

CS 43: Computer Networks

Reliable Transport and TCP

October 22, 2020



Transport Layer

Today

- Principles of reliability
- Class of protocols: Automatic Repeat Requests

Moving down a layer!

Application Layer

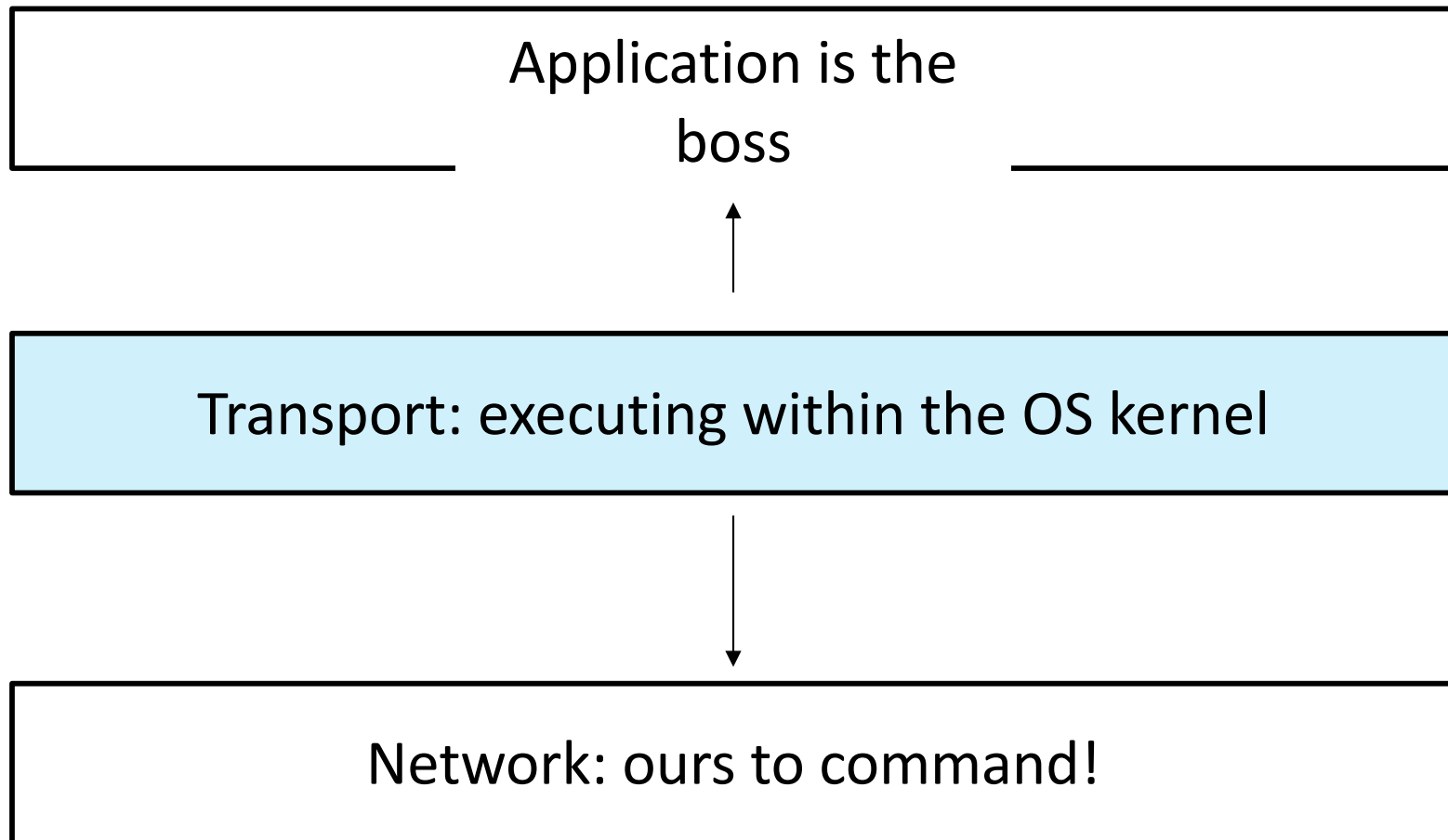
Transport: end-to-end connections, reliability

Network: routing

Link (data-link): framing, error detection

Physical: 1's and 0's/bits across a medium
(copper, the air, fiber)

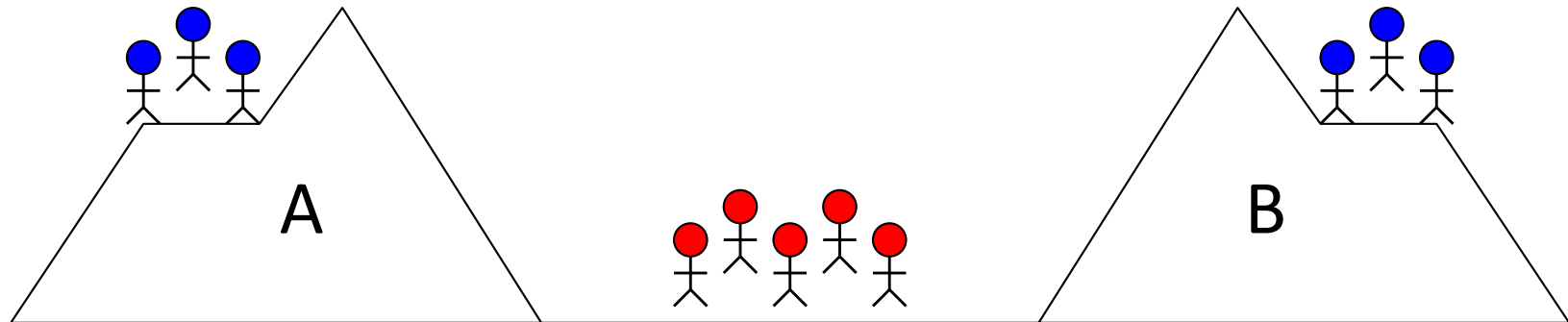
Transport Layer perspective



Today

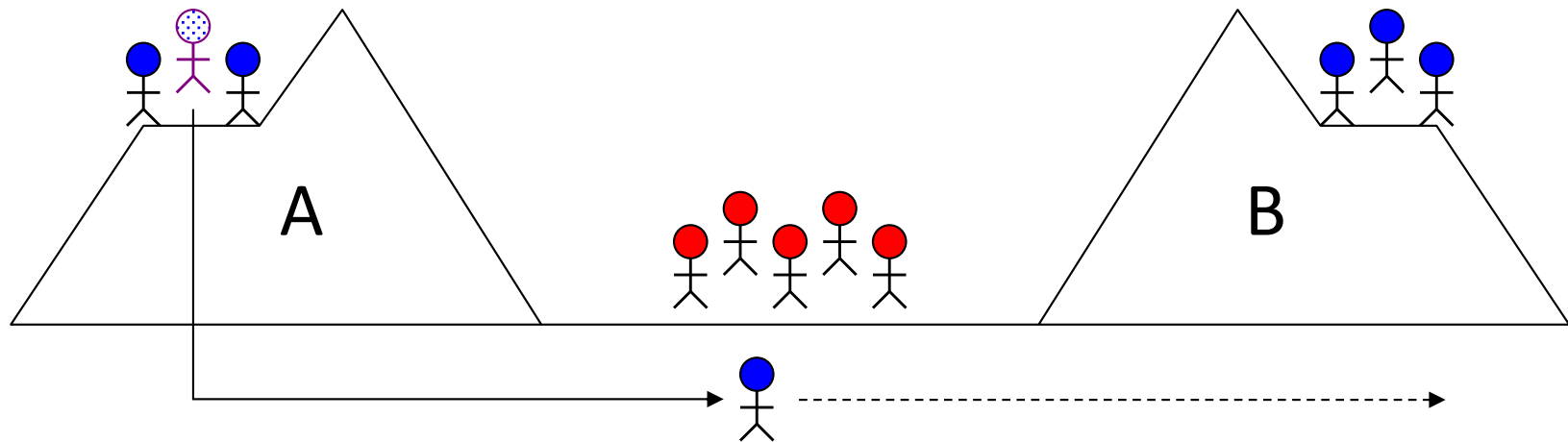
- Principles of reliability
 - The Two Generals Problem
- Automatic Repeat Requests
 - Stop and Wait
 - Timeouts and Losses
 - Pipelined Transmission

The Two Generals Problem



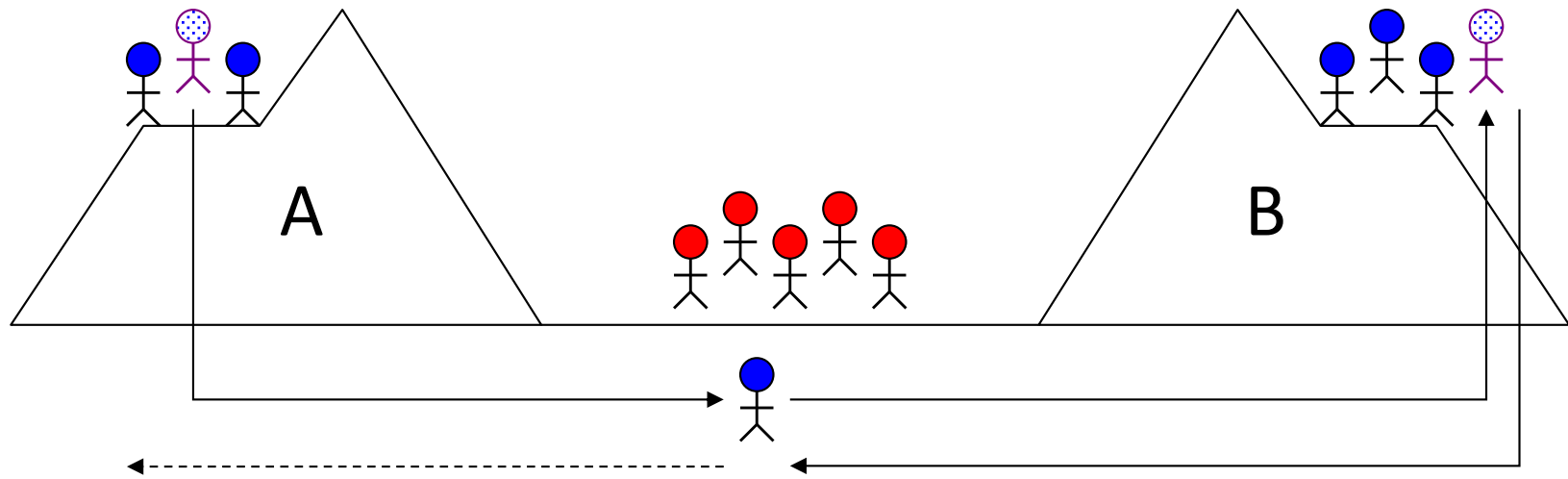
- Two army divisions (blue) surround enemy (red)
 - Each division led by a general
 - Both must agree when to simultaneously attack
 - If either side attacks alone, defeat
- Generals can only communicate via messengers
 - Messengers may get captured (unreliable channel)

The Two Generals Problem



- How to coordinate?
 - Send messenger: “Attack at dawn”
 - What if messenger doesn’t make it?

The Two Generals Problem

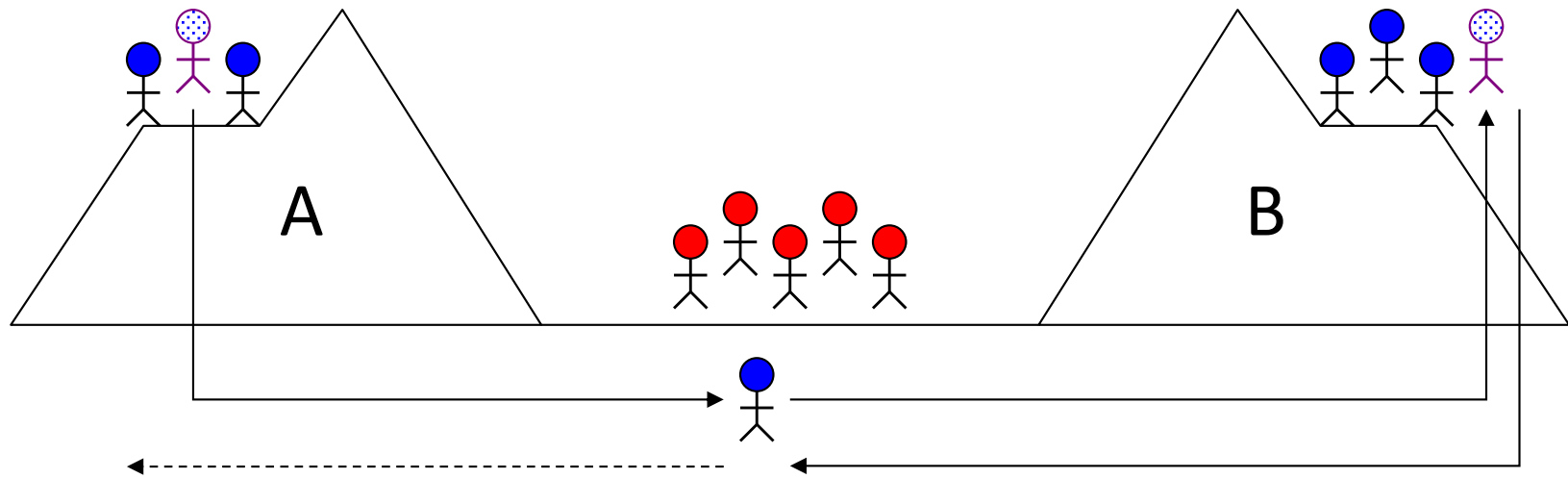


- How to be sure messenger made it?
 - Send acknowledgment: “I delivered message”

In the “two generals problem”, can the two armies reliably coordinate their attack? (using what we just discussed)

- A. Yes (explain how)
- B. No (explain why not)

The Two Generals Problem



- Result
 - Can't create perfect channel out of faulty one
 - Can only increase probability of success

Engineering

- Concerns
 - Message corruption
 - Message duplication
 - Message loss
 - Message reordering
 - Performance
- Our toolbox
 - Checksums
 - Timeouts
 - Acks & Nacks
 - Sequence numbering
 - Pipelining

Engineering

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We use these to build **Automatic Repeat Request (ARQ)** protocols.

(We'll briefly talk about alternatives at the end.)

Automatic Repeat Request (ARQ)

- Intuitively, ARQ protocols act like you would when using a cell phone with bad reception.
 - Receiver: Message garbled? Ask to repeat.
 - Sender: Didn't hear a response? Speak again.
- Refer to book for building state machines.
 - We'll look at TCP's states soon

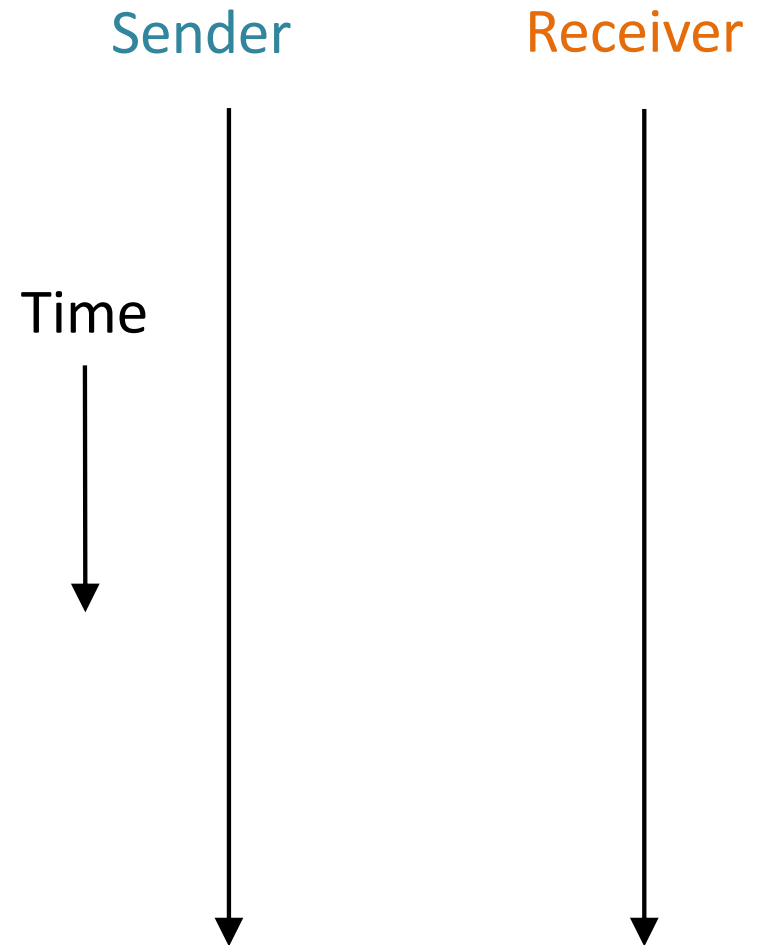
ARQ Broad Classifications

1. Stop-and-wait

Stop and Wait

We have:

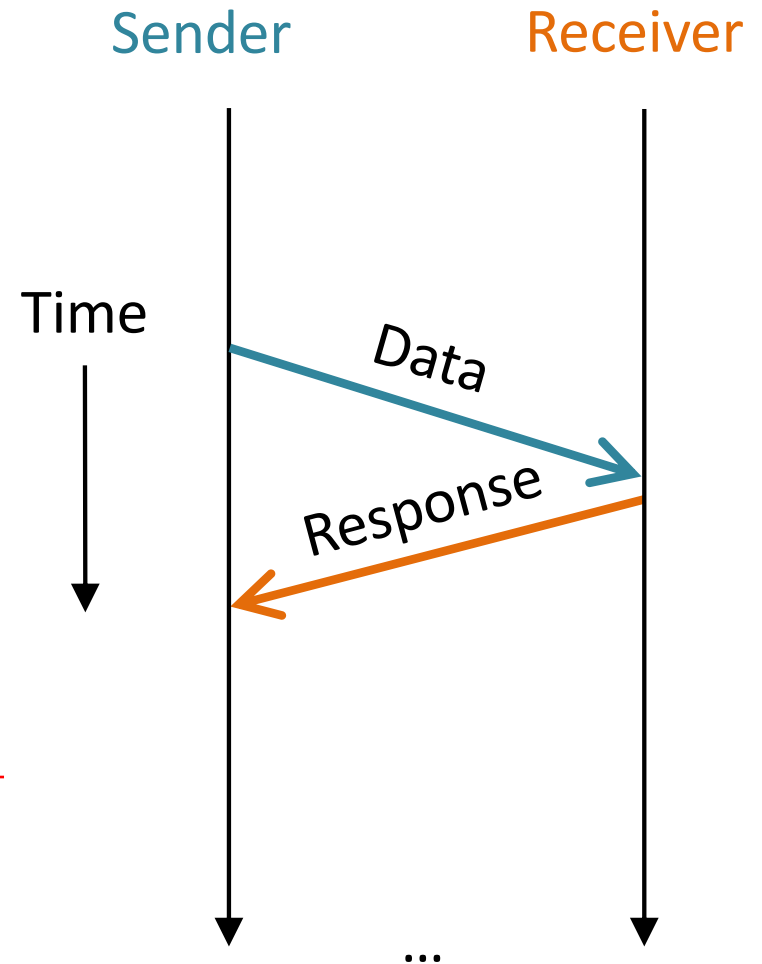
- a sender
- a receiver
- time: represented by downwards arrow



Stop and Wait

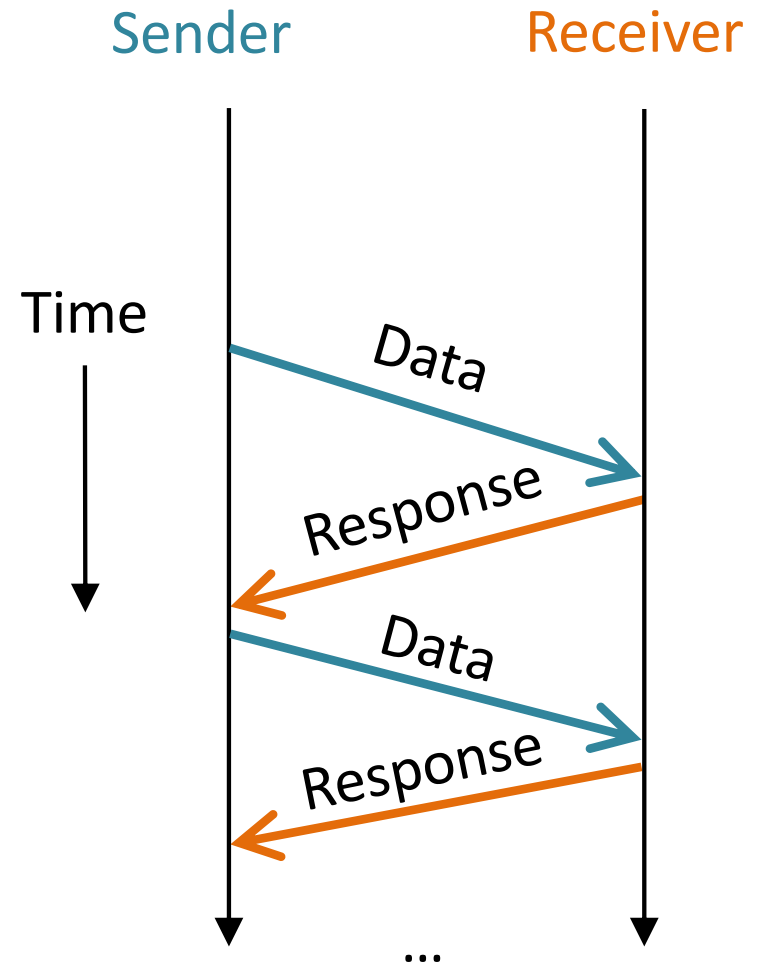
Sender sends data and waits till they get the response message from the receiver.

Buffer data, and don't send till response received



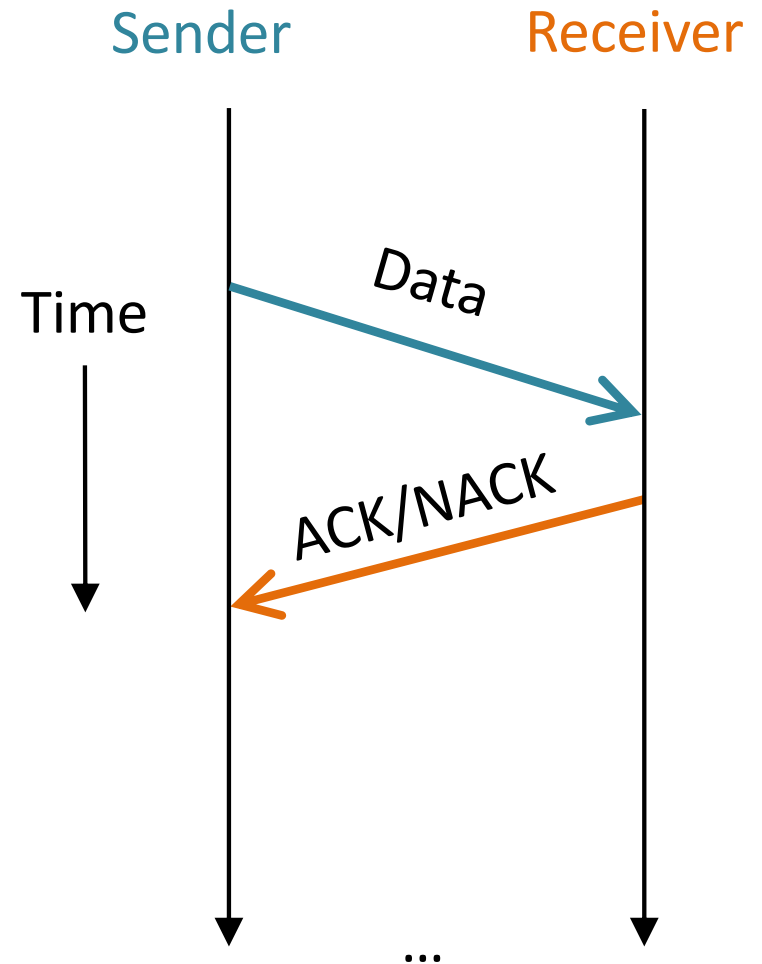
Stop and Wait

- Up next: concrete problems and mechanisms to solve them.
- These mechanisms will build upon each other
- Questions?



Corruption?

- Error detection mechanism: checksum
 - Data good – receiver sends back ACK
 - Data corrupt – receiver sends back NACK

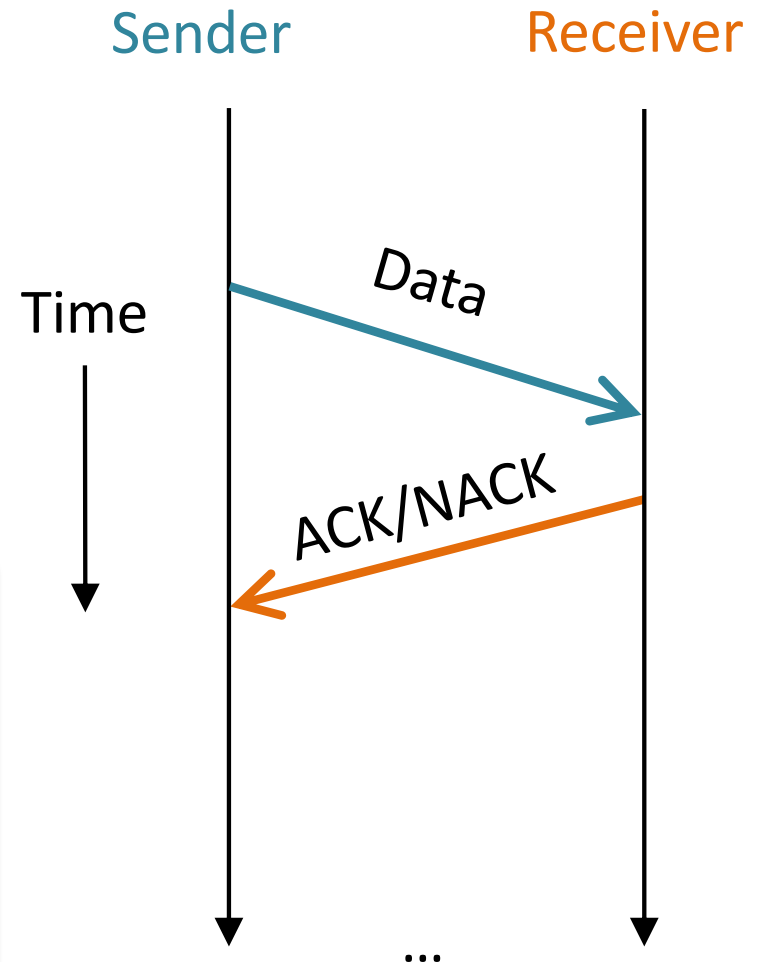


Could we do this with just ACKs or just NACKs?

Error detection mechanism:
checksum

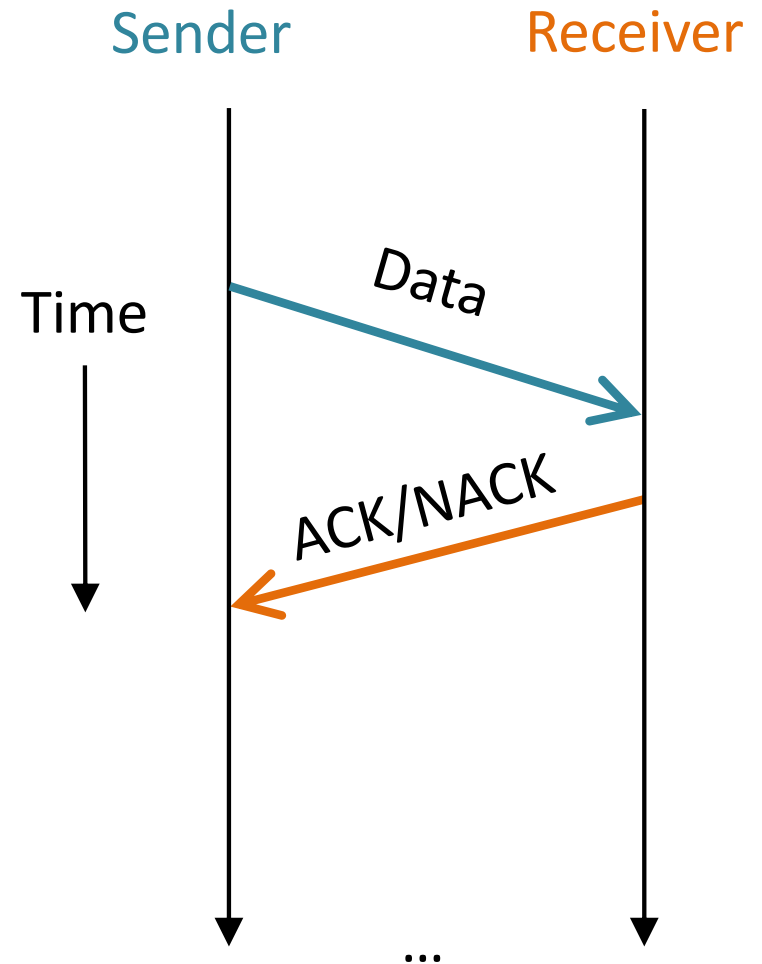
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- B. Yes, we could do without one of them, but we'd need some other mechanism.
- C. Yes, we could get by without one of them.

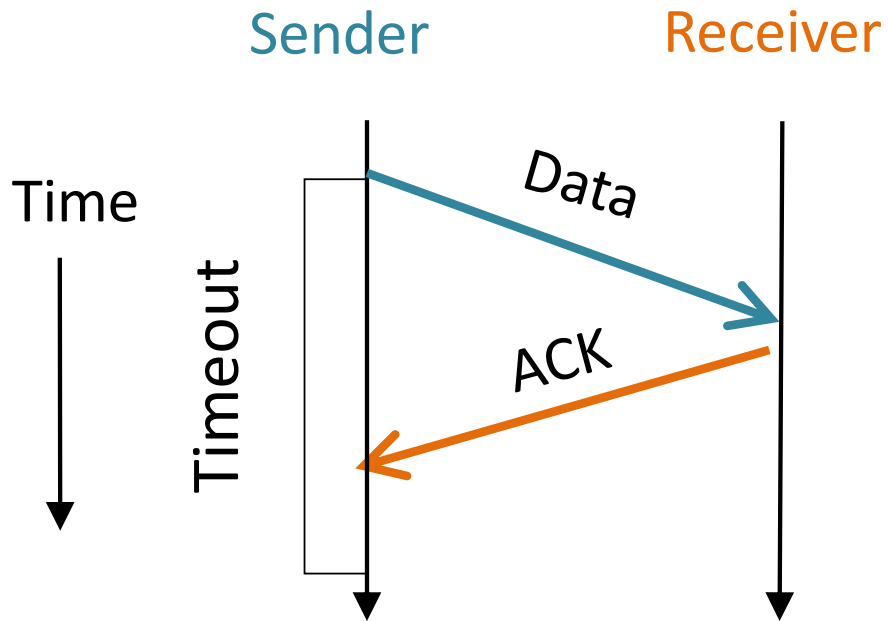


Could we do this with just ACKs or just NACKs?

- **With only ACK**, we could get by with a timeout.
 - **With only NACK**, we couldn't advance (no good).
- A. No, we need them both.
 - B. Yes, we could do without one of them, but we'd need some other mechanism.
 - C. Yes, we could get by without one of them.

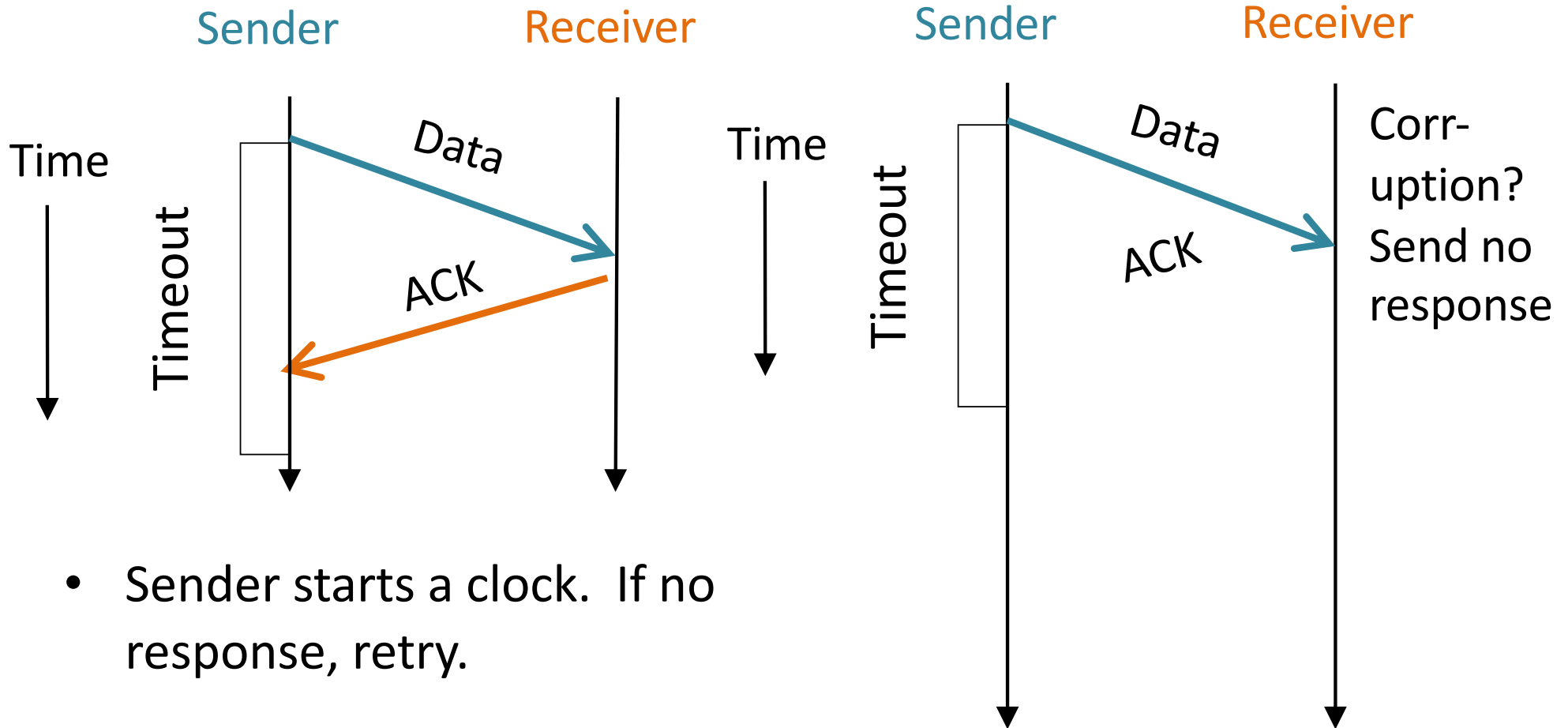


Timeouts and Losses



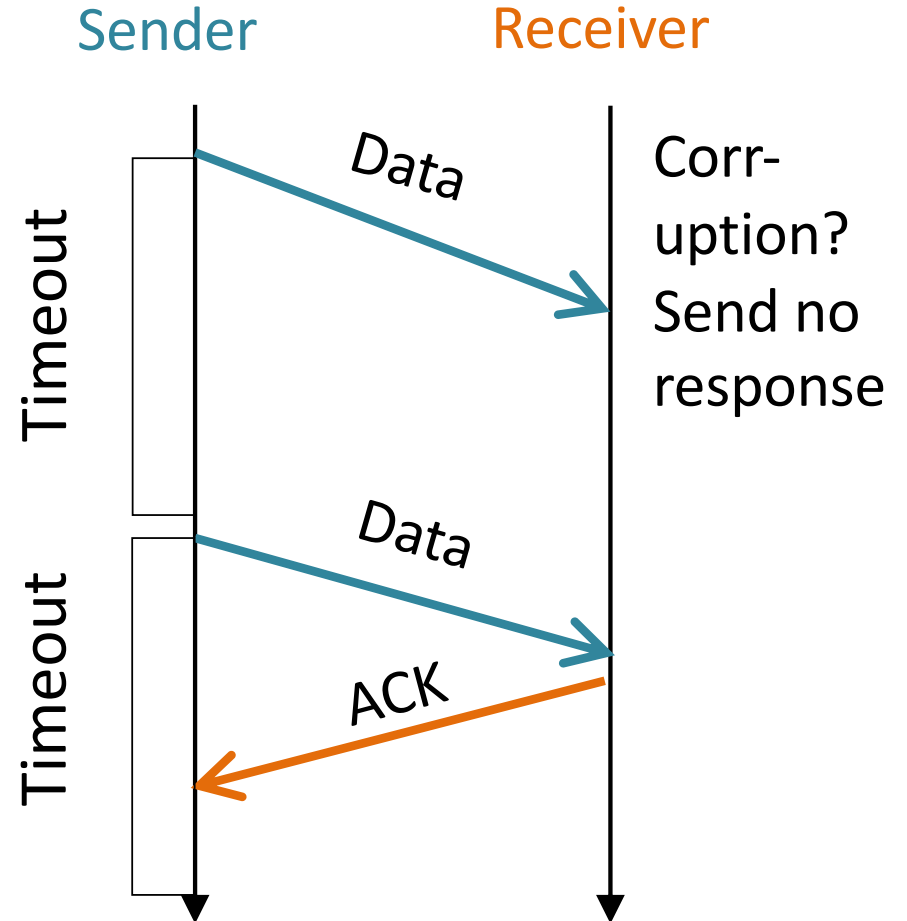
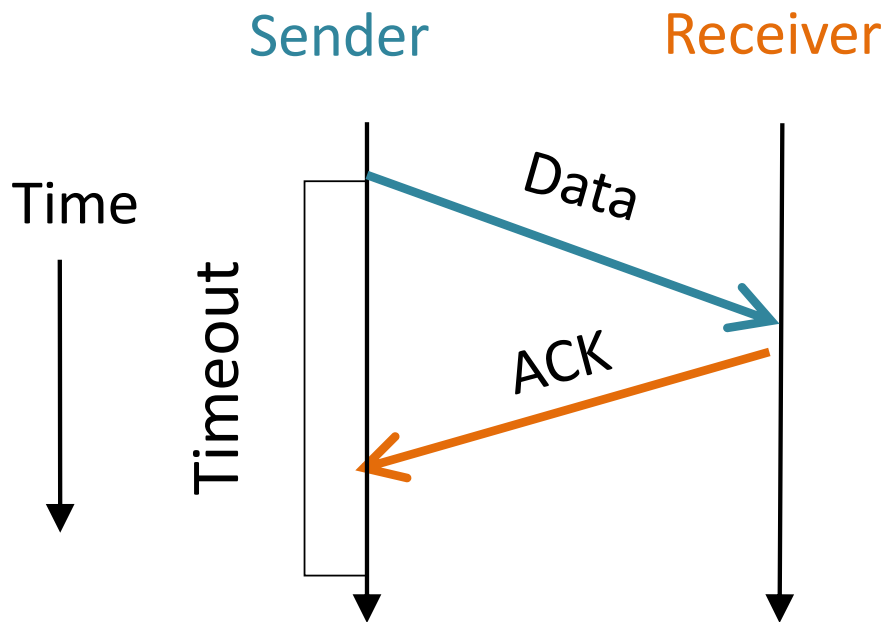
- Sender starts a clock. If no response, retry.

Timeouts and Losses



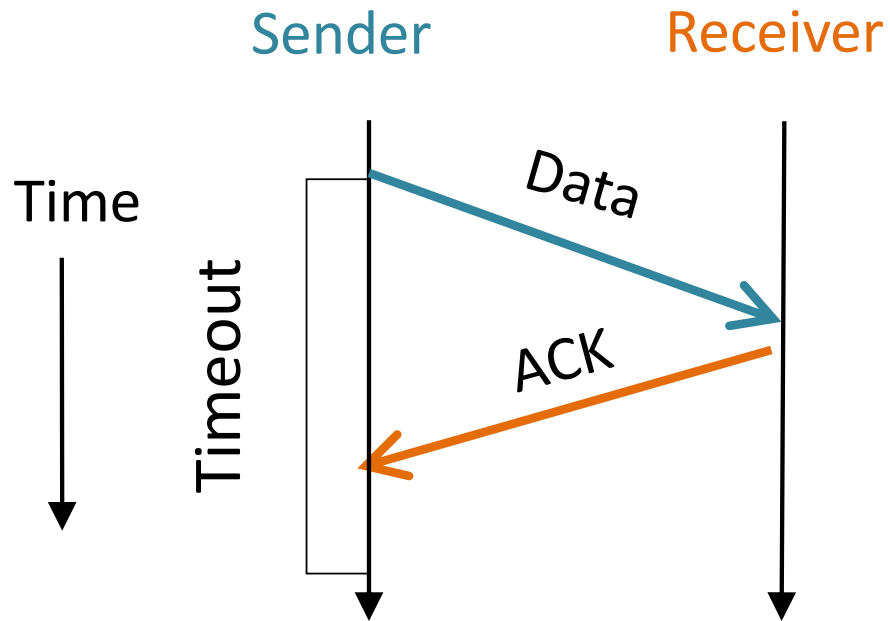
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Timeouts and Losses



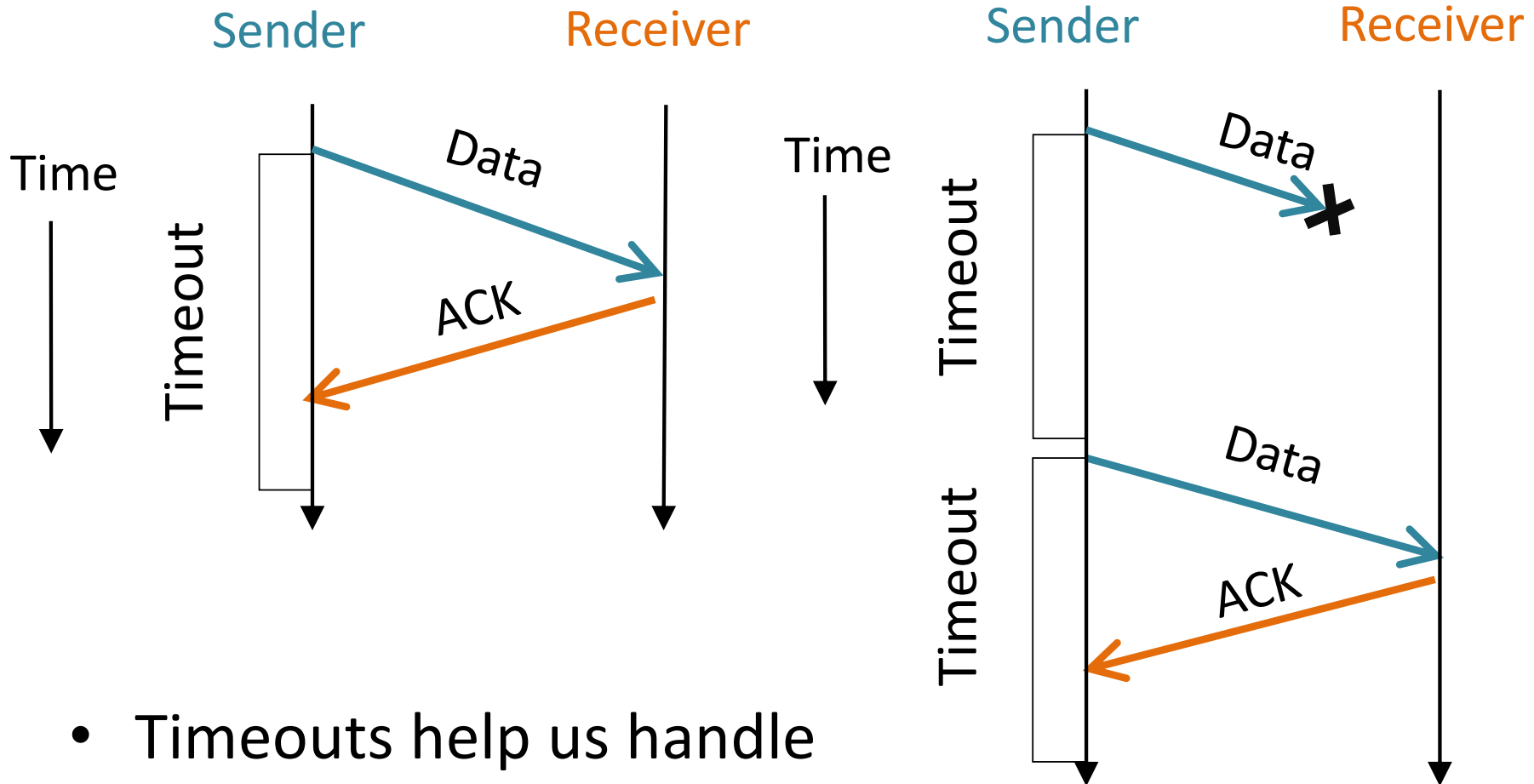
- Sender starts a clock. If no response, retry.
- Probably not a great idea for handling corruption, but it works.

Timeouts and Losses



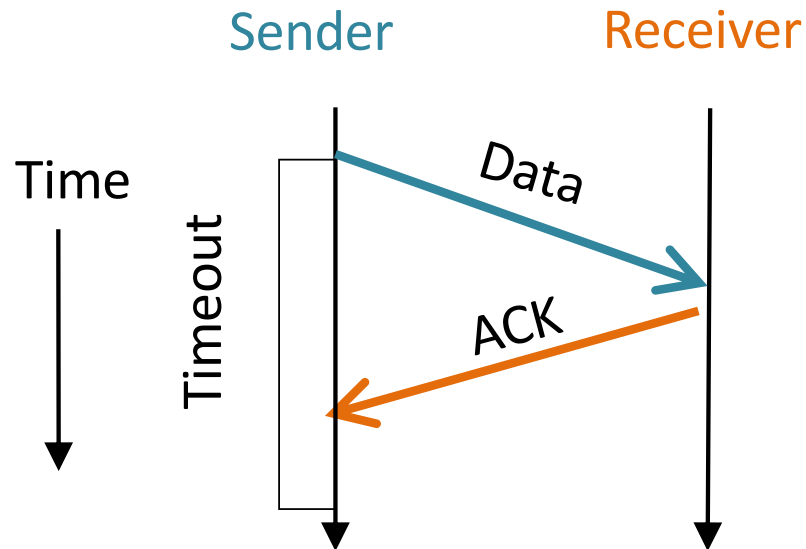
- Timeouts help us handle message losses too!

Timeouts and Losses



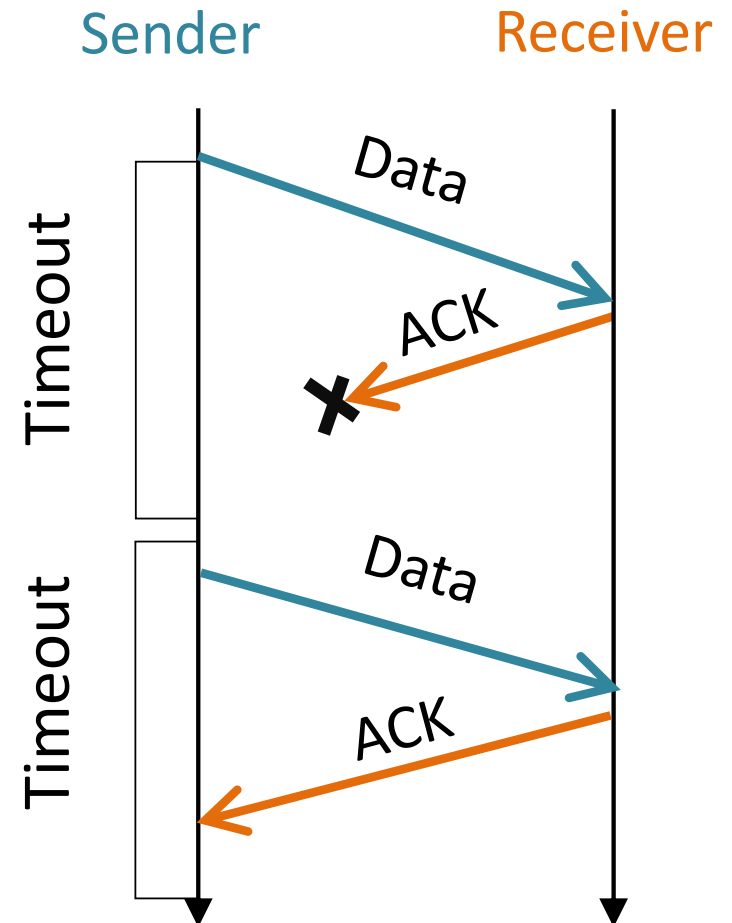
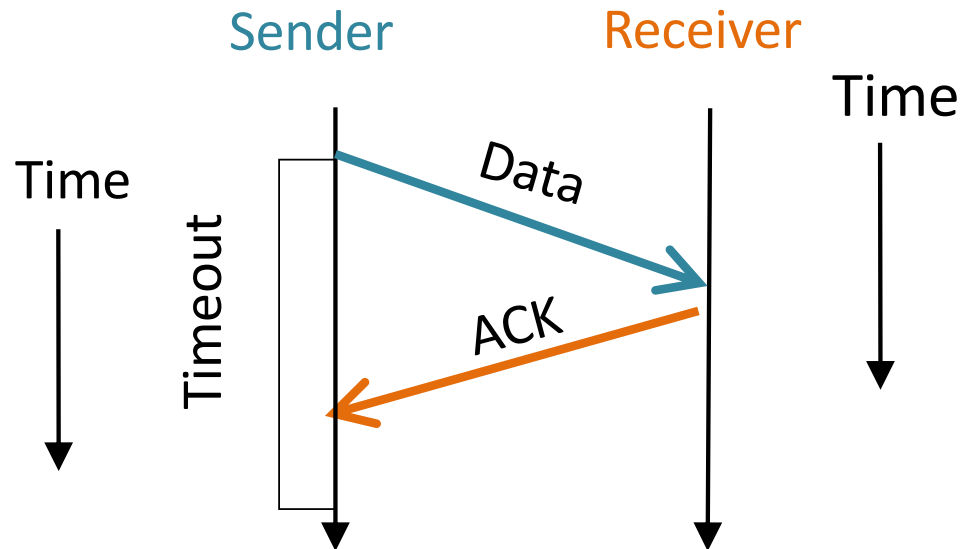
- Timeouts help us handle message losses too!

Adding timeouts might create new problems for us to worry about. How many?
Examples?



- A. No new problems (why not?)
- B. One new problem (which is..)
- C. Two new problems (which are..)
- D. More than two new problems (which are..)

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Sequence Numbering

Sender

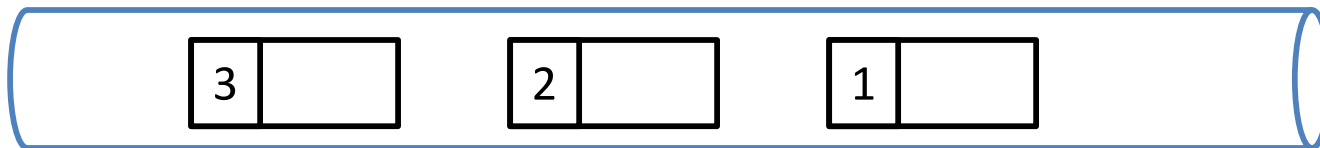
- Add a monotonically increasing label to each msg

Receiver

- Ignore messages with numbers we've seen before
- When pipelining (a few slides from now)
 - Detect gaps in the sequence (e.g., 1,2,4,5)

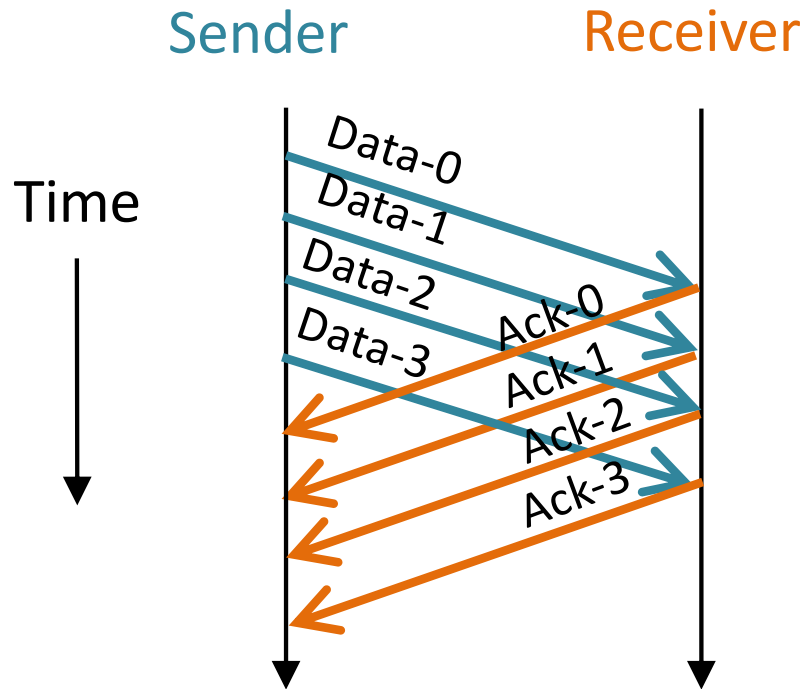


Sender



Receiver

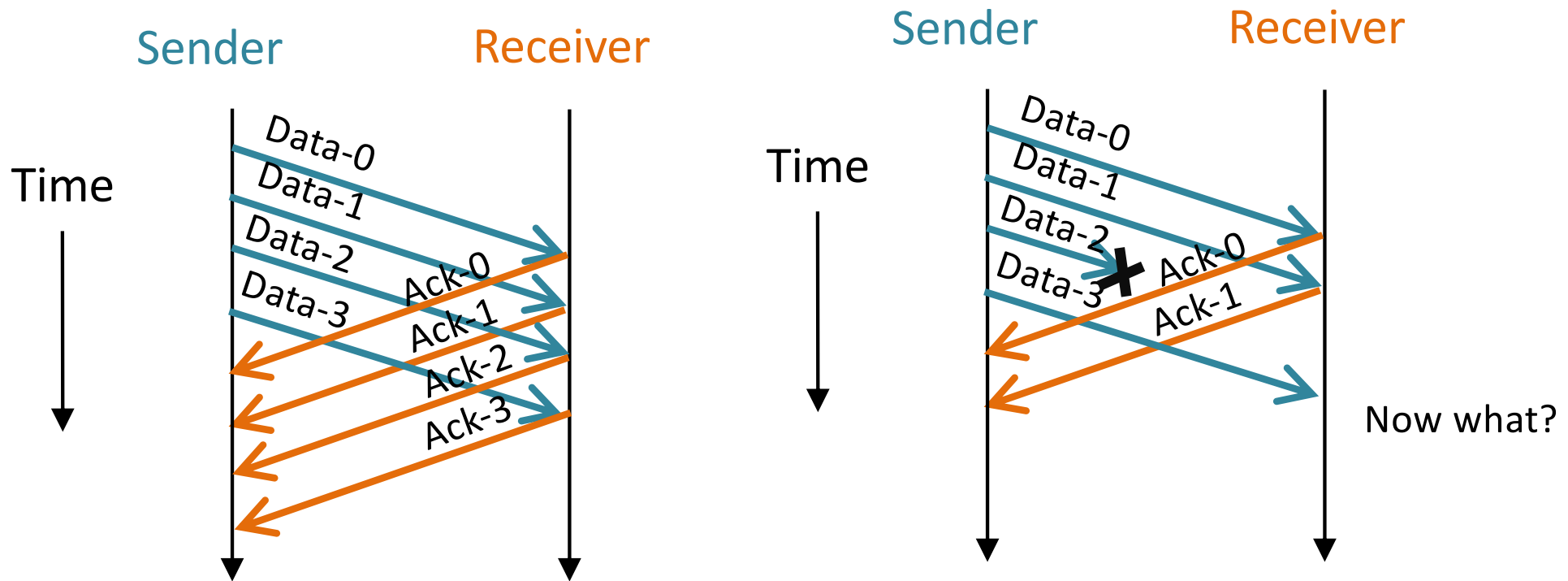
Pipelined Transmission



Keep multiple segments “in flight”

- Allows sender to make efficient use of the link
- Sequence numbers ensure receiver can distinguish segments

Pipelined Transmission



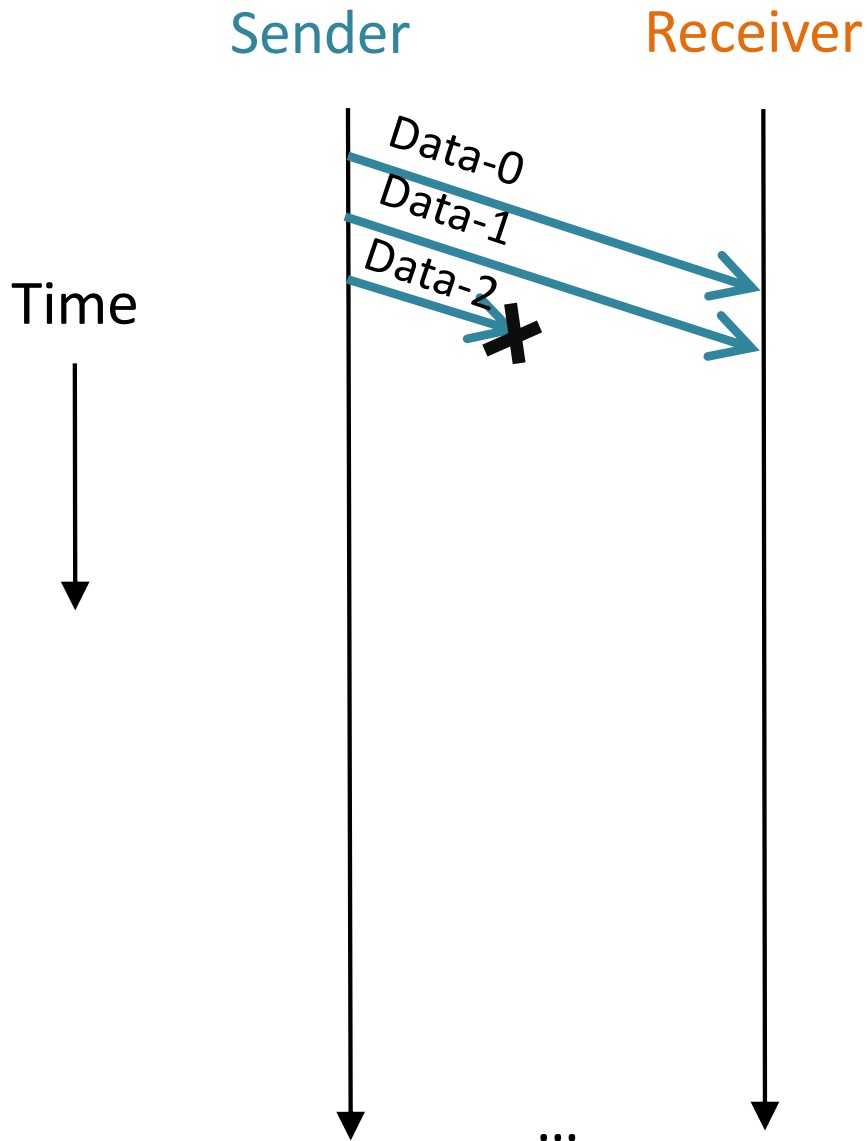
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ARQ Broad Classifications

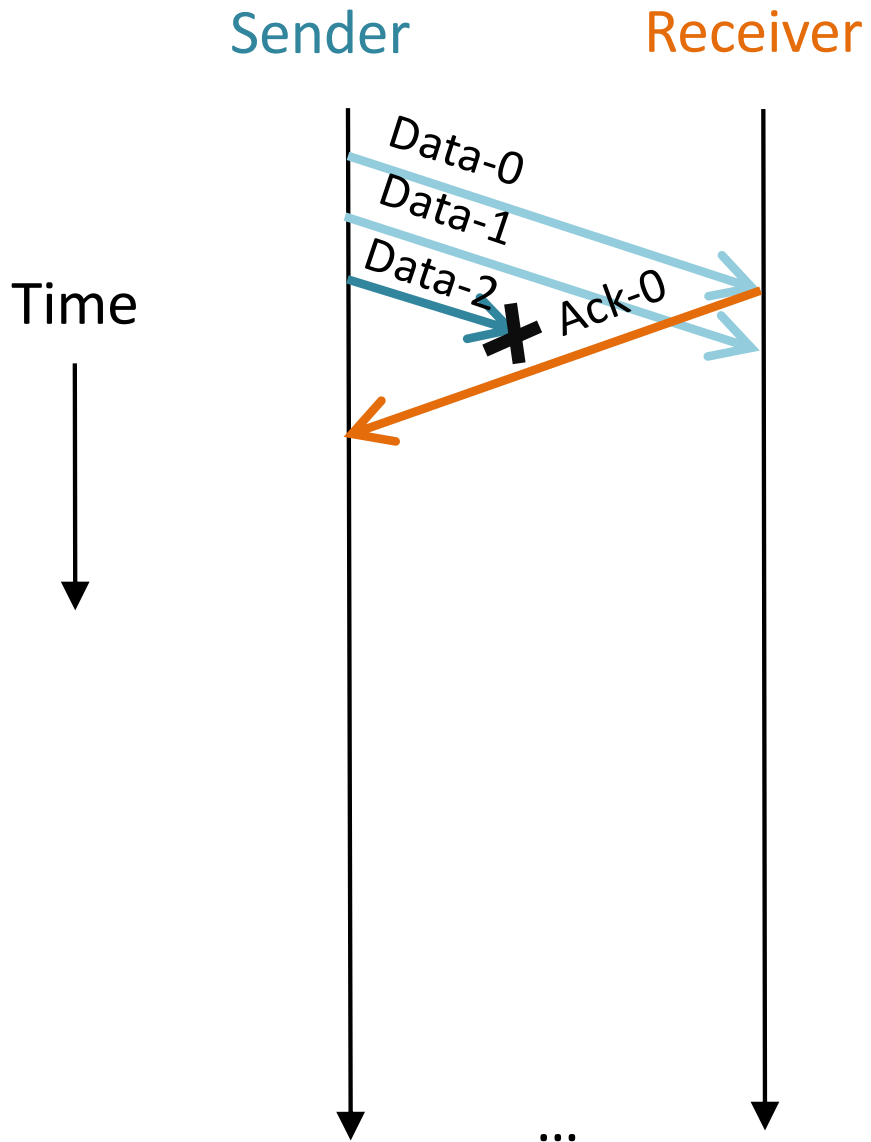
1. Stop-and-wait
2. Go-back-N

Go-Back-N

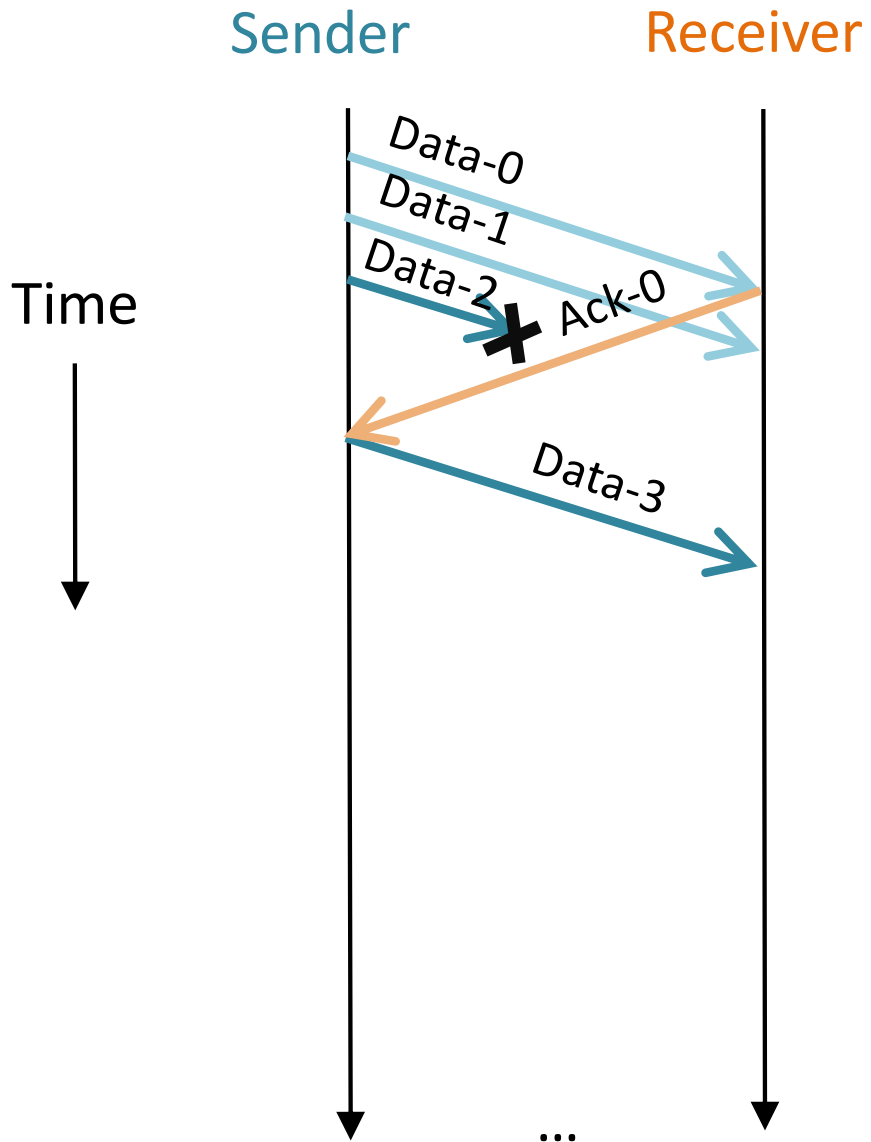


- Retransmit from point of loss
 - Segments between loss event and retransmission are ignored
 - “Go-back-N” if a timeout event occurs

Go-Back-N



Go-Back-N

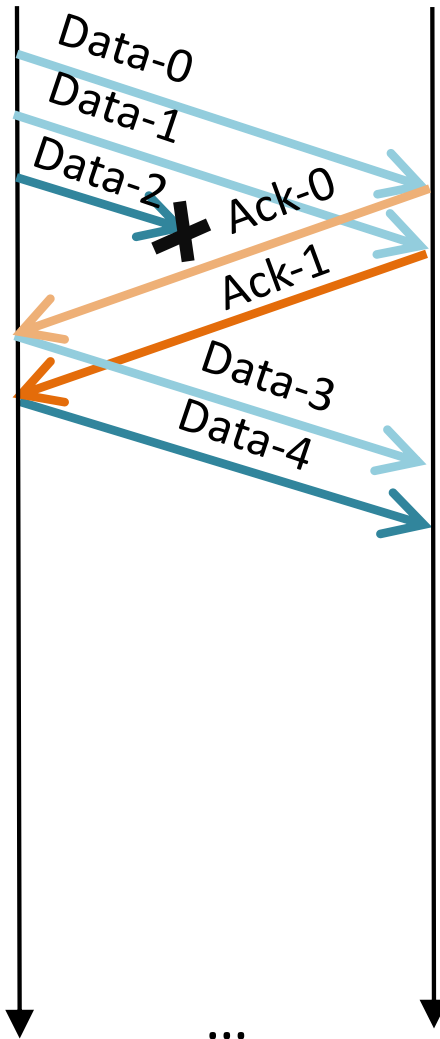


Go-Back-N

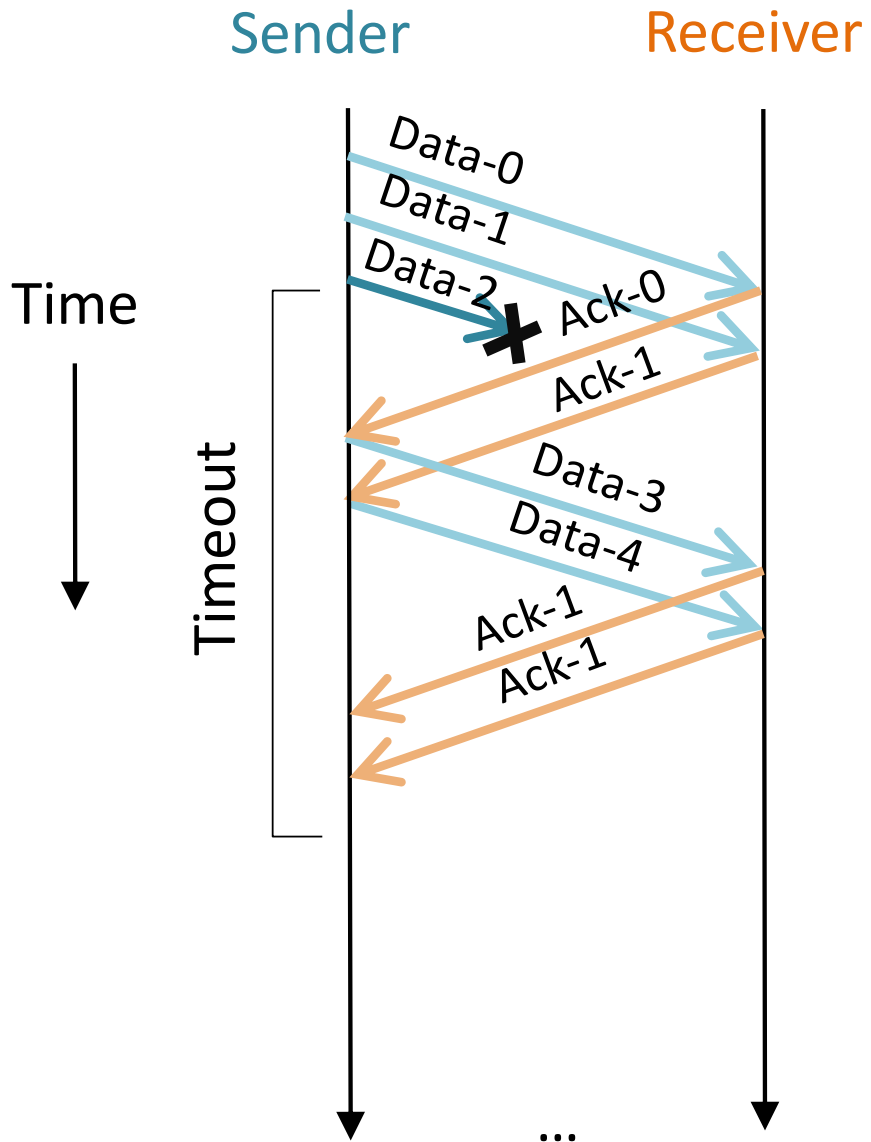
Sender

Receiver

Time



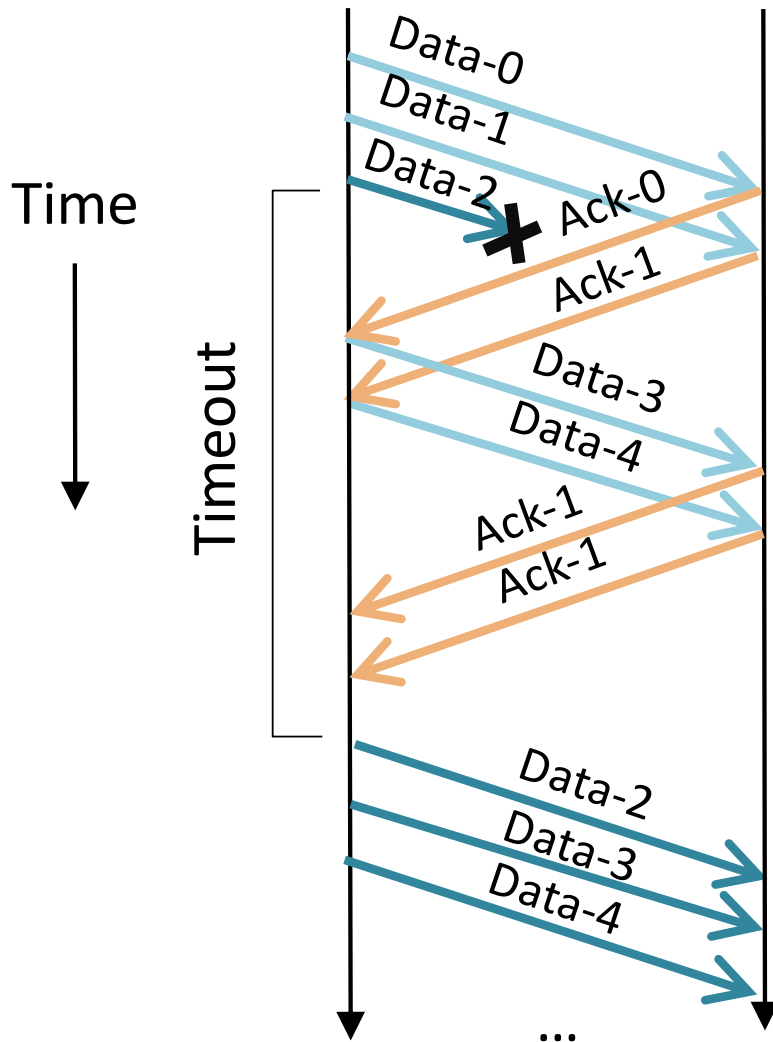
Go-Back-N



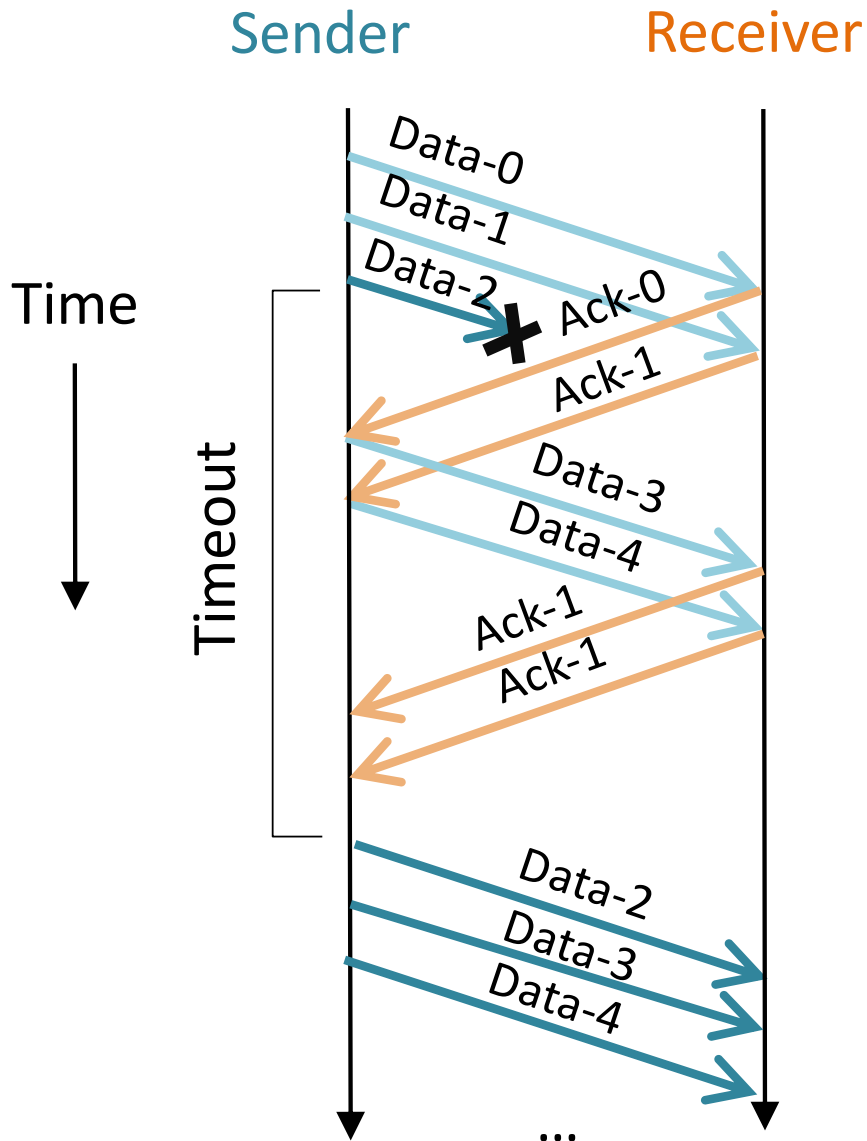
Go-Back-N

Sender

Receiver

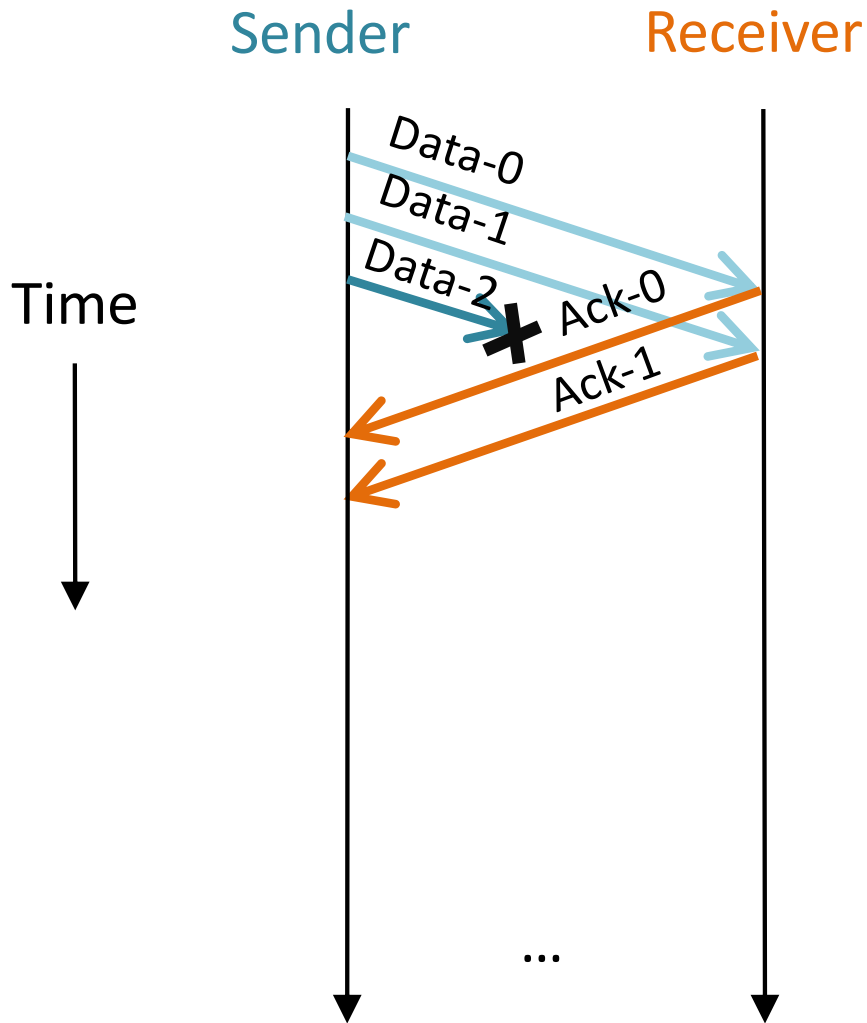


Go-Back-N



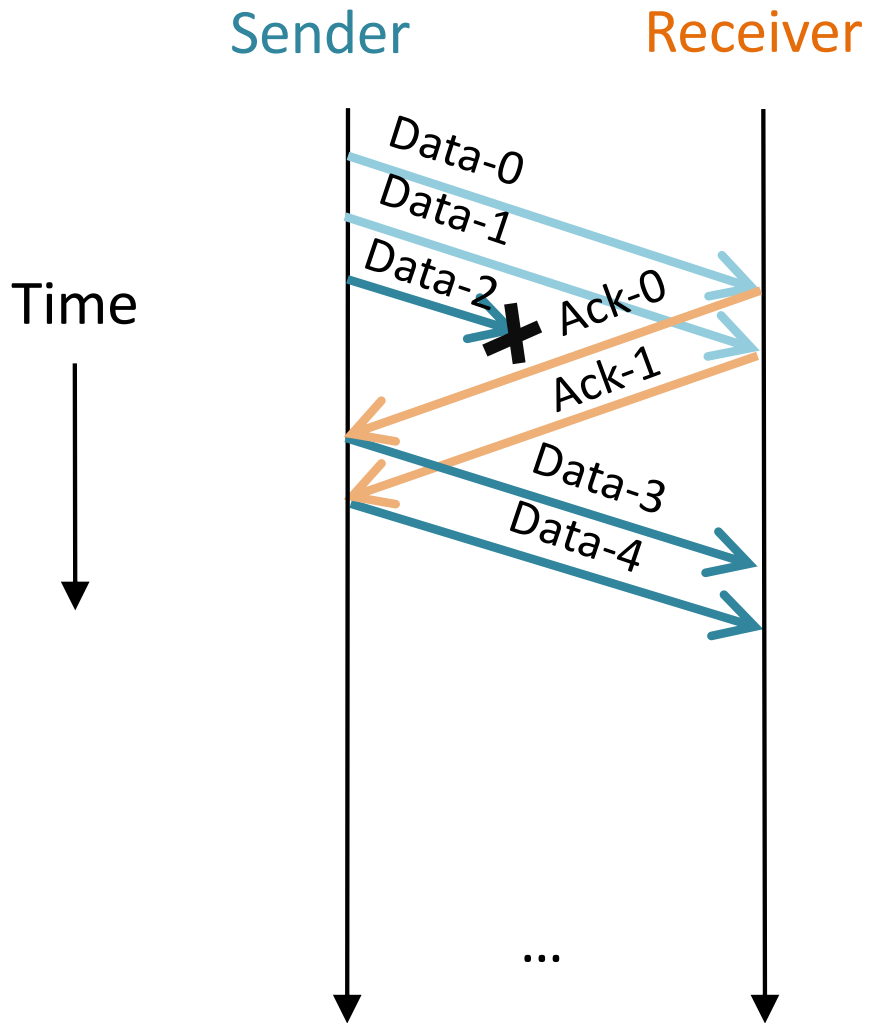
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Selective Repeat

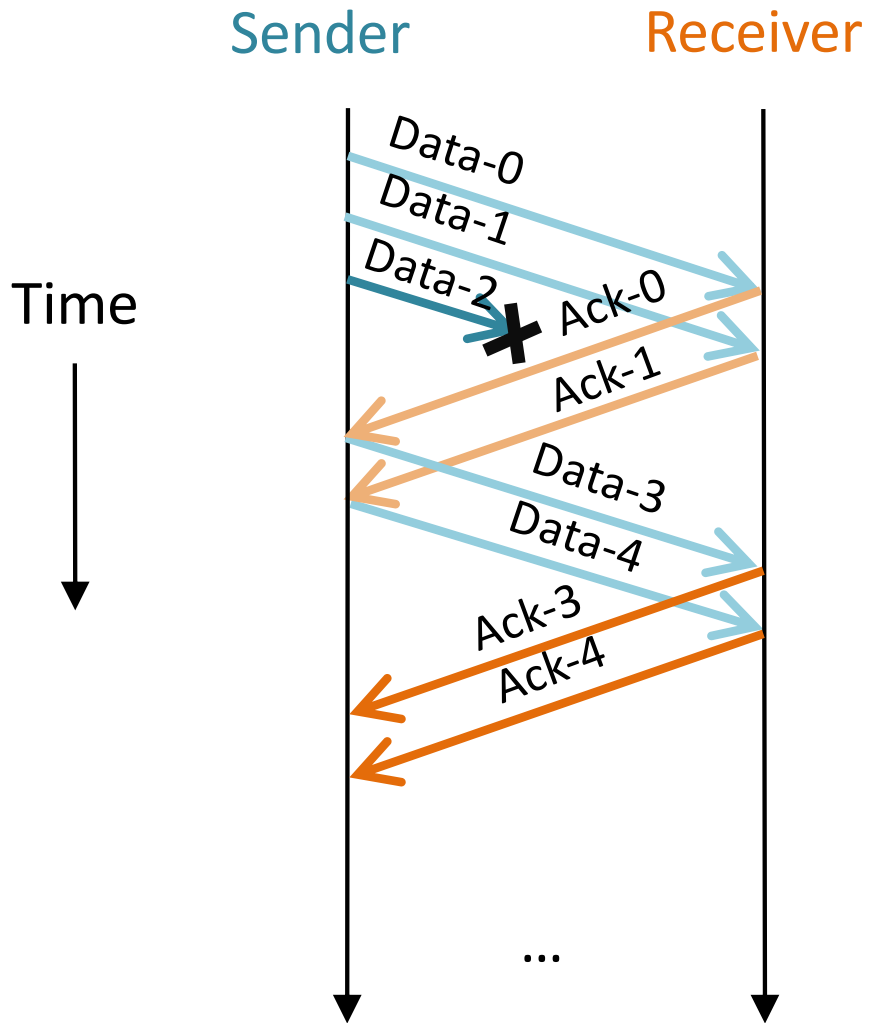


- Receiver ACKs each segment individually (not cumulative)
- Sender only resends those not ACKed

Selective Repeat



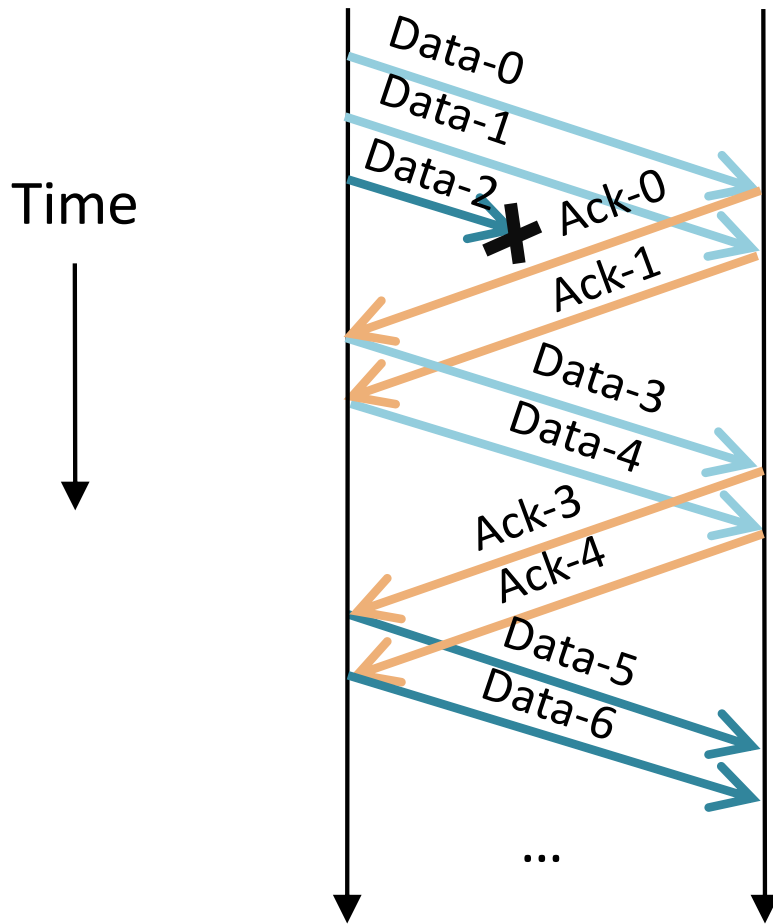
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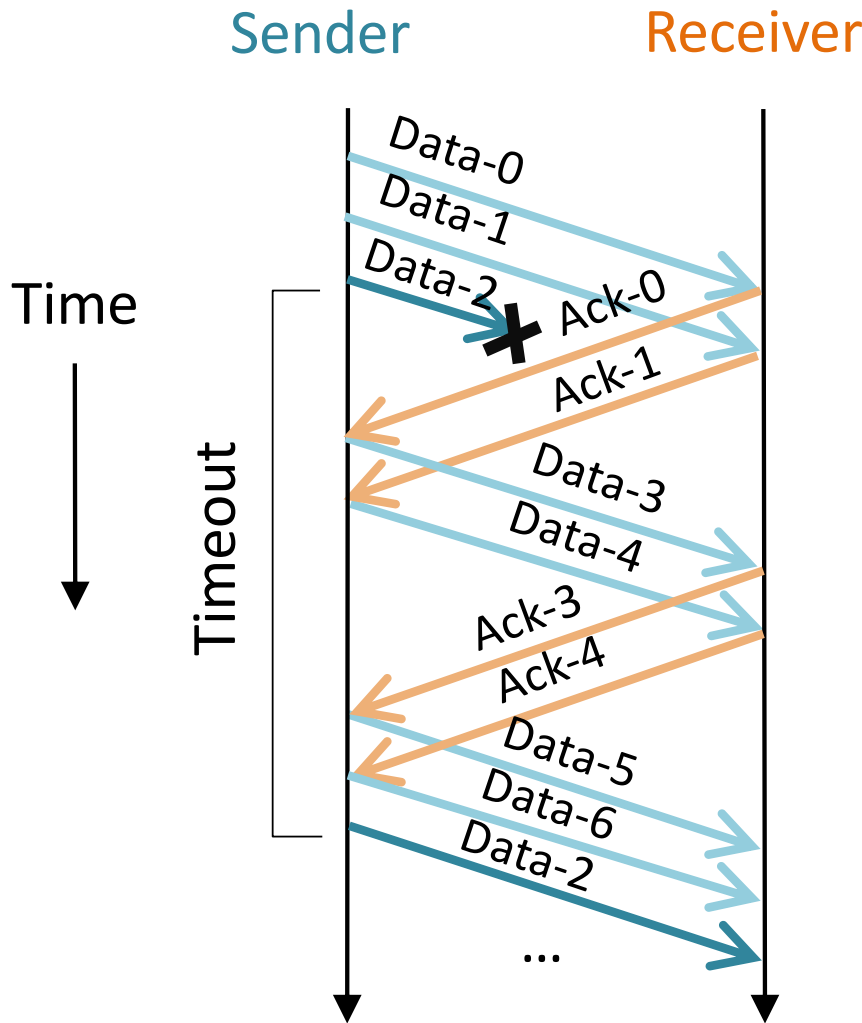
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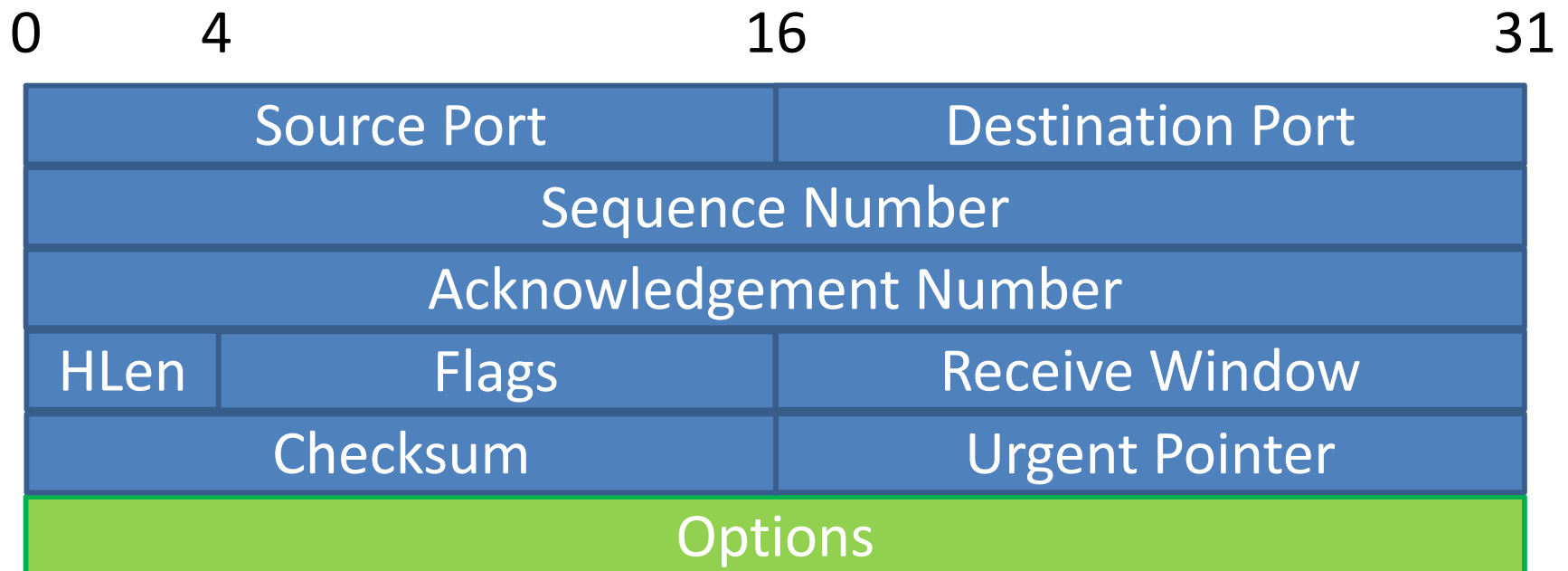
ARQ Alternatives

- Can't afford the RTT's or timeouts?
- When?
 - Broadcasting, with lots of receivers
 - Very lossy or long-delay channels (e.g., space)
- Use redundancy – send more data
 - Simple form: send the same message N times
 - More efficient: use “erasure coding”
 - For example, encode your data in 10 pieces such that the receiver can piece it together with any subset of size 8.

Transmission Control Protocol

Reliable, in-order, bi-directional byte streams

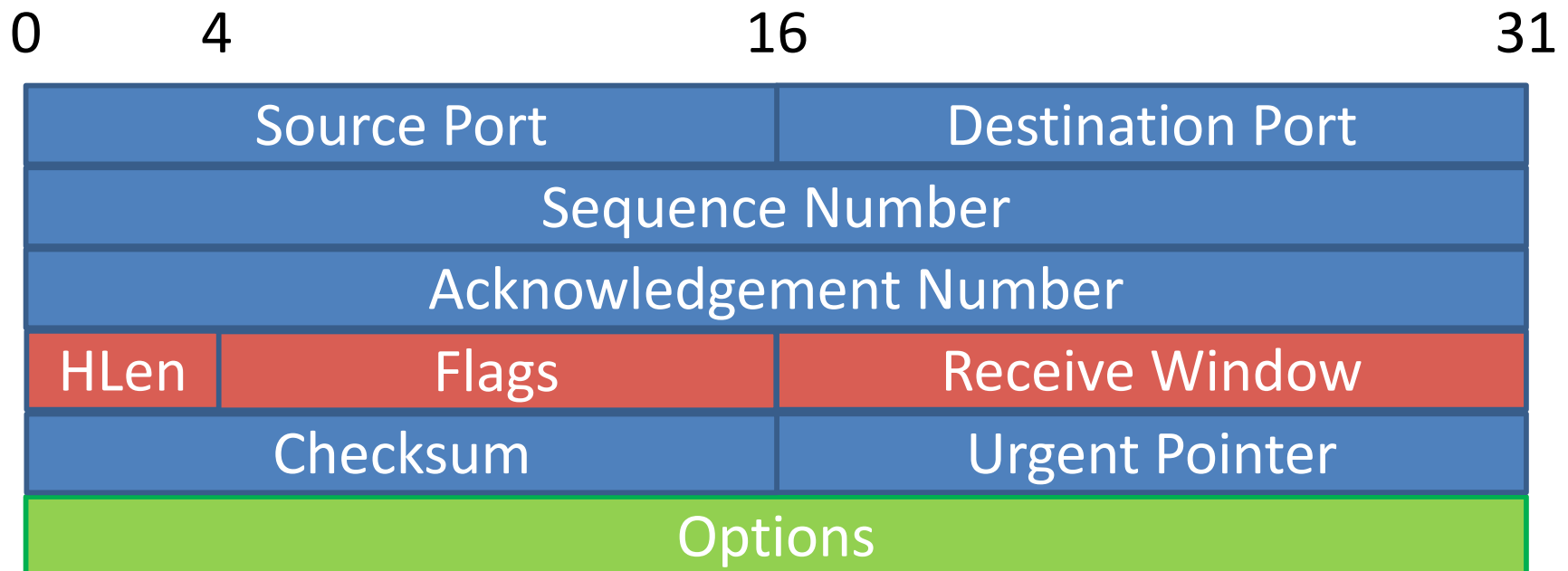
- Port numbers for demultiplexing
- Flow control
- Congestion control, approximate fairness



Transmission Control Protocol

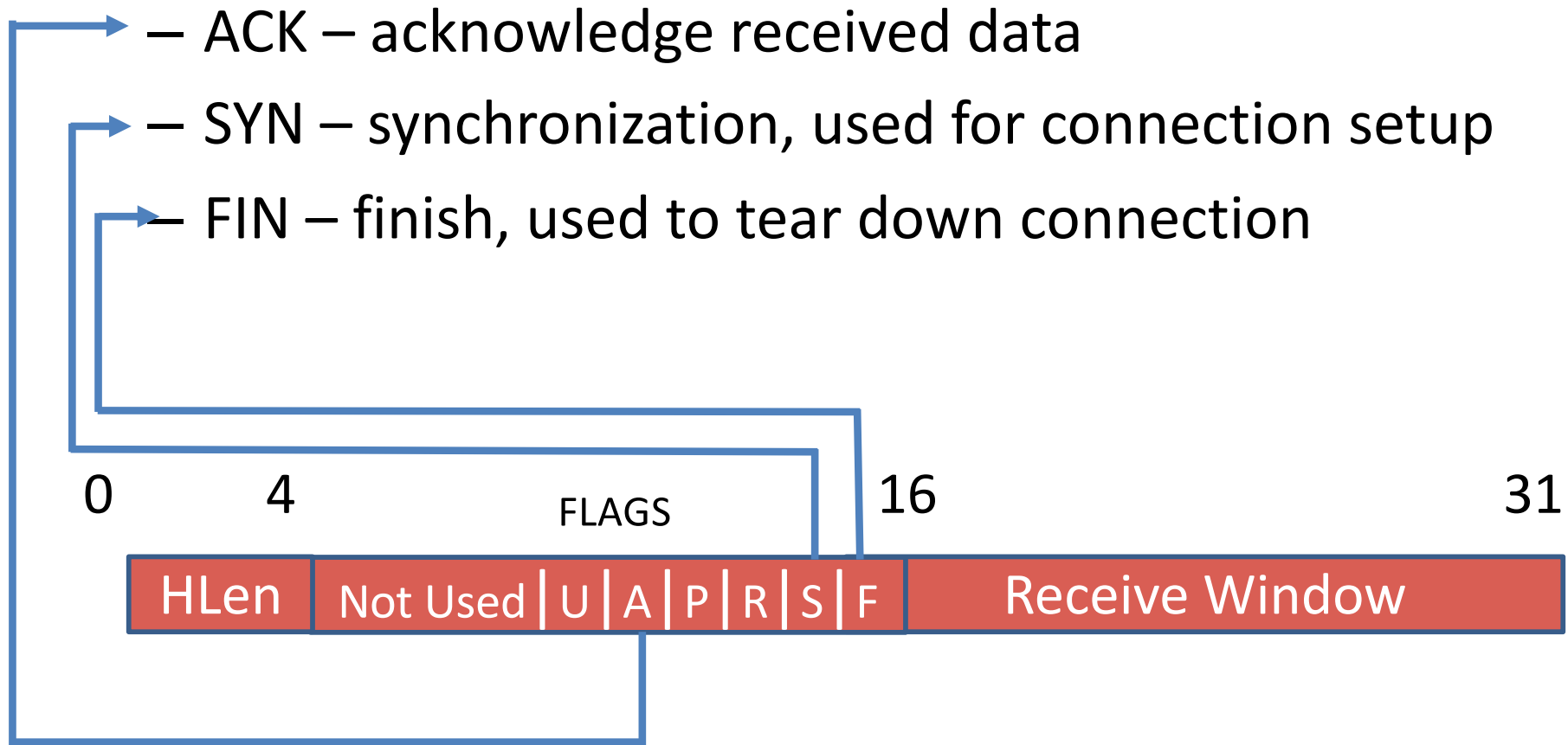
Reliable, in-order, bi-directional byte streams

- Port numbers for demultiplexing
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Transmission Control Protocol

- Important TCP flags (1 bit each)
 - ACK – acknowledge received data
 - SYN – synchronization, used for connection setup
 - FIN – finish, used to tear down connection



Practical Reliability Questions

- What does connection establishment look like?
- How do we choose sequence numbers?
- How do the sender and receiver keep track of outstanding pipelined segments?
- How should we choose timeout values?
- How many segments should be pipelined?

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A connection...

1. Requires stored state at two hosts.
2. Requires stored state within the network.
3. Establishes a path between two hosts.

- A. 1
- B. 1 & 3
- C. 1, 2 & 3
- D. 2
- E. 2 & 3

A connection...

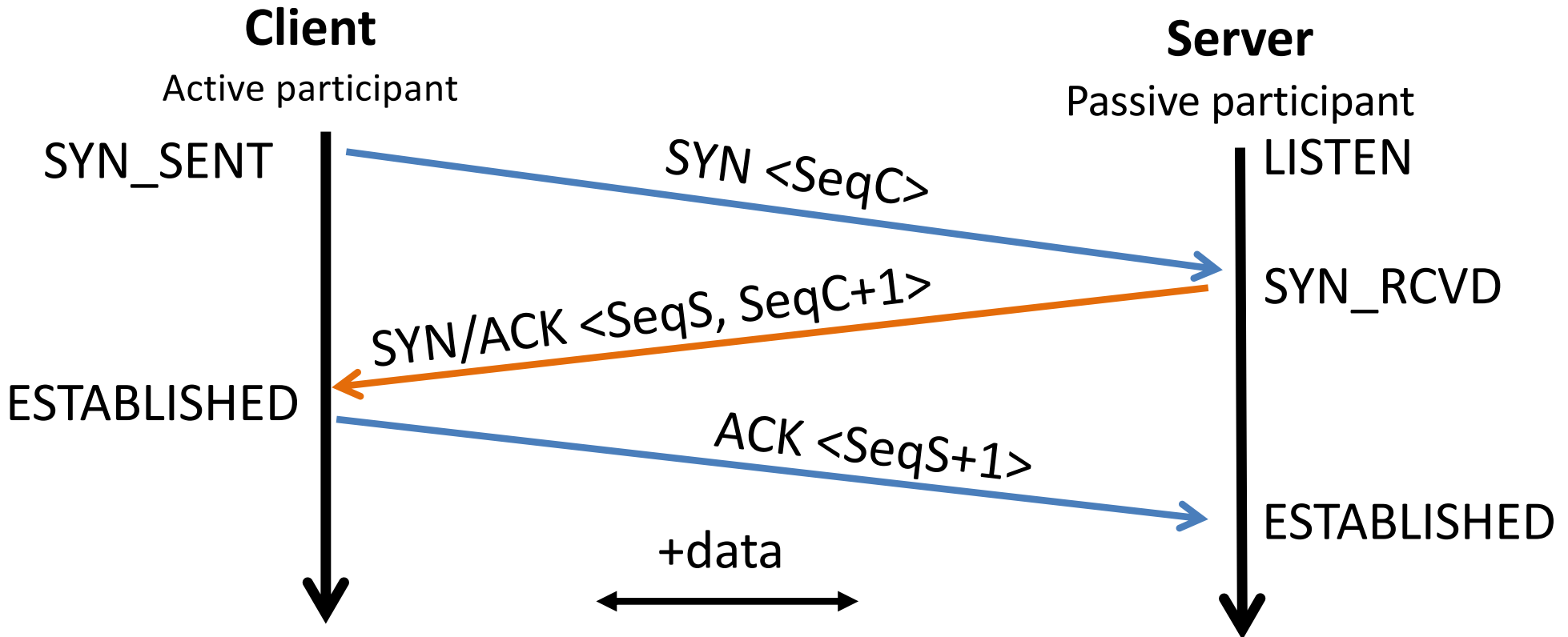
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Connections

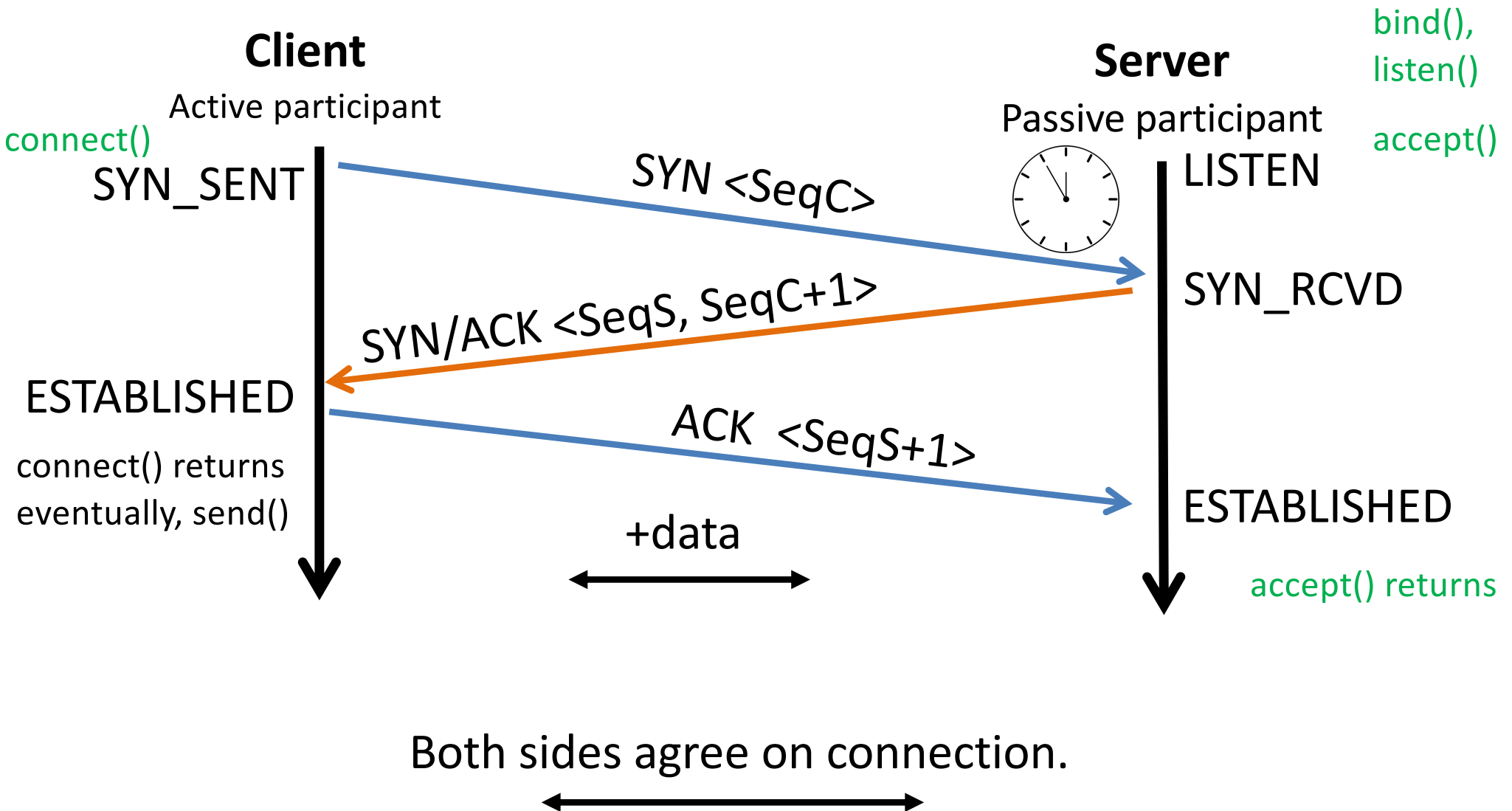
- In TCP, hosts must establish a connection prior to communicating.
- Exchange initial protocol state.
 - sequence #s to use.
 - maximum segment size
 - Initial window sizes, etc. (several parameters)

Three Way Handshake

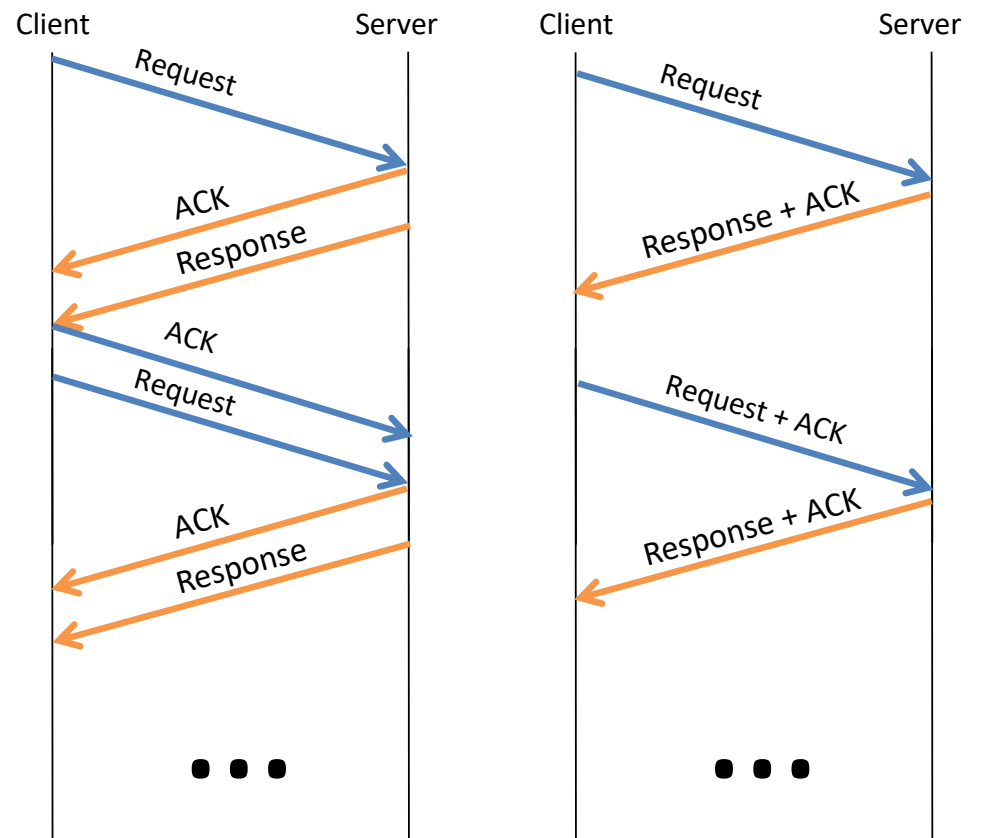


- Each side:
 - Notifies the other of starting sequence number
 - ACKs the other side's starting sequence number

Three Way Handshake



Piggybacking



Without
Piggybacking

With
Piggybacking

Initiator/Receiver

- Assumed distinct “sender” and “receiver” roles
- In reality, **usually both sides of a connection send some data**
- request/response is a common pattern

Initiator

Active participant

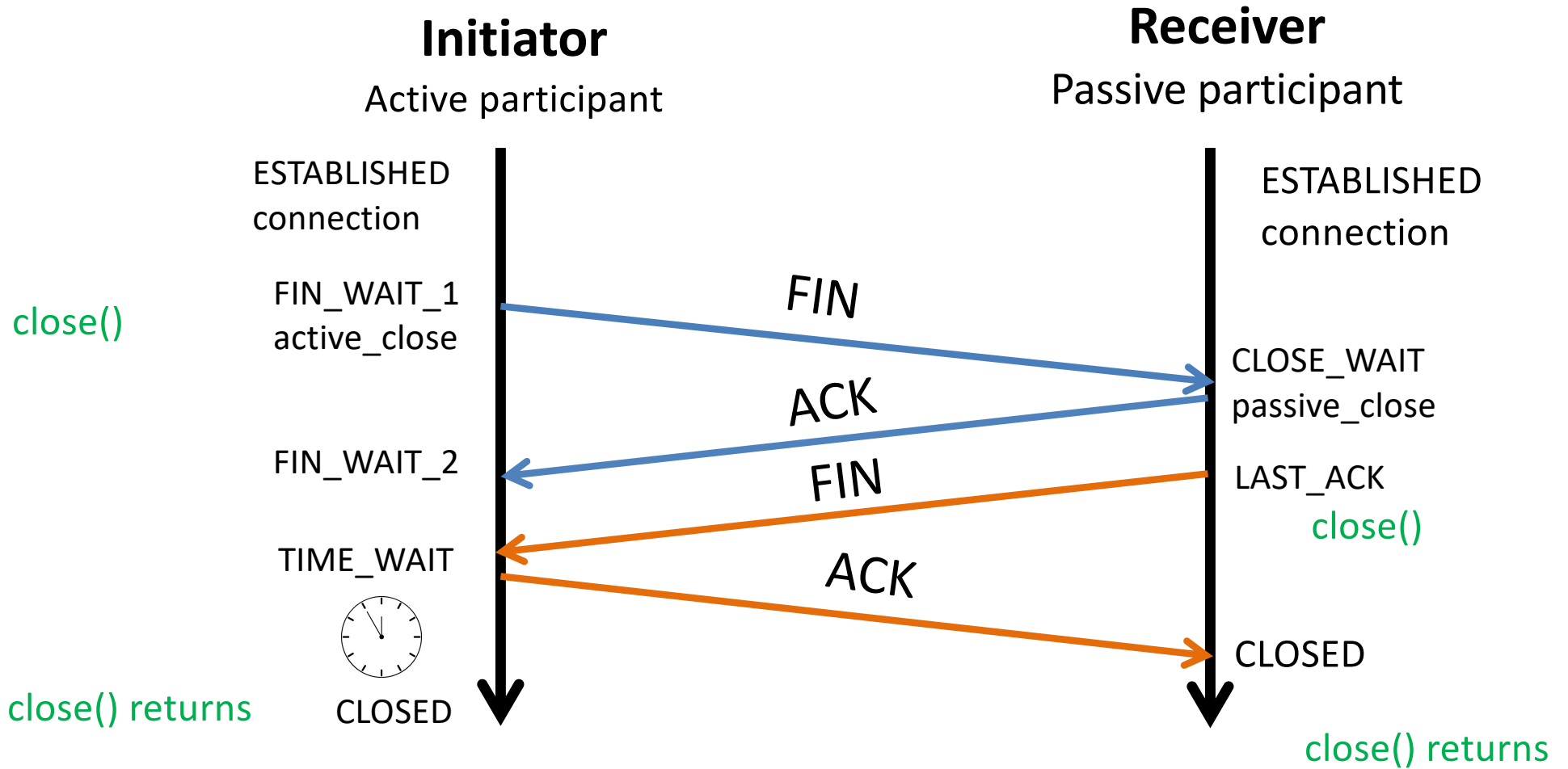
Receiver

Passive participant

Connection Teardown

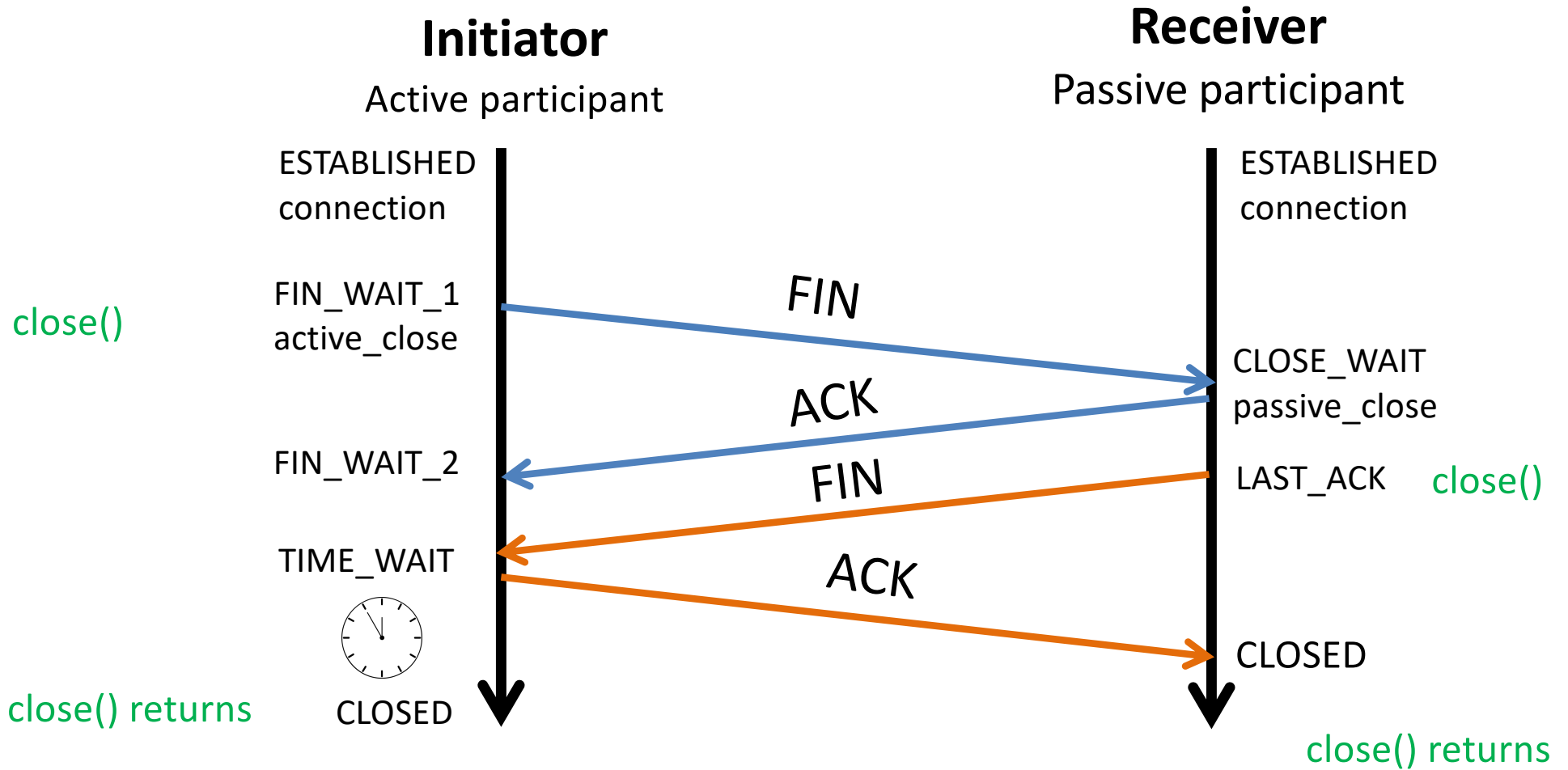
- Orderly release by sender and receiver when done
 - Delivers all pending data and “hangs up”
- Cleans up state in sender and receiver
- Each side may terminate independently

TCP Connection Teardown



Both sides agree on closing the connection.

TCP Connection Teardown



Both sides agree on closing the connection.



The TIME_WAIT State

- We wait $2 * \text{MSL}$ (maximum segment lifetime) before completing the close. The MSL is arbitrary (usually 60 sec)
- ACK might have been lost and so FIN will be resent
 - Could interfere with a subsequent connection
- This is why we used `SO_REUSEADDR` socket option in lab 2
 - Says to skip this waiting step and immediately abort the connection

Practical Reliability Questions

- What does connection establishment look like?
- **How do we choose sequence numbers?**
- How should we choose timeout values?
- How do the sender and receiver keep track of outstanding pipelined segments?
- How many segments should be pipelined?

How should we choose the initial sequence number?

A. Start from zero

B. Start from one

C. Start from a random number

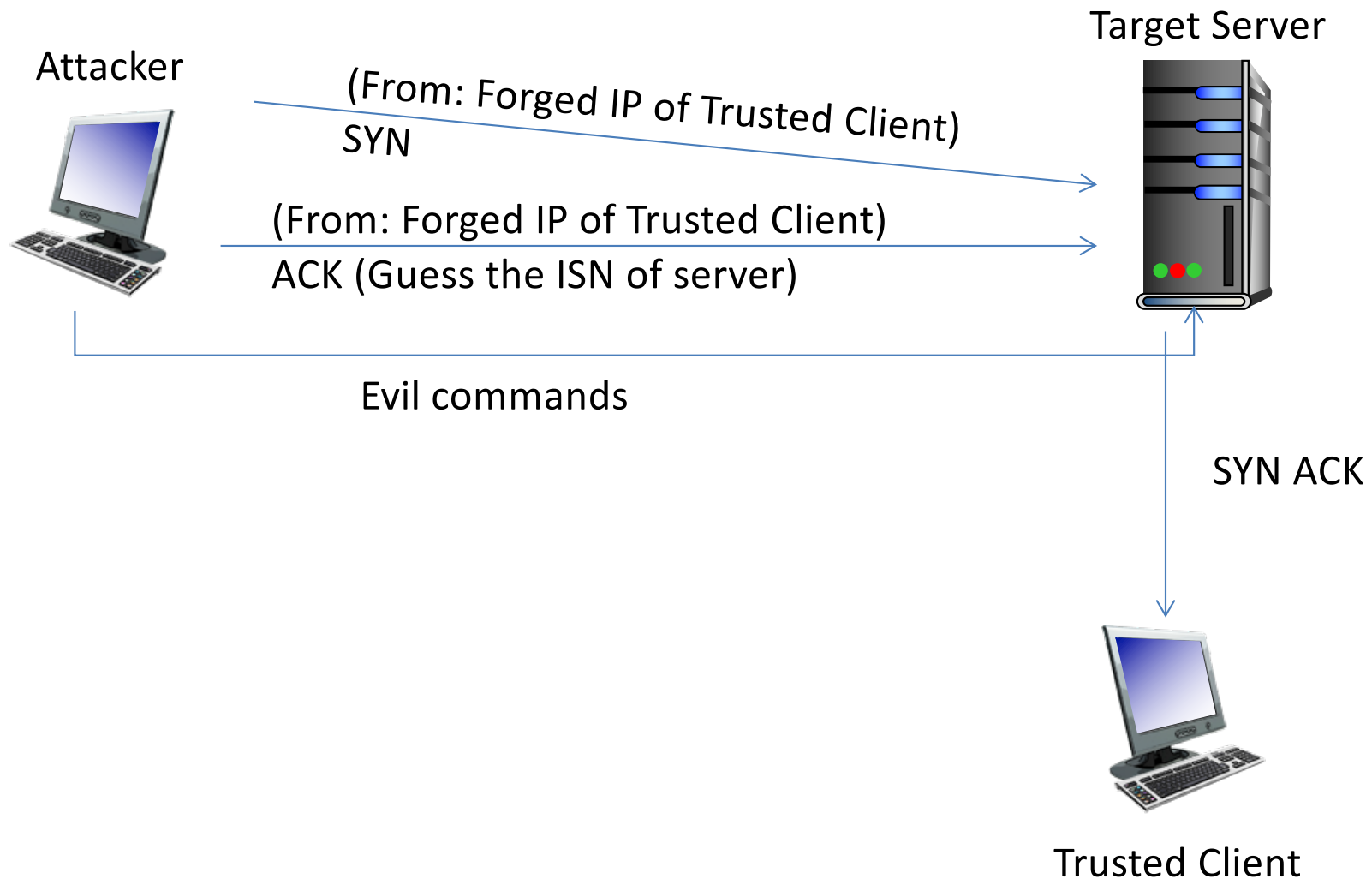
D. Start from some other value (such as...?)

What can go wrong with sequence numbers?
-How they're chosen?
-In the course of using them?

Sequencing

- Initial sequence numbers (ISN) chosen at random
 - Does not start at 0 or 1 (anymore).
 - Helps to prevent against forgery attacks.
- TCP sequences bytes rather than segments
 - Example: if we're sending 1500-byte segments
 - Randomly choose ISN (suppose we picked 1150)
 - First segment (sized 1500) would use number 1150
 - Next would use 2650

Sequence Prediction Attack (1996)



Practical Reliability Questions

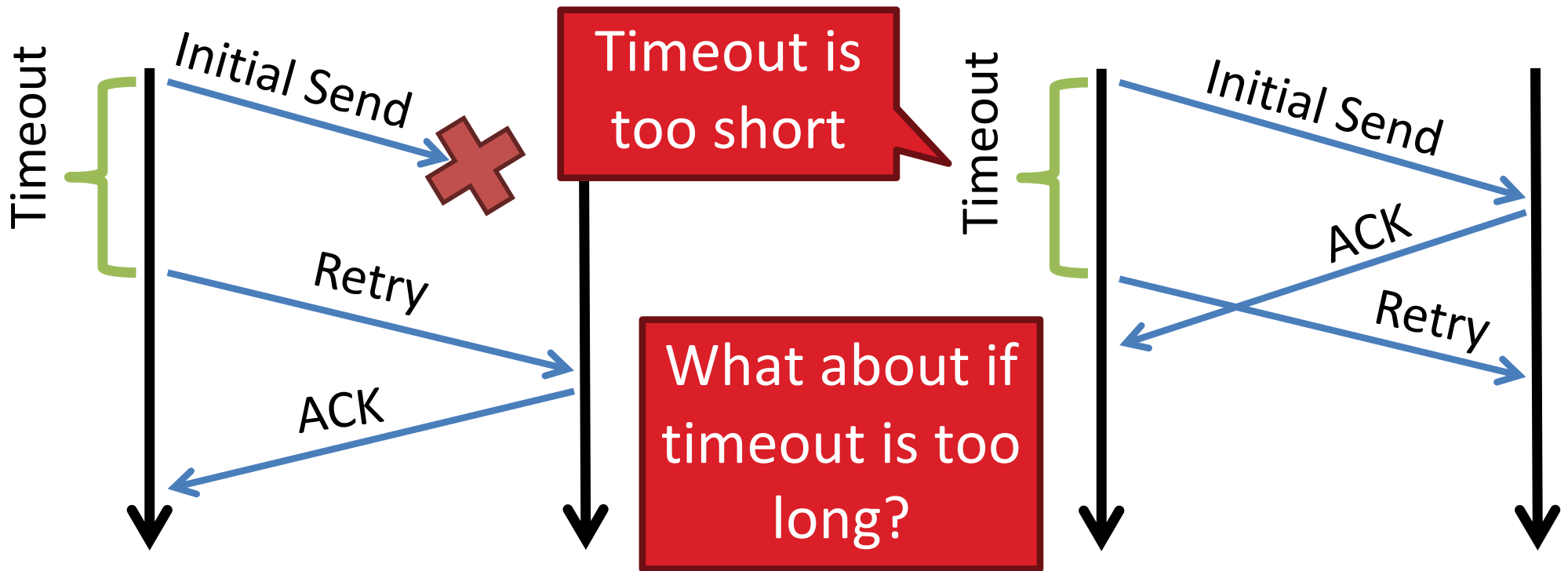
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- **How should we choose timeout values?**
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Timeouts

- How long should we wait before timing out and retransmitting a segment?
- Too short: needless retransmissions
- Too long: slow reaction to losses
- Should be (a little bit) longer than the RTT

Retransmission Time Outs (RTO)

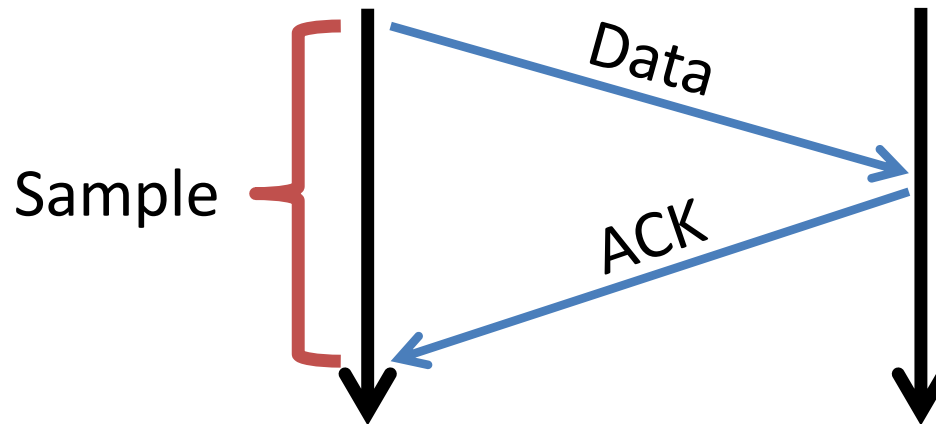
- Problem: time-out is linked to round trip time



Estimating RTT

- **Problem: RTT changes over time**
 - Routers buffer packets in queues
 - Queue lengths vary
 - Receiver may have varying load
- Sender takes measurements
 - Use statistics to decide future timeouts for sends
 - Estimate RTT and variance
- Apply “smoothing” to account for changes

Round Trip Time Estimation: Exponentially Weighted Moving Average (EWMA)



$$\text{EstimatedRTT} = (1 - a) * \text{EstimatedRTT} + a * \text{SampleRTT}$$

– a is usually 1/8.

In words current estimate is a blend of:

- 7/8 of the previous estimate
- 1/8 of the new sample.

$$\text{DevRTT} = (1 - B) * \text{DevRTT} + B * | \text{SampleRTT} - \text{EstimatedRTT} |$$

- B is usually 1/4

Example RTT Estimation

- Suppose EstimateRTT = 64, Dev = 8
- Latest sample: 120

$$\text{New estimate} = 7/8 * 64 + 1/8 * 120 = 56 + 15 = 71$$

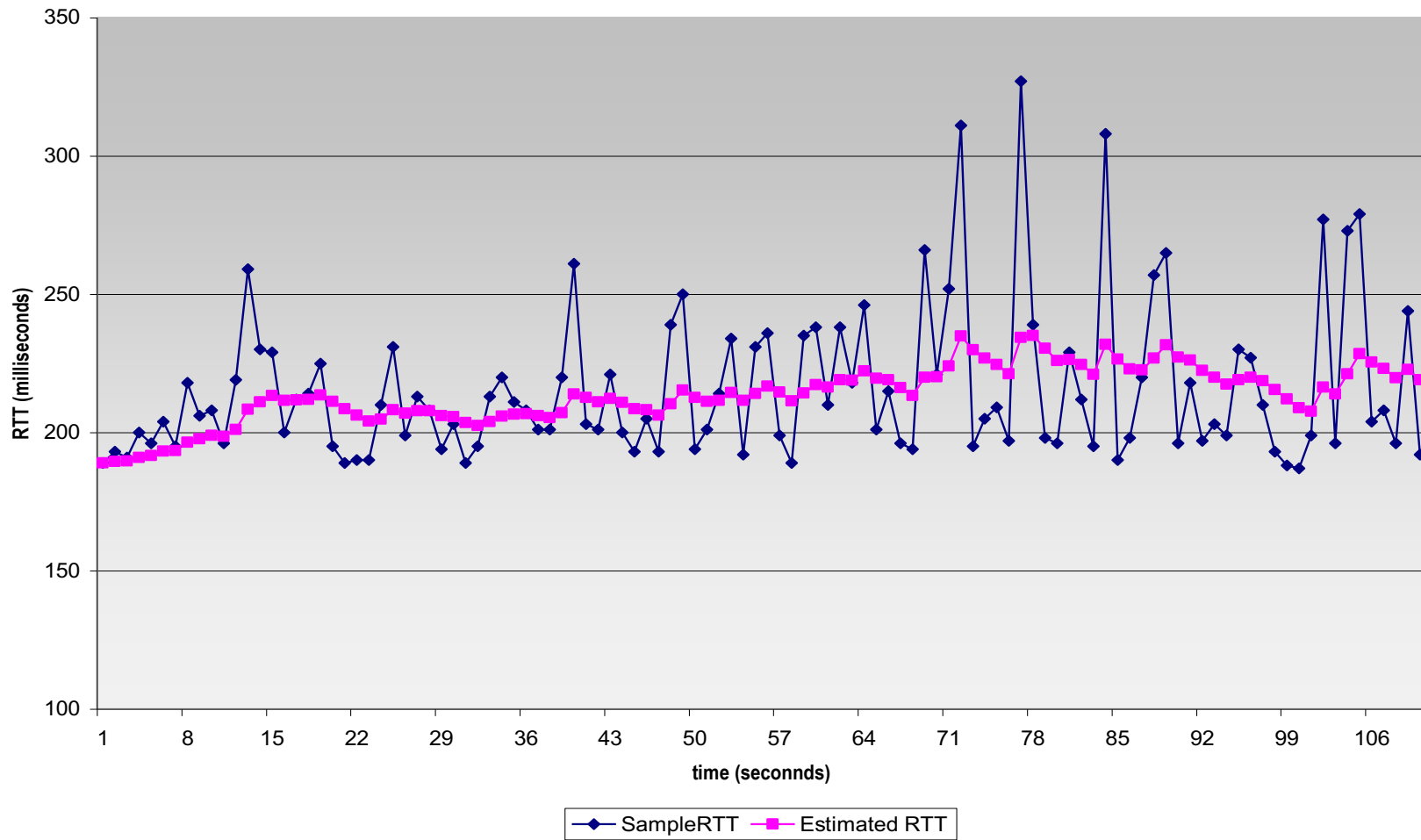
$$\text{New dev} = 3/4 * 8 + 1/4 * |120 - 71| = 6 + 12 = 18$$

- Another sample: 400

$$\text{New estimate} = 7/8 * 71 + 1/8 * 400 = 62 + 50 = 112$$


$$\text{New dev} = 3/4 * 18 + 1/4 * |400 - 112| = 13 + 72 = 85$$

Example RTT Estimation (Smoothing)



TCP Timeout Value

$$\text{TimeoutInterval} = \text{EstimatedRTT} + 4 * \text{DevRTT}$$

 ↑ ↑

estimated RTT “safety margin”