

CS 352

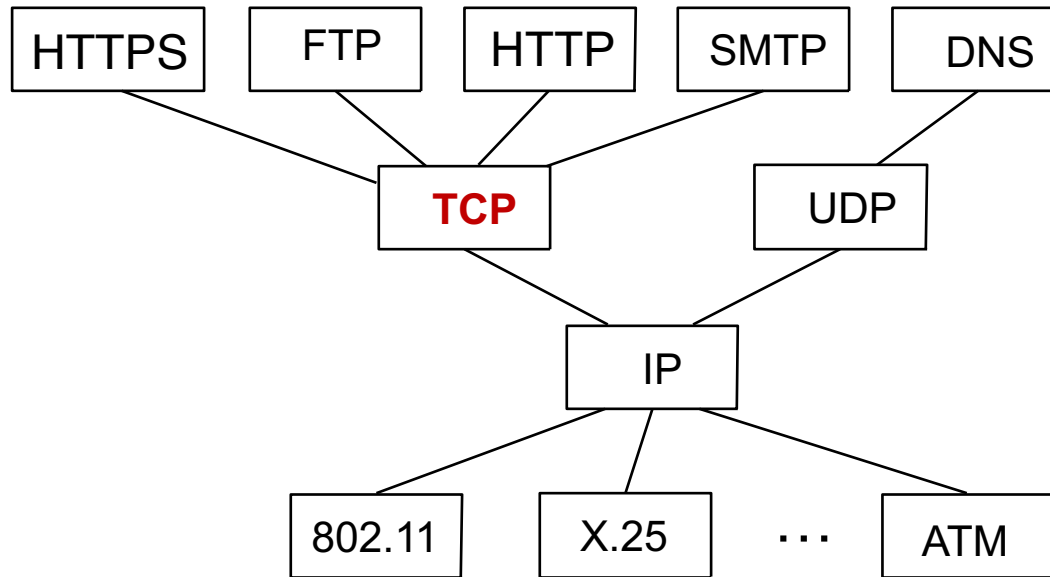
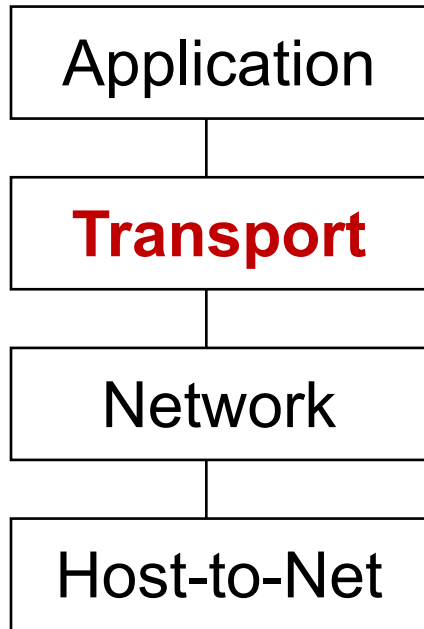
Reliability: Stop and Wait

CS 352, Lecture 9.1

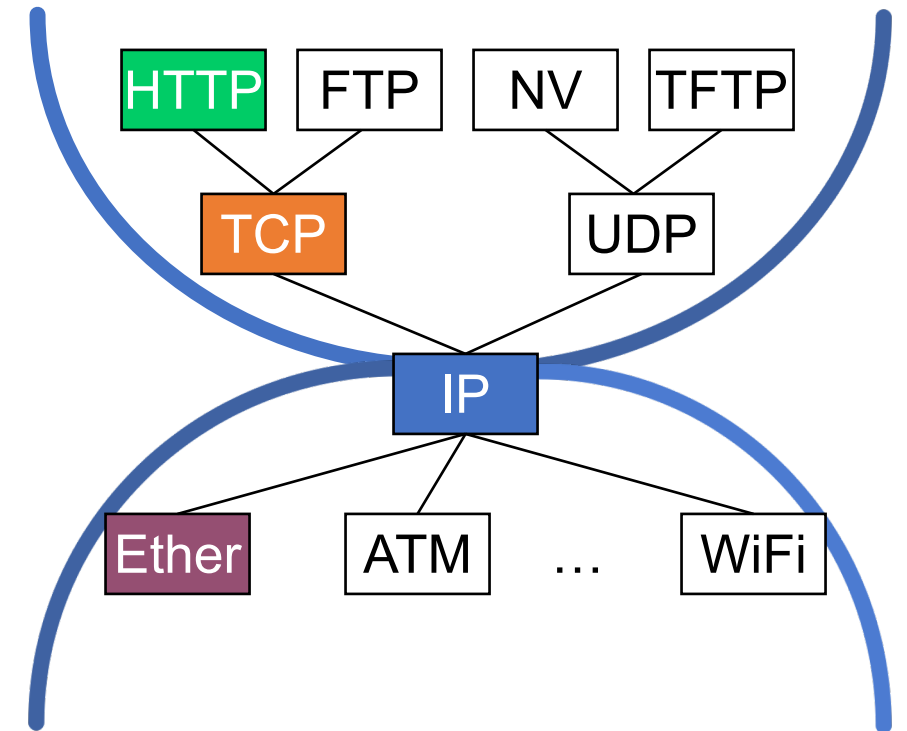
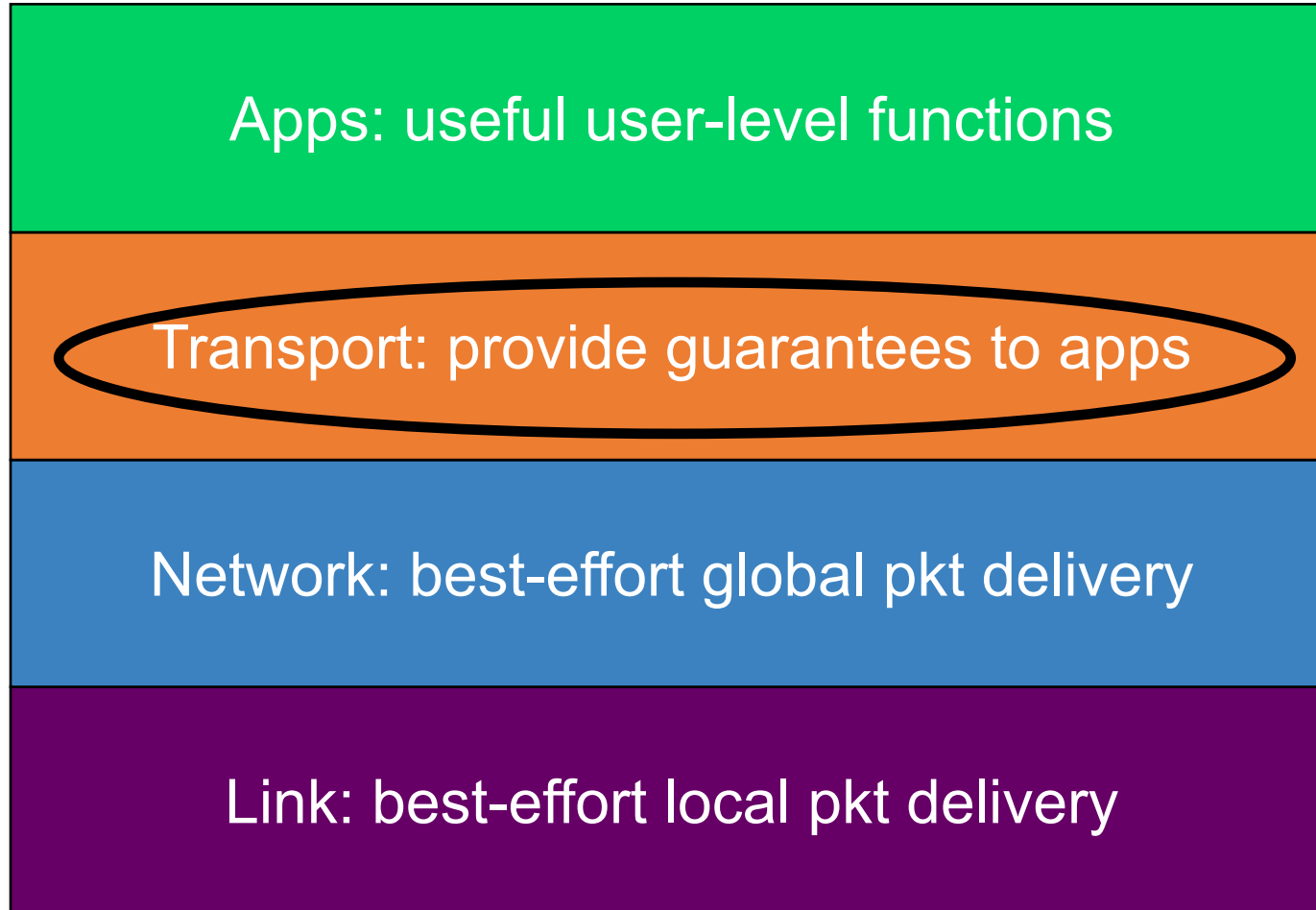
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Transport

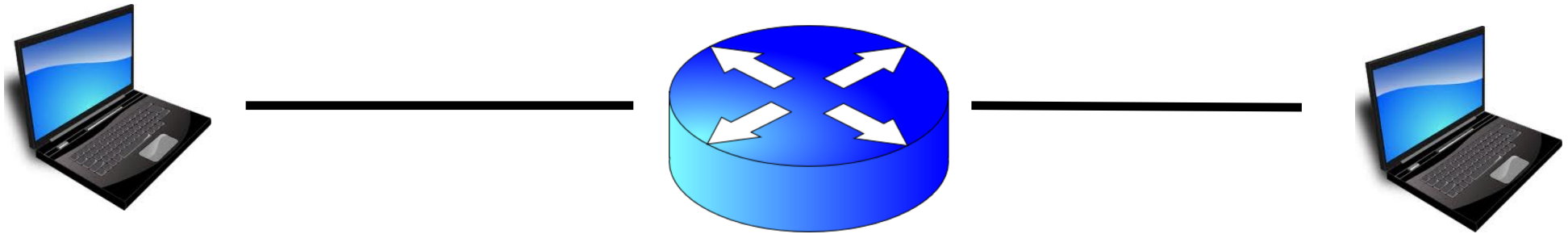


Modularity through layering



How do apps get perf guarantees?

- The network core provides no guarantees on packet delivery



- Transport software on the endpoint oversees implementing guarantees on top of a best-effort network

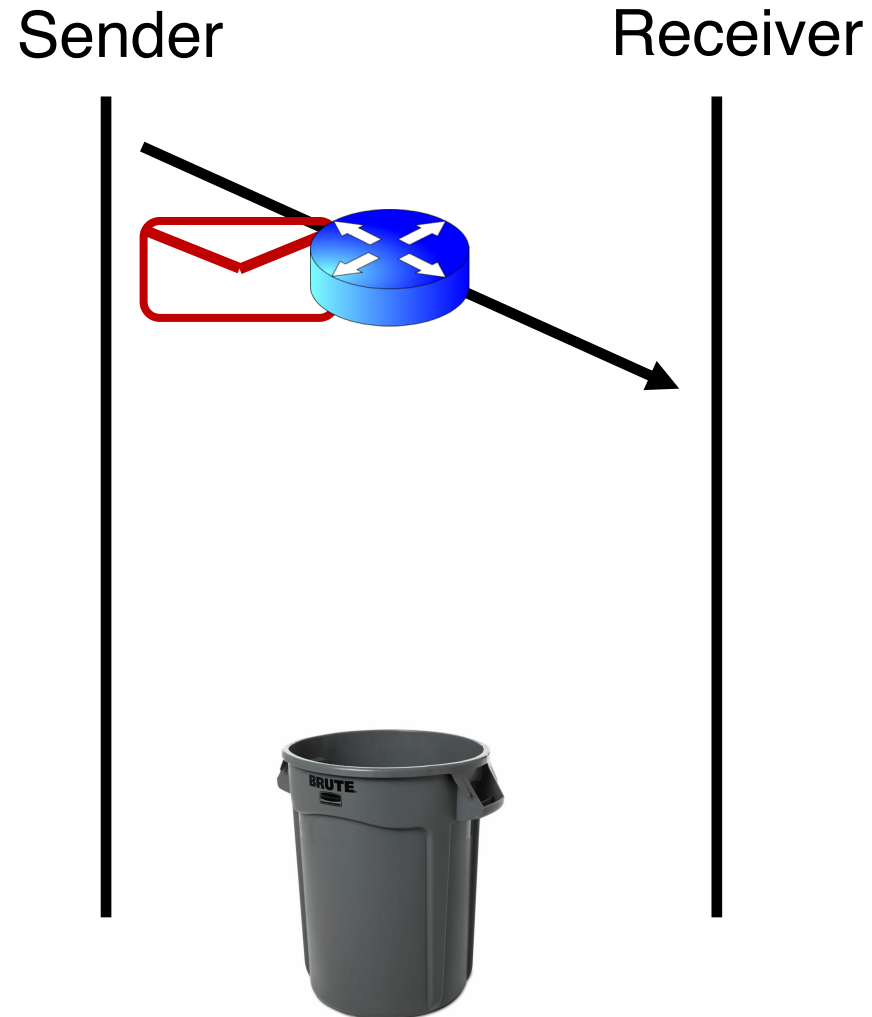
- Three important kinds of guarantees

- **Reliability**
 - Ordered delivery
 - Resource sharing in the network core

} Transmission
Control Protocol
(TCP)

Reliable data delivery

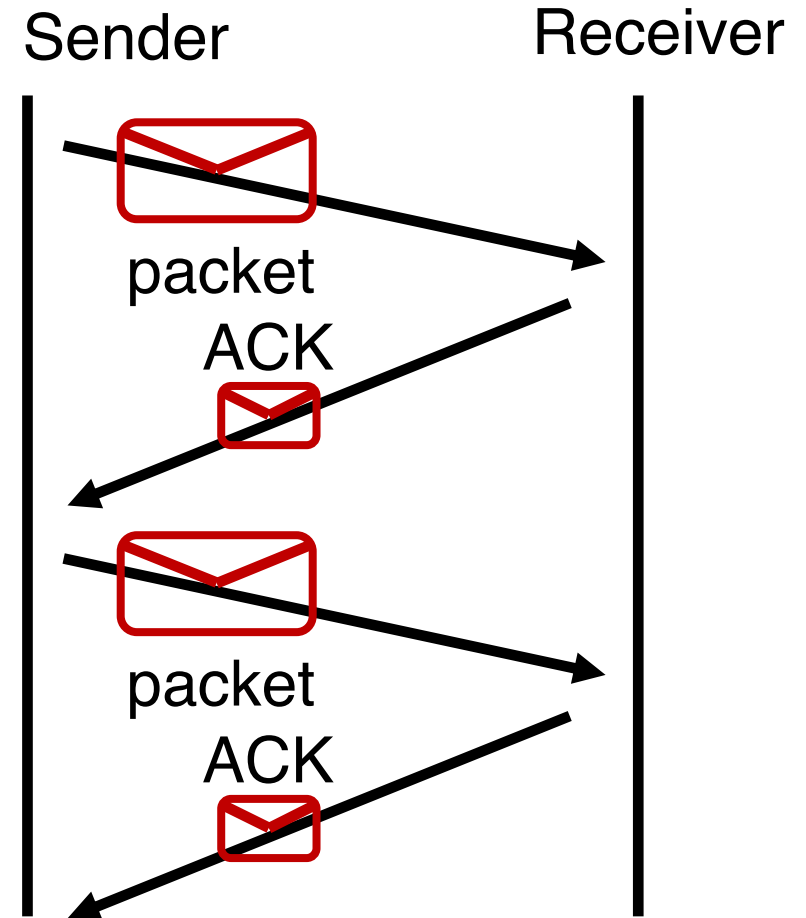
Packet loss



- How might a sender and receiver ensure that data is delivered reliably (despite some packets being lost)?
- TCP uses three mechanisms

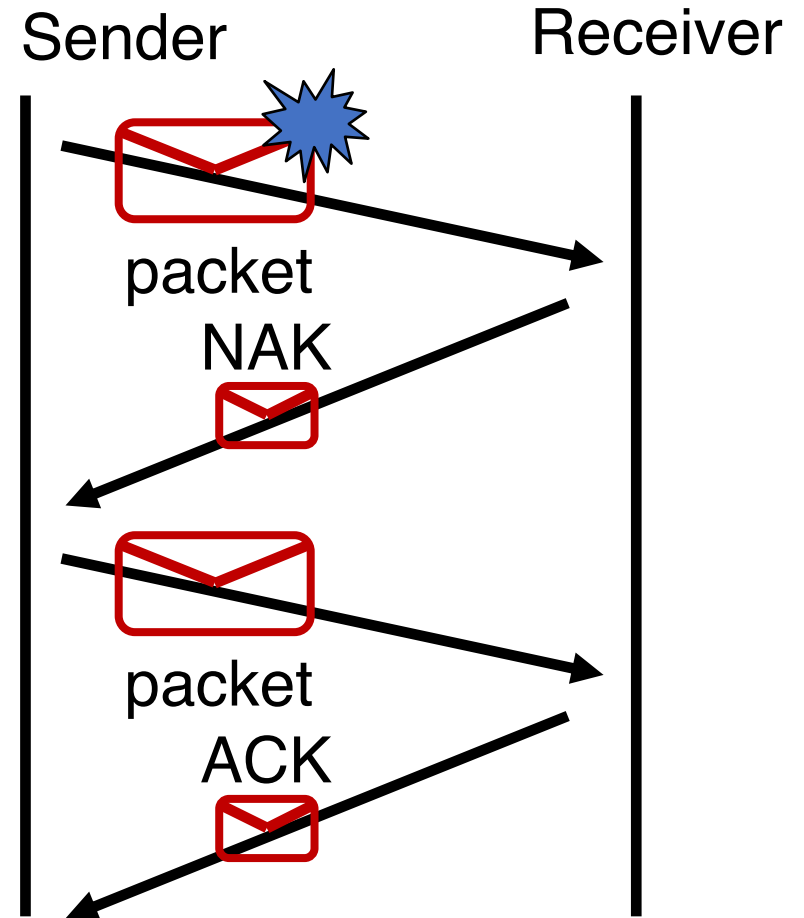
Coping with packet loss: (1) ACK

- Key idea: Receiver returns an **acknowledgment** (ACK) per packet sent
- If sender receives an ACK, it knows that the receiver got the packet.



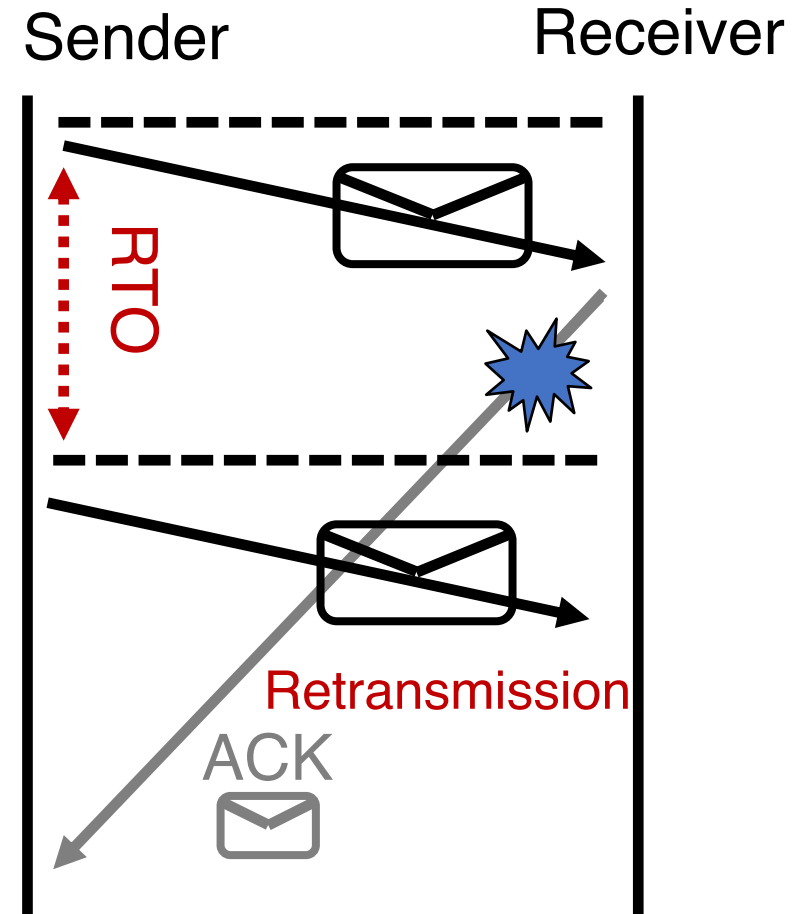
Coping with packet **corruption**: (1) ACK

- ACKs also work to detect packet corruption on the way to the receiver
 - One possibility: A receiver could send a negative acknowledgment, or a **NAK**, if it receives a corrupted packet
 - Q: How to detect corrupted packet?
 - **One method: Checksum!**
- TCP only uses positive ACKs.



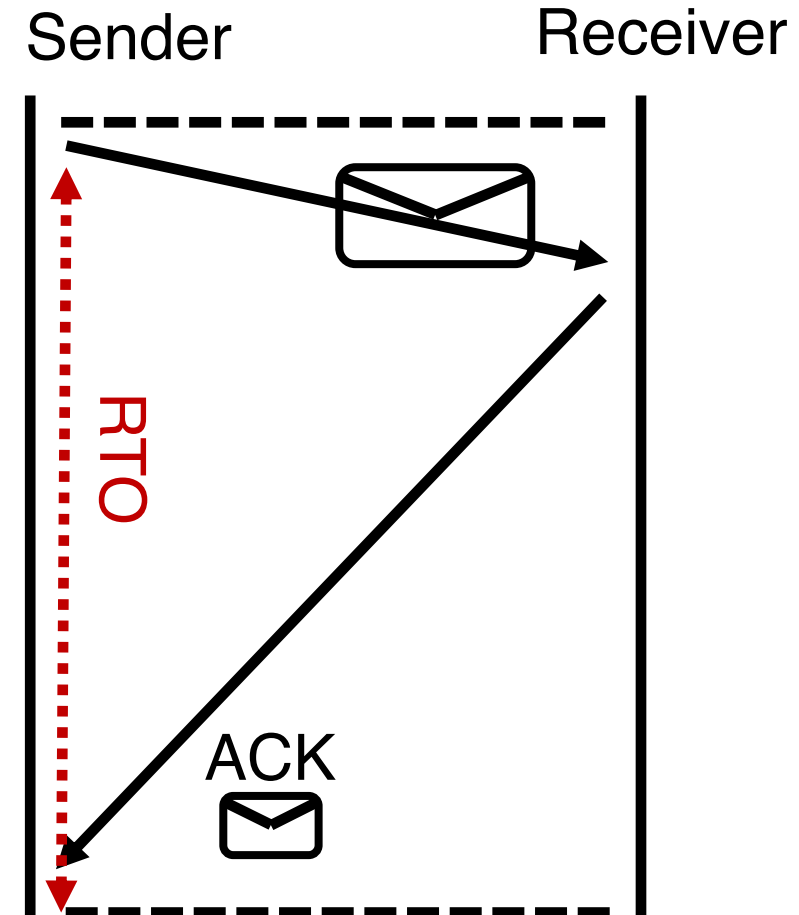
Coping with packet loss: (2) RTO

- What if a packet is dropped?
- Key idea: Wait for a duration of time (called **retransmission timeout** or RTO) before **re-sending** the packet
- In TCP, **the onus is on the sender** to retransmit lost data when ACKs are not received
- Note that retransmission works also if **ACKs are lost or delayed**



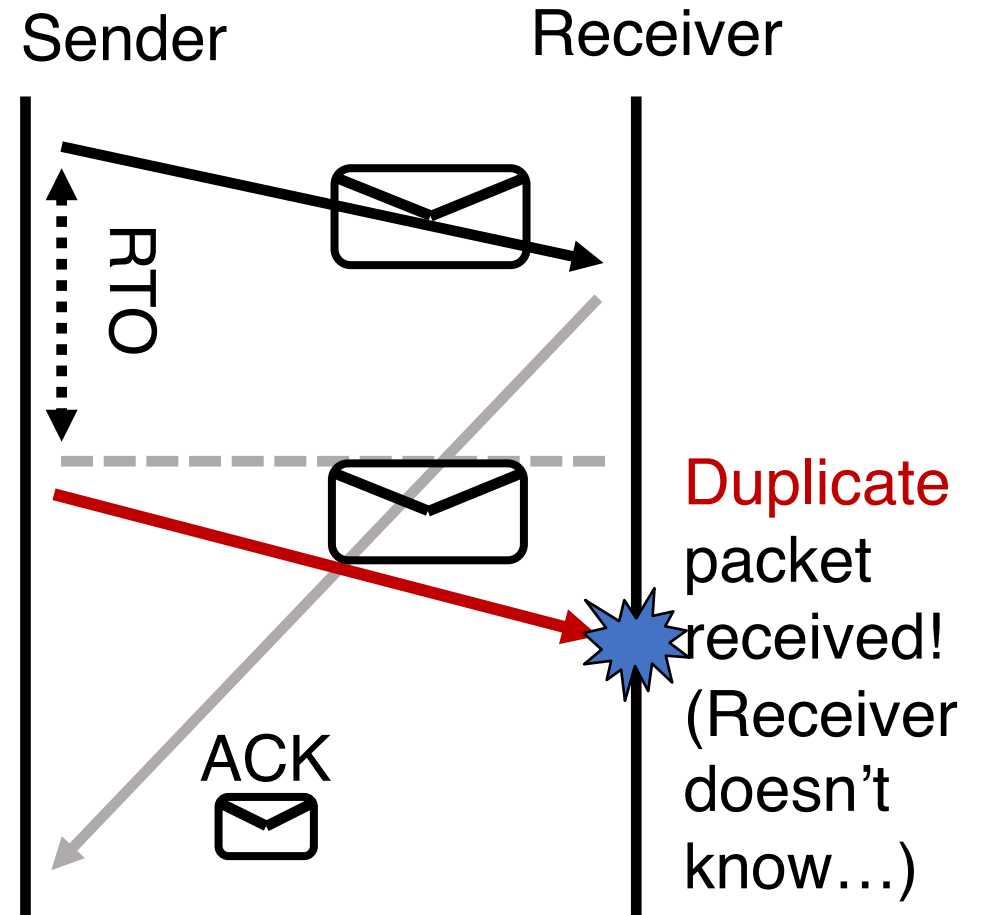
How should the RTO be set?

- A good RTO must **predict** the **round-trip time** (RTT) between the sender and receiver
 - RTT: the time to send a single packet and receive a (corresponding) single ACK at the sender
- Intuition: If an ACK hasn't returned, and our (best estimate of) RTT has elapsed, the packet was likely dropped.
- RTT can be measured directly at the sender. No receiver involvement needed.



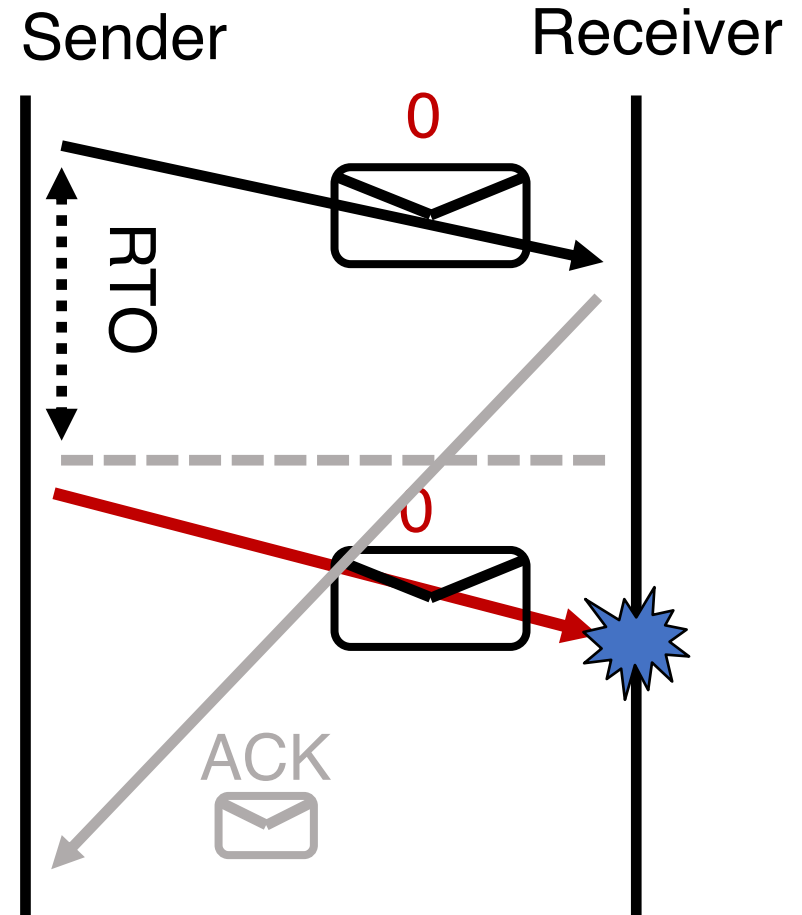
Coping with packet duplication

- If ACKs delayed beyond the RTO, sender may retransmit the **same** data
 - Receiver wouldn't know that it just received duplicate data from this retransmitted packet
- Add some identification to each packet to help distinguish between adjacent transmissions
 - This is known as the **sequence number**



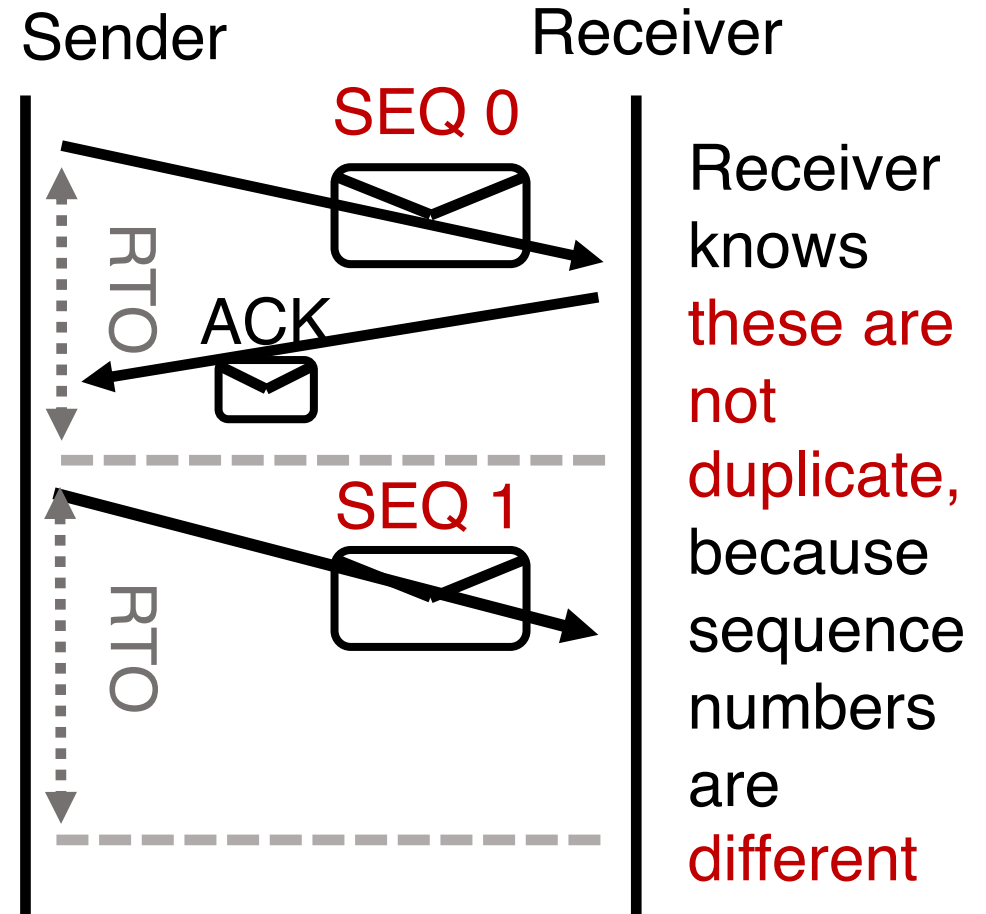
Coping with packet loss: (3) Sequence #s

- A bad scenario: Suppose an ACK was delayed beyond the RTO; sender ended up retransmitting the packet.
- At the receiver: **sequence number helps disambiguate a fresh transmission from a retransmission**
 - Sequence number same as earlier: retransmission
 - Fresh sequence number: fresh data



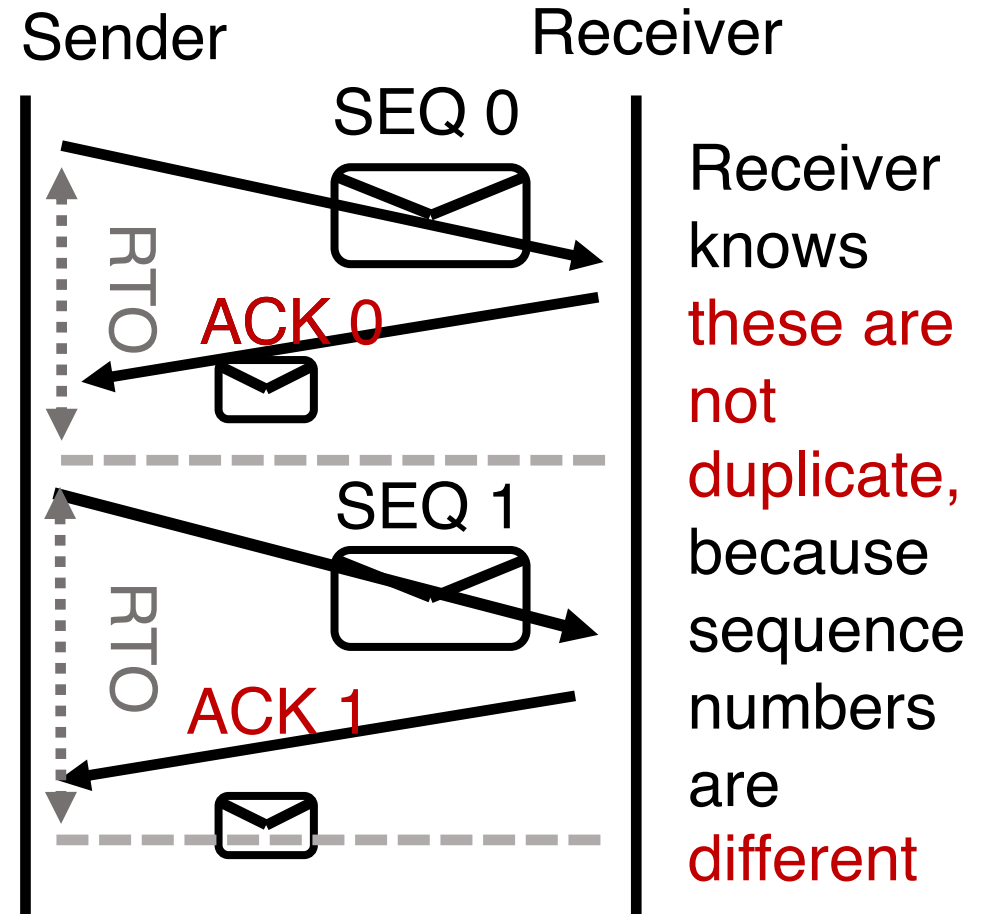
Coping with packet loss: (3) Sequence #s

- A good scenario: packet successfully received and ACK returned within RTO
- Sequence numbers of successively transmitted packets are different



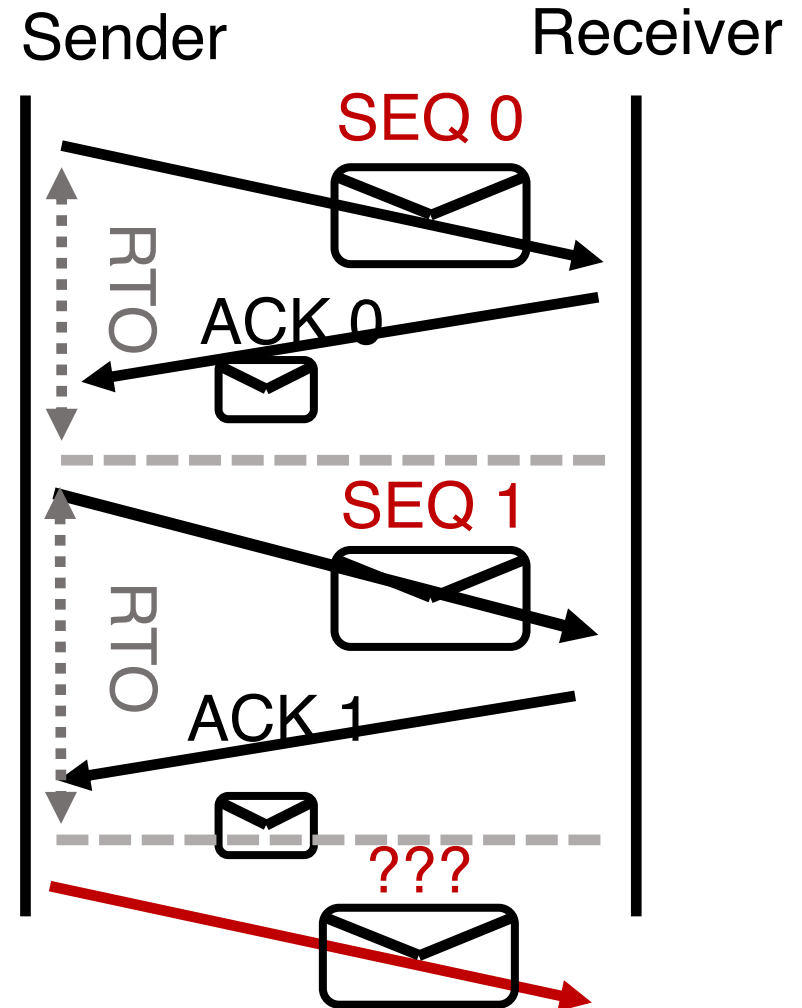
Coping with packet loss: (3) Sequence #s

- A good scenario: packet successfully received and ACK returned within RTO
- Sequence numbers of successively transmitted packets are different
- Further, the receiver informs the sender which packet was ACK'ed using an **ACK sequence number**



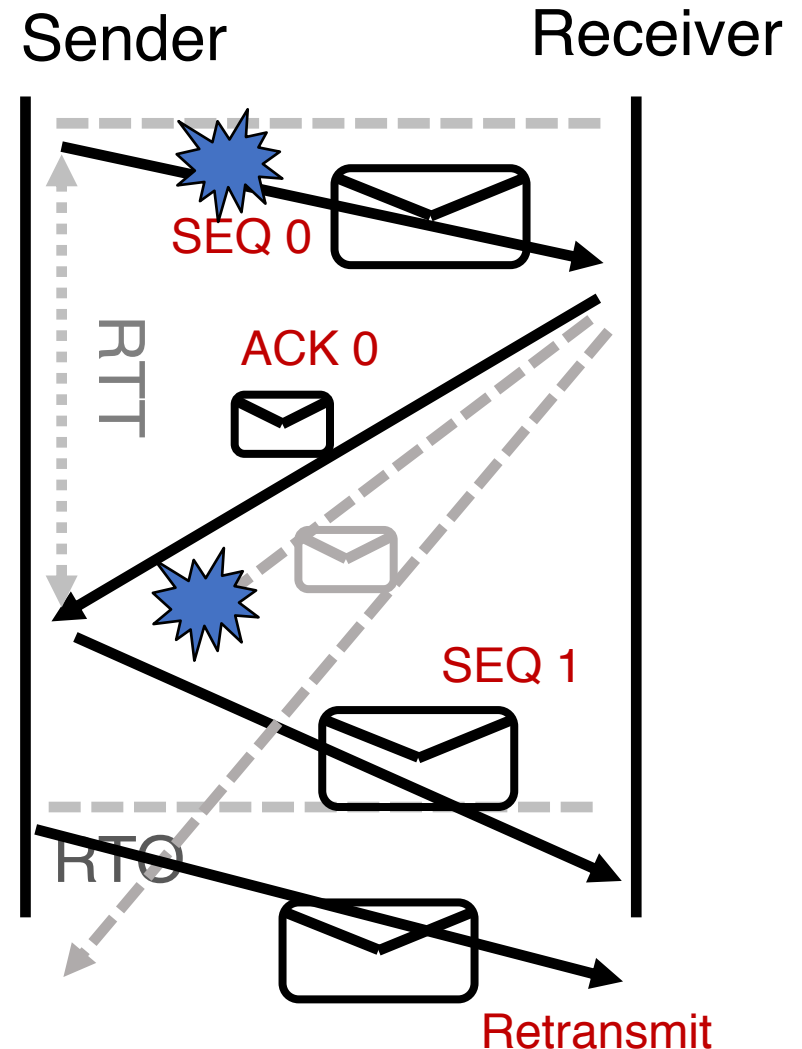
Q: What is the seq# of third packet?

- Goal: Avoid ambiguity on which packet was received/ACK'ed from both the sender and receiver's perspective
- One possibility: keep incrementing the seq #: 2, 3, ...
- Alternative: since seq # 0 was successfully ACK'ed earlier, it is OK to reuse seq #0 for next transmission.
 - Seq #s reused if enough time elapsed



Summary: Stop-and-Wait Reliability

- Sender sends a single packet, then waits for an ACK to know the packet was successfully received. Then the sender transmits the next packet.
- If ACK is not received until a timeout (RTO), sender **retransmits** the packet
- Disambiguate duplicate vs. fresh packets using sequence numbers that change on “adjacent” packets



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Reliability: TCP Metadata

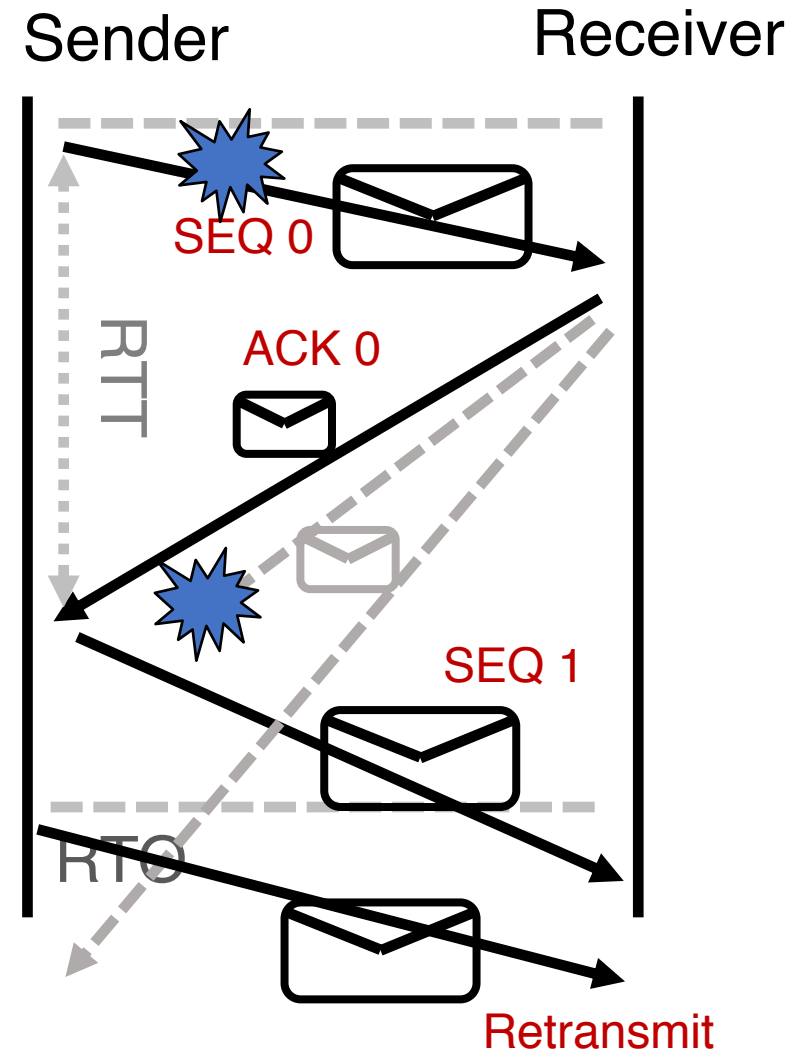
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Review: Stop-and-Wait Reliability

- Sender sends a single packet, then waits for an ACK to know the packet was successfully received. Then the sender transmits the next packet.
- If ACK is not received until a timeout (RTO), sender **retransmits** the packet
- Disambiguate duplicate vs. fresh packets using sequence numbers that change on “adjacent” packets



Q1: Where are seq & ACK #s written to?

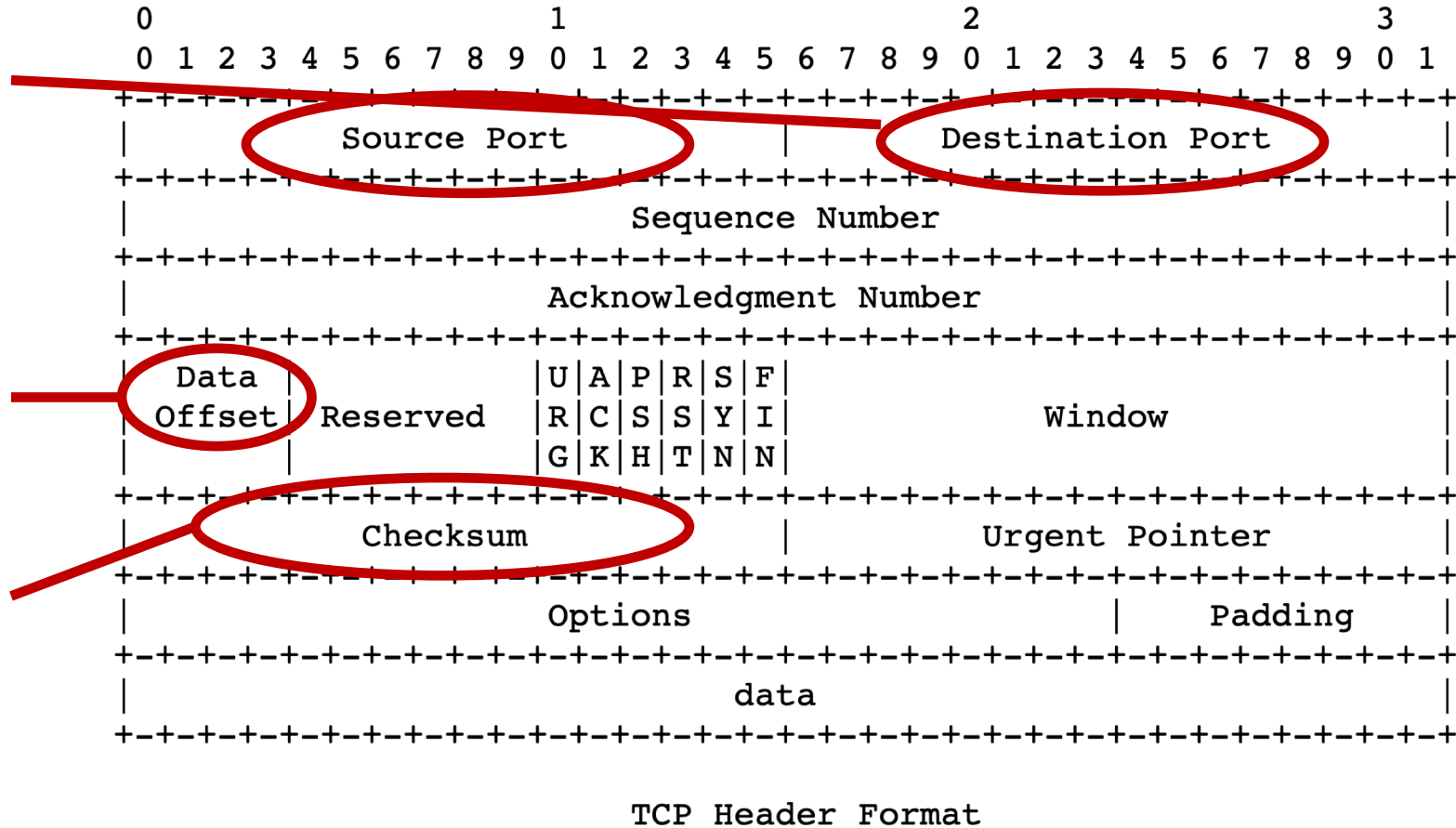
- Naturally, in the packet header!

TCP header structure

Source port, destination port (connection demultiplexing)

Size of the TCP header (in 32-bit words)

Basic error detection through checksums (similar to UDP)



Note that one tick mark represents one bit position.

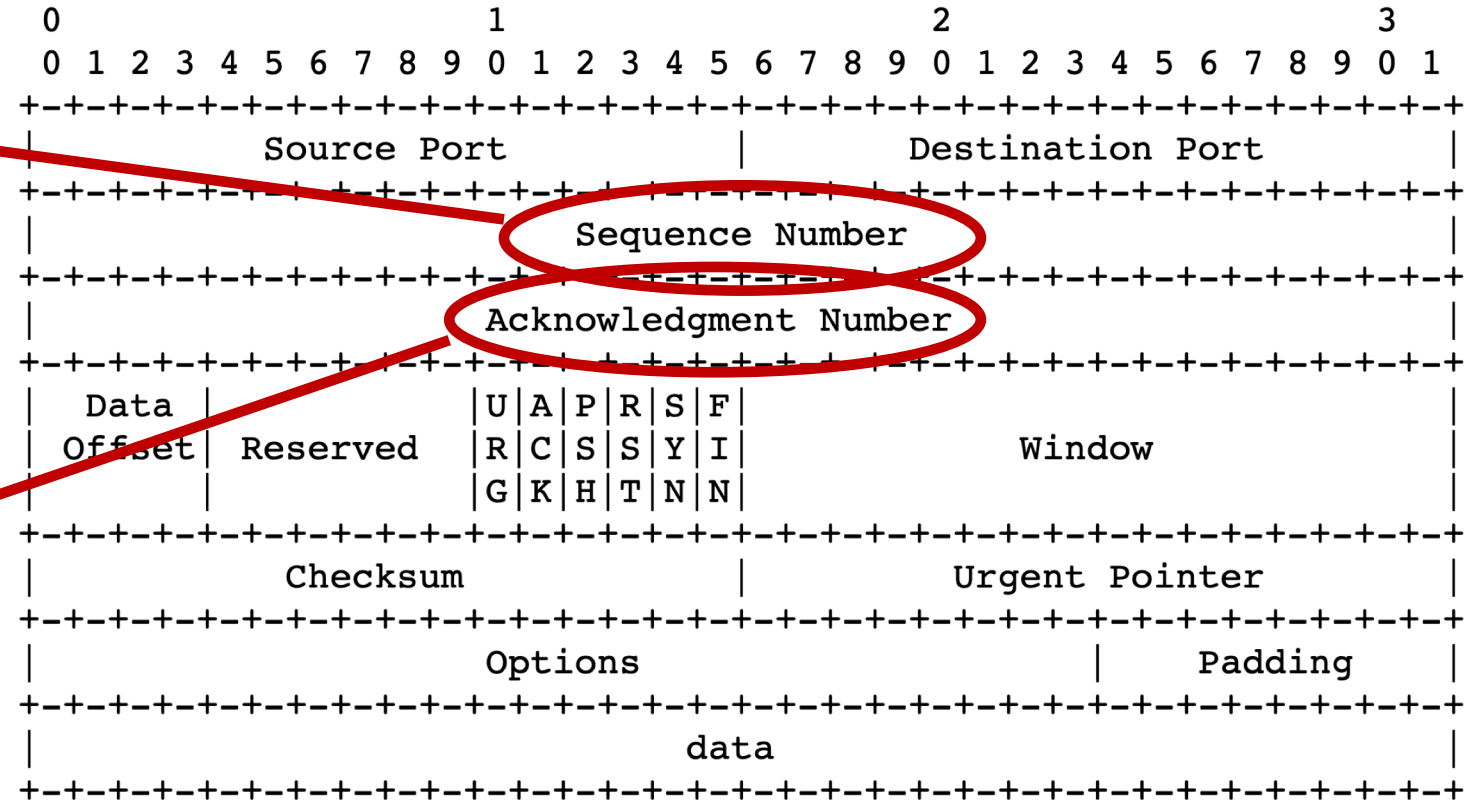
TCP header structure

Identifies data in the packet from sender's perspective

TCP uses byte seq #s

Identifies the data being ACKed from the receiver's perspective.

TCP uses next seq # that the receiver is expecting.



TCP Header Format

Note that one tick mark represents one bit position.

A TCP exchange

- A small demo

