender}: *utilization* is fraction of time sender busy sending

$$U_{sender} = \frac{L/R}{RTT + L/R} = \frac{.008}{30+0.008} = 0.00027$$

- if RTT=30 msec, and 1kB pkt every 30 msec (0.03 seconds) is transmitted over 1 Gbps link, then throughput is 33kB/sec

- network protocol (rdt3.0) limits use of physical resources!

rdt3.0: stop-and-wait operation



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