CS 348 Computer Networks



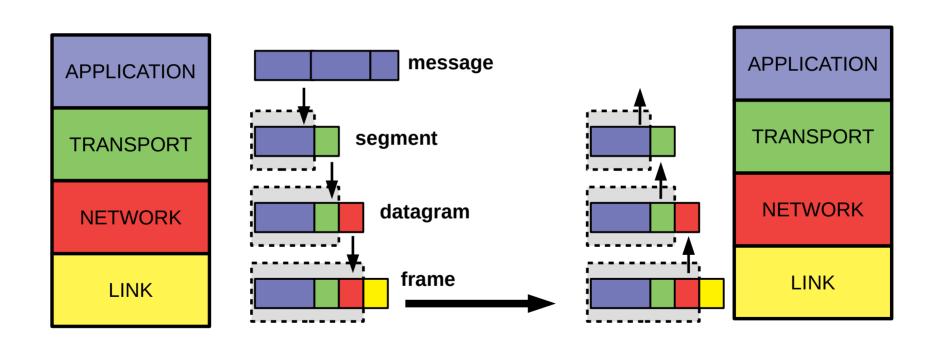
Lec 15 Principles of Reliable Data Transfer

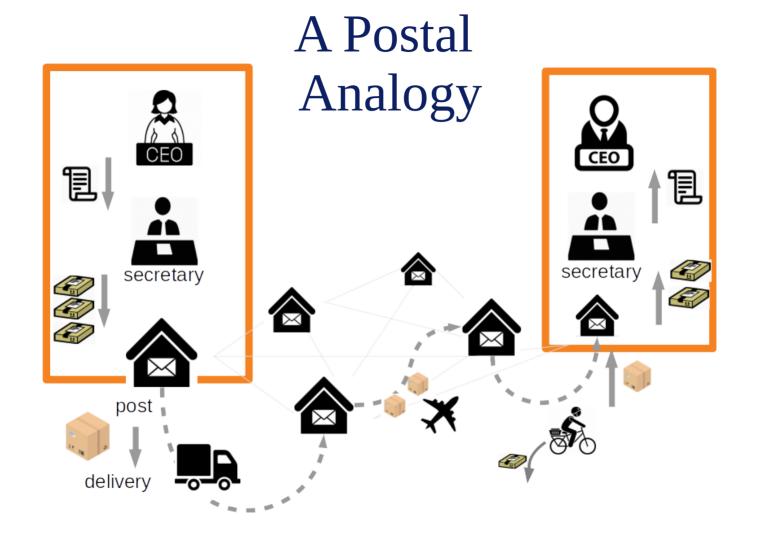
Spring 2020 IIT Goa

Course Instructor: Dr. Neha Karanjkar

Note: These slides are adapted from "Computer Networking: A Top-down Approach" by Kurose & Ross, 7th ed

Recall: The TCP/IP Model





Recall: Service Models

• **Network Layer:** Best-effort host-to-host delivery of a single datagram

NW layer: "Give me a payload and a destination IP address. I will create a single datagram containing the payload and make the best effort to deliver it to the destination, but can offer no guarantee of delivery (the packet can get corrupted or lost)."

• **Transport Layer:** Process-to-process message delivery service for applications, which can be reliable and in-order (TCP) or best-effort (UDP)

TCP: "Give me a message (as a sequence of Bytes) and a <dest IP, dest port> address, and I will split it into segments and deliver them reliably and in-order, to the correct process (socket) at the destination host."

UDP: "Give me a message and a <dest IP, dest port> address, and I will make a best-effort delivery of that message as a single packet to the correct process (socket) at the destination.

Reliable Data Transfer

• How can reliable data transfer be possible using the unreliable delivery service provided by the Network layer?

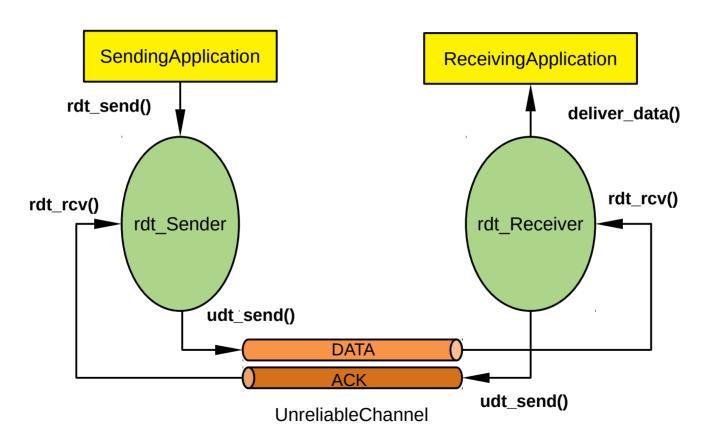
Reliable Data Transfer

- How can reliable data transfer be possible using the unreliable delivery service provided by the Network layer?
 - using acknowledgements, sequence numbers, timers etc

Reliable Data Transfer

- Some simplfying **assumptions** we make (to being with)
 - One side is the sender and the other is the receiver
 - The sender sends "data", the receiver sends back "acknowledgements"
 - There's a lossy, unreliable channel
 - A packet can get corrupted (with probability P_corruption)
 - A packet can get lost (with probability P_loss)
 - Packets are delivered in-order (if at all). Packets don't get reordered.

A Template



rdt1.0 A perfectly reliable channel

(P_corruption=0, P_loss=0)





rdt1.0: sending side

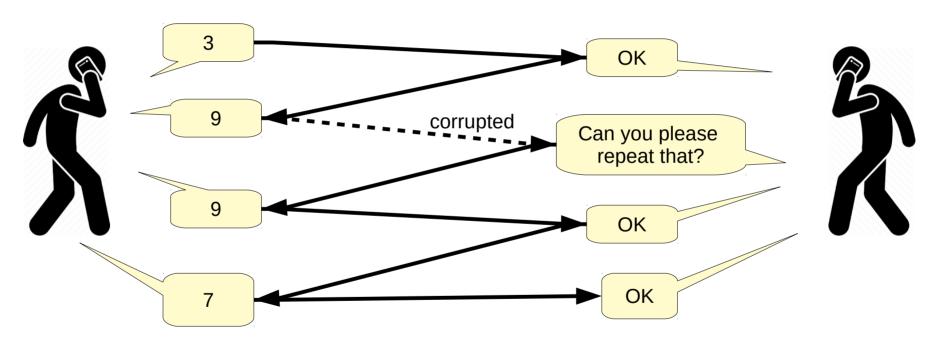
rdt1.0: receiving side

rdt2.0 Bit errors are possible

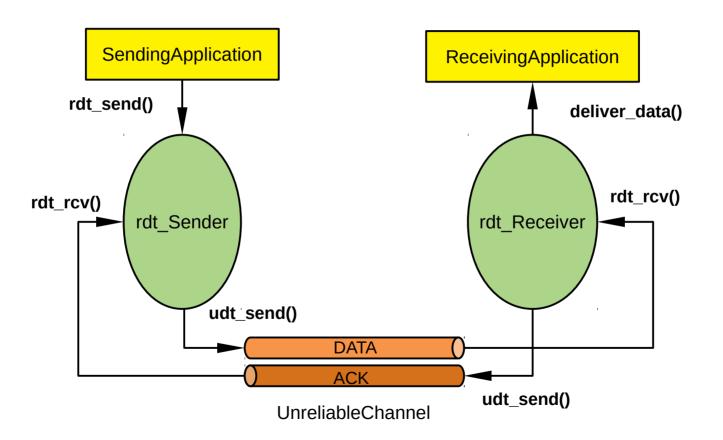
(P_corruption≠0, P_loss=0)

- The receiver needs:
 - Some mechanism to find out if the packet was corrupted
 - Send ACK/NAK
- The sender should re-transmit upon receiving a NAK

Analogy: person A is dictating a phone number to person B over a noisy phone connection

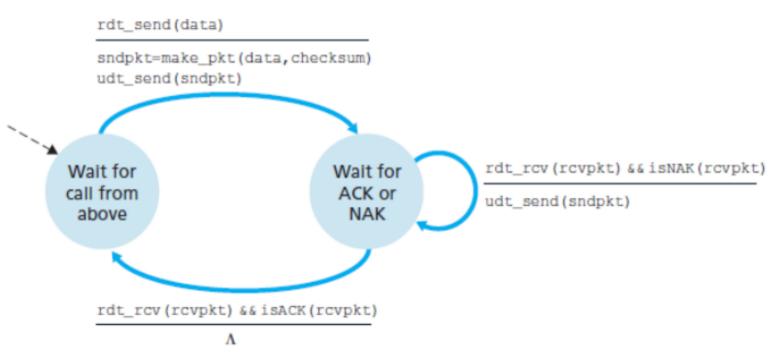


A Template



rdt2.0 Bit errors are possible

(P_corruption≠0, P_loss=0)



rdt2.0: sending side

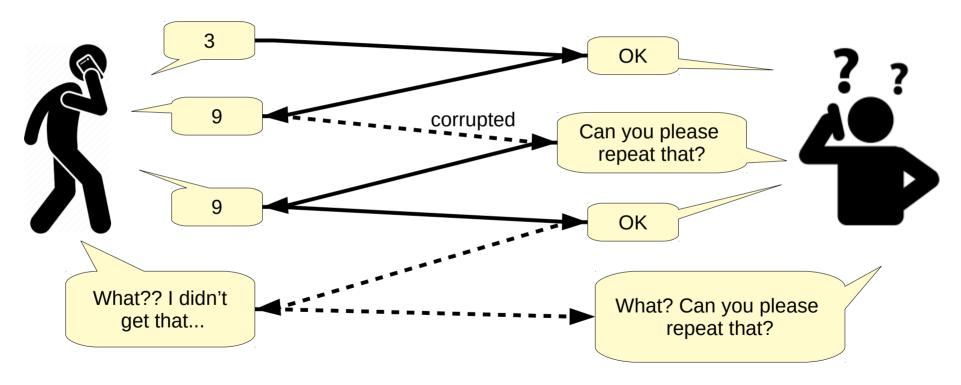
rdt2.0 Bit errors are possible

(P_corruption≠0, P_loss=0)

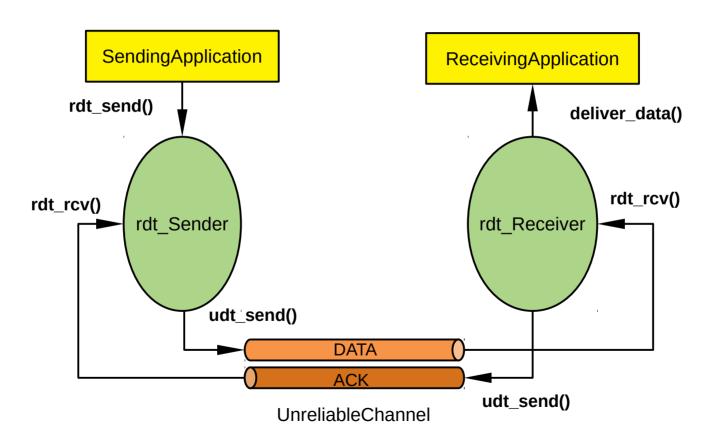
```
rdt_rcv(rcvpkt) && corrupt(rcvpkt)
          sndpkt=make_pkt(NAK)
          udt_send(sndpkt)
Wait for
call from
 below
          rdt_rcv(rcvpkt) && notcorrupt(rcvpkt)
          extract(rcvpkt,data)
          deliver data(data)
          sndpkt=make_pkt(ACK)
          udt send(sndpkt)
```

rdt2.1 Bit errors are possible (even in the ACK/NAK packets)

Analogy: person A is dictating a phone number to person B over a noisy phone connection



A Template

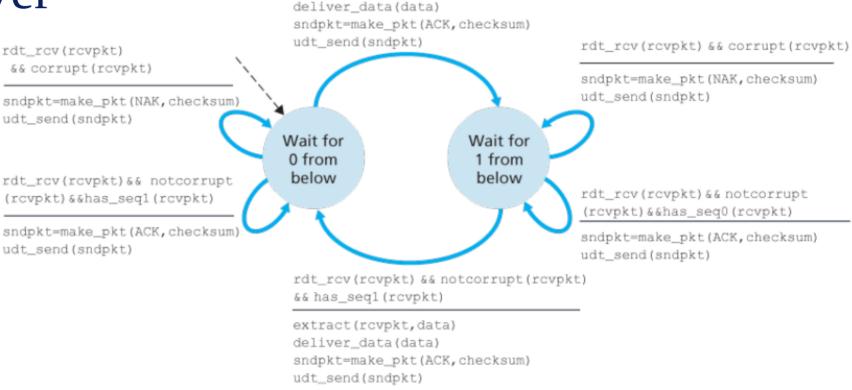


rdt_send(data) rdt2.1 sndpkt=make_pkt(0,data,checksum) udt_send(sndpkt) rdt_rcv(rcvpkt)&& sender (corrupt(rcvpkt) | | isNAK (rcvpkt)) Wait for Wait for udt_send(sndpkt) call 0 from ACK or above NAK 0 rdt_rcv(rcvpkt) rdt_rcv(rcvpkt) && notcorrupt (rcvpkt) && notcorrupt (rcvpkt) && isACK(rcvpkt) && isACK(rcvpkt) Λ Λ Wait for Wait for call 1 from ACK or NAK 1 above rdt_rcv(rcvpkt)&& (corrupt(rcvpkt) | | isNAK (rcvpkt)) rdt_send(data) udt_send(sndpkt) sndpkt=make_pkt(1,data,checksum)

udt_send(sndpkt)

Figure 3.11 rdt2.1 sender

rdt2.1 receiver



rdt_rcv(rcvpkt) && notcorrupt(rcvpkt)

&& has_seq0 (rcvpkt)

extract (rcvpkt, data)

Figure 3.12 rdt2.1 receiver

rdt2.2: We can use ACK 0 /ACK 1 to acknowledge the last successfully received packet. No need of a NAK.

rdt2.2 sender

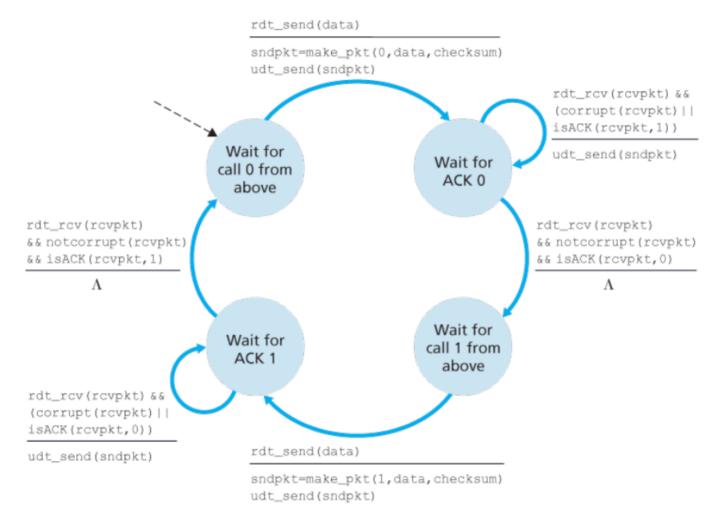


Figure 3.13 rdt2.2 sender

rdt2.2 rdt_rcv(rcvpkt) && notcorrupt(rcvpkt) && has_seq0 (rcvpkt) receiver extract (rcvpkt, data) deliver_data(data) rdt_rcv(rcvpkt) && sndpkt=make_pkt(ACK,0,checksum) (corrupt (rcvpkt) | | udt_send(sndpkt) has_seq0(rcvpkt)) sndpkt=make_pkt(ACK,0,checksum) udt_send(sndpkt) Wait for Wait for rdt rcv(rcvpkt) && 0 from 1 from (corrupt(rcvpkt) | | below below has_seq1 (rcvpkt)) sndpkt=make_pkt(ACK,1,checksum) udt send(sndpkt) rdt_rcv(rcvpkt) && notcorrupt(rcvpkt) && has_seq1 (rcvpkt)

extract(rcvpkt,data) deliver_data(data)

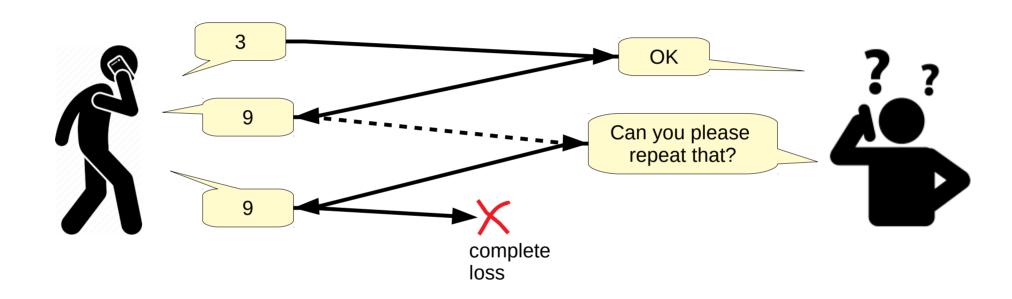
udt_send(sndpkt)

sndpkt=make_pkt(ACK,1,checksum)

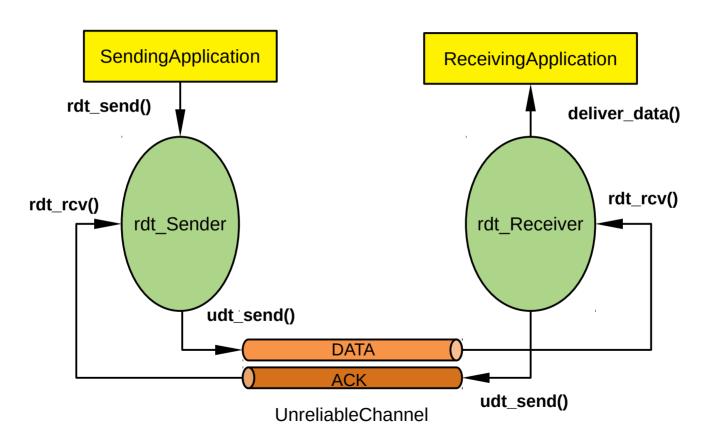
Figure 3.14 rdt2.2 receiver

rdt3.0 Bit errors are possible and packet loss is also possible

(P_corruption≠0, P_loss≠0)



A Template



rdt3.0 sender

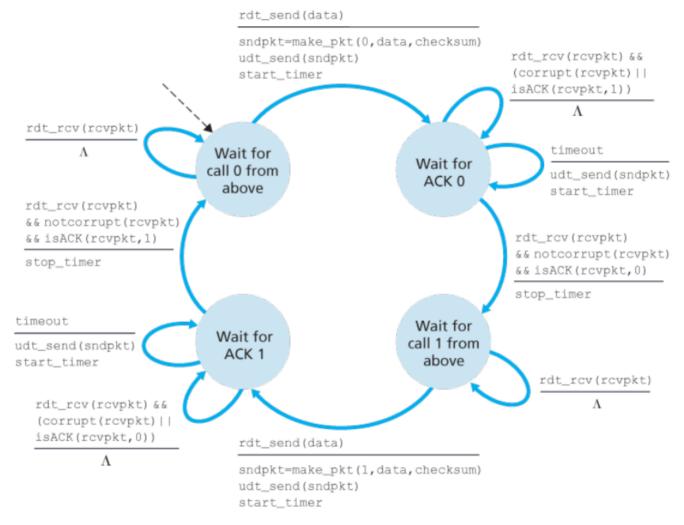


Figure 3.15 rdt3.0 sender

rdt3.0 receiver ??

Exercise

• [LAB] Design, simulate and test these protocols
Using Python+SimPy

https://nehakaranjkar.github.io/ProtocolSimulation.html

References and Reading Assignment

• **Kurose and Ross 6**th **ed:** Section 3.4

So far...

- Structure and Physical components of the Internet
- Design of the Internet: Layering and Encapsulation
- The Applications Layer:
 - Sockets Interface
 - The Web and HTTP
 - DNS
- The Transport Layer: how it works
 - Basic services, UDP
- **Principles of Reliable Data Transfer**
 - Pipelined data transfer (Sliding window protocols)
 - TCP details
 - Congestion and Flow control