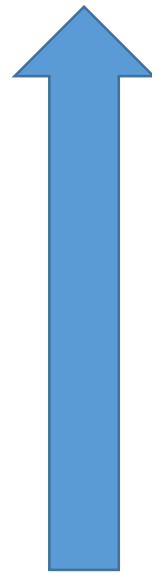
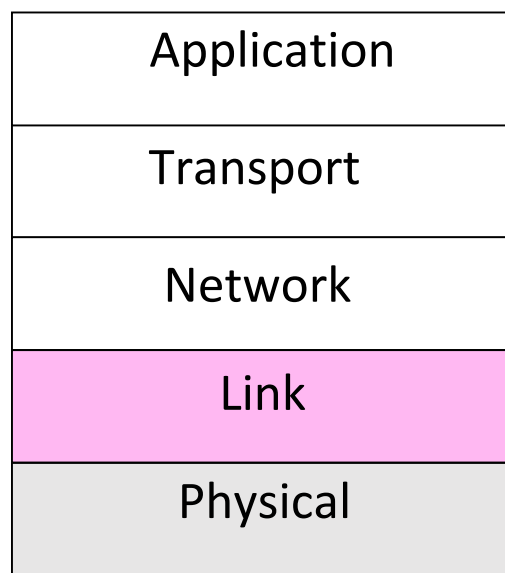


# Link Layer

(continued)

# Where we are in the Course

- Moving on up to the Link Layer!



# Topics

1. Framing
  - Delimiting start/end of frames
2. Error detection and **correction**
  - Handling errors
3. Retransmissions
  - Handling loss
4. Multiple Access
  - 802.11, classic Ethernet
5. Switching
  - Modern Ethernet

# Hamming Code

- Gives a method for constructing a code with a distance of 3
  - Uses  $n = 2^k - k - 1$ , e.g.,  $n=4, k=3$
  - Put check bits in positions  $p$  that are powers of 2, starting with position 1
  - Check bit in position  $p$  is parity of positions with a  $p$  term in their values

# Hamming Code (2)

- Example: data=0101, 3 check bits
  - 7 bit code, check bit positions 1, 2, 4
  - Check 1 covers positions 1, 3, 5, 7
  - Check 2 covers positions 2, 3, 6, 7
  - Check 4 covers positions 4, 5, 6, 7

0 1 0 0 1 0 1     $\longrightarrow$   
1 2 3 4 5 6 7

$$p_1 = 0 + 1 + 1 = 0, \quad p_2 = 0 + 0 + 1 = 1, \quad p_4 = 1 + 0 + 1 = 0$$

# Hamming Code (3)

- To decode:
  - Recompute check bits (with parity sum including the check bit)
  - Arrange as a binary number
  - Value (syndrome) tells error position
  - Value of zero means no error
  - Otherwise, flip bit to correct

# Hamming Code (4)

- Example, continued

→ 0 1 0 0 1 0 1  
1 2 3 4 5 6 7

$$p_1 = 0 + 0 + 1 + 1 = 0, \quad p_2 = 1 + 0 + 0 + 1 = 0,$$

$$p_4 = 0 + 1 + 0 + 1 = 0$$

Syndrome = 000, no error

Data = 0 1 0 1

# Hamming Code (5)

- Example, continued

→ 0 1 0 0 1 0 **0**  
1 2 3 4 5 6 7

$$p_1 = 0+0+1+0 = 1, \quad p_2 = 1+0+0+0 = 1,$$

$$p_4 = 0+1+0+0 = 1$$

Syndrome = **1 1 1**, flip position 7

Data = 0 1 0 1 (correct after flip!)



# Detection vs. Correction

- Which is better will depend on the pattern of errors.  
For example:
  - 1000 bit messages with a bit error rate (BER) of 1 in 10000
- Which has less overhead?

# Detection vs. Correction

- Which is better will depend on the pattern of errors.  
For example:
  - 1000 bit messages with a bit error rate (BER) of 1 in 10000
- Which has less overhead?
  - It still depends! We need to know more about the errors

# Detection vs. Correction (2)

Assume bit errors are random

- Messages have 0 or maybe 1 error (1/10 of the time)

Error correction:

- Need ~10 check bits per message
- Overhead:

Error detection:

- Need ~1 check bits per message plus 1000 bit retransmission
- Overhead:

# Detection vs. Correction (3)

Assume errors come in bursts of 100

- Only 1 or 2 messages in 1000 have significant (multi-bit) errors

Error correction:

- Need  $\gg 100$  check bits per message
- Overhead:

Error detection:

- Need 32 check bits per message plus 1000 bit resend 2/1000 of the time
- Overhead:

# Detection vs. Correction (4)

- Error correction:
  - Needed when errors are expected
  - Or when no time for retransmission
- Error detection:
  - More efficient when errors are not expected
  - And when errors are large when they do occur

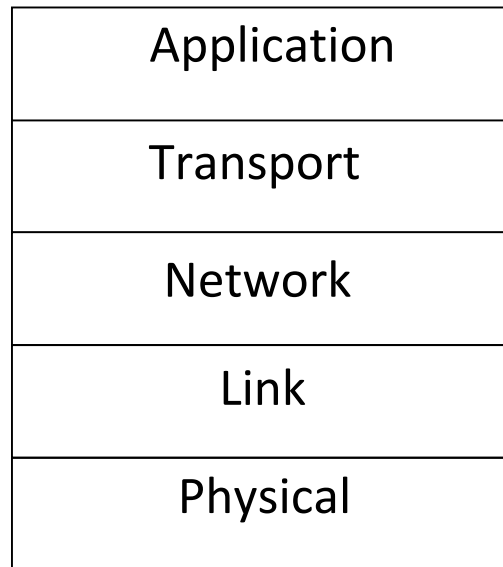
# Error Correction in Practice

- Heavily used in physical layer
  - LDPC is the future, used for demanding links like 802.11, DVB, WiMAX, LTE, power-line, ...
  - Convolutional codes widely used in practice
- Error detection (w/ retransmission) is used in the link layer and above for residual errors
- Correction also used in the application layer
  - Called Forward Error Correction (FEC)
  - Normally with an erasure error model
  - E.g., Reed-Solomon (CDs, DVDs, etc.)

Retransmissions

# Context on Reliability

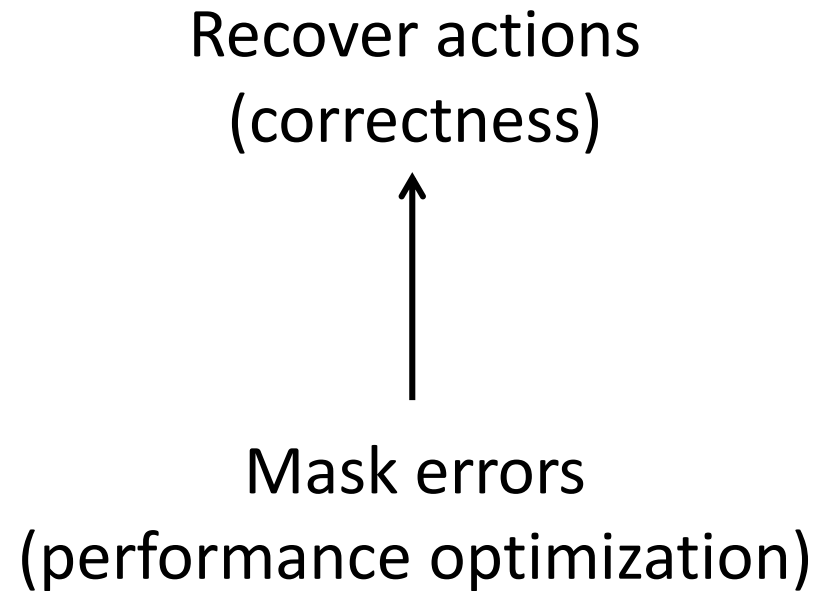
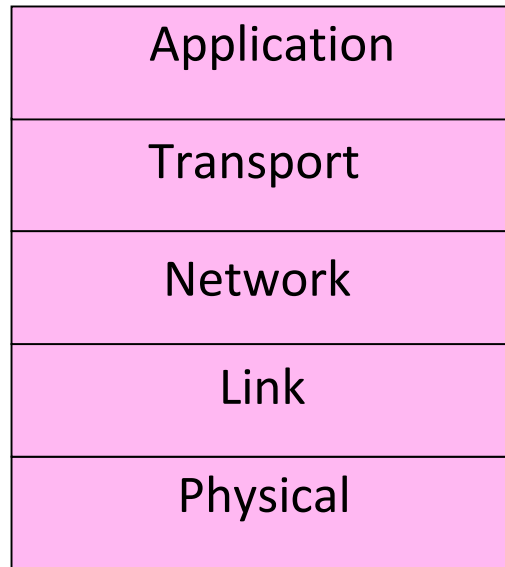
- Where in the stack should we place reliability functions?





# Context on Reliability (2)

- Everywhere! It is a key issue
  - Different layers contribute differently

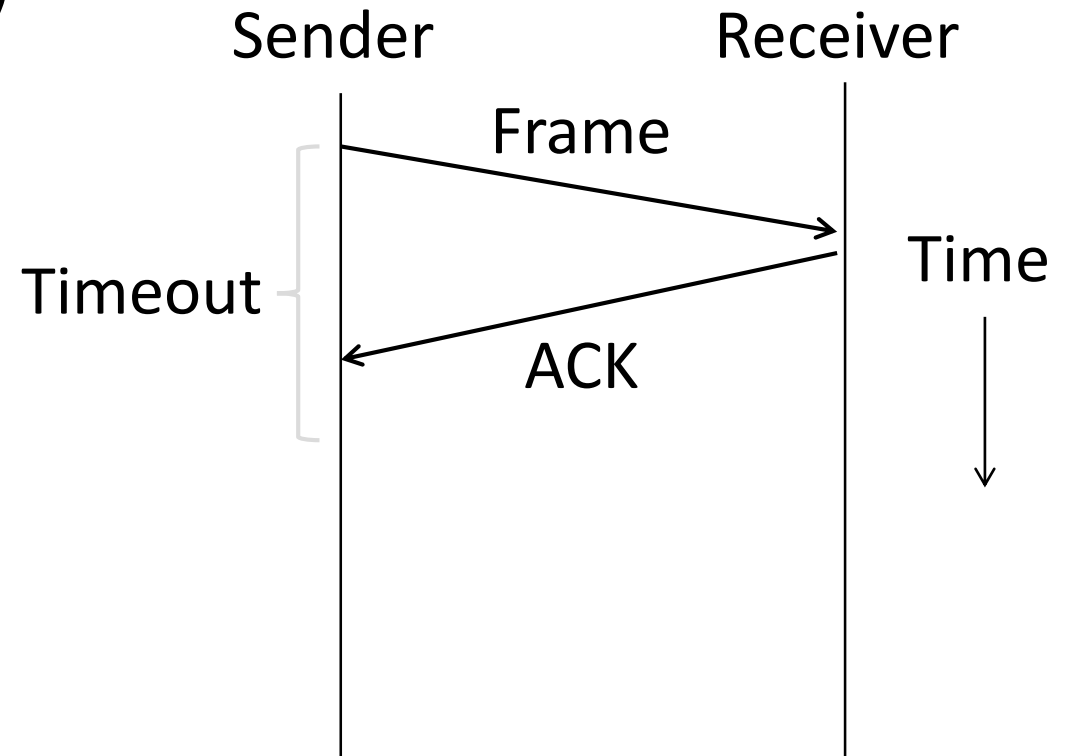


# ARQ (Automatic Repeat reQuest)

- ARQ often used when errors are common or must be corrected
  - E.g., WiFi, and TCP (later)
- Rules at sender and receiver:
  - Receiver automatically acknowledges correct frames with an ACK
  - Sender automatically resends after a timeout, until an ACK is received

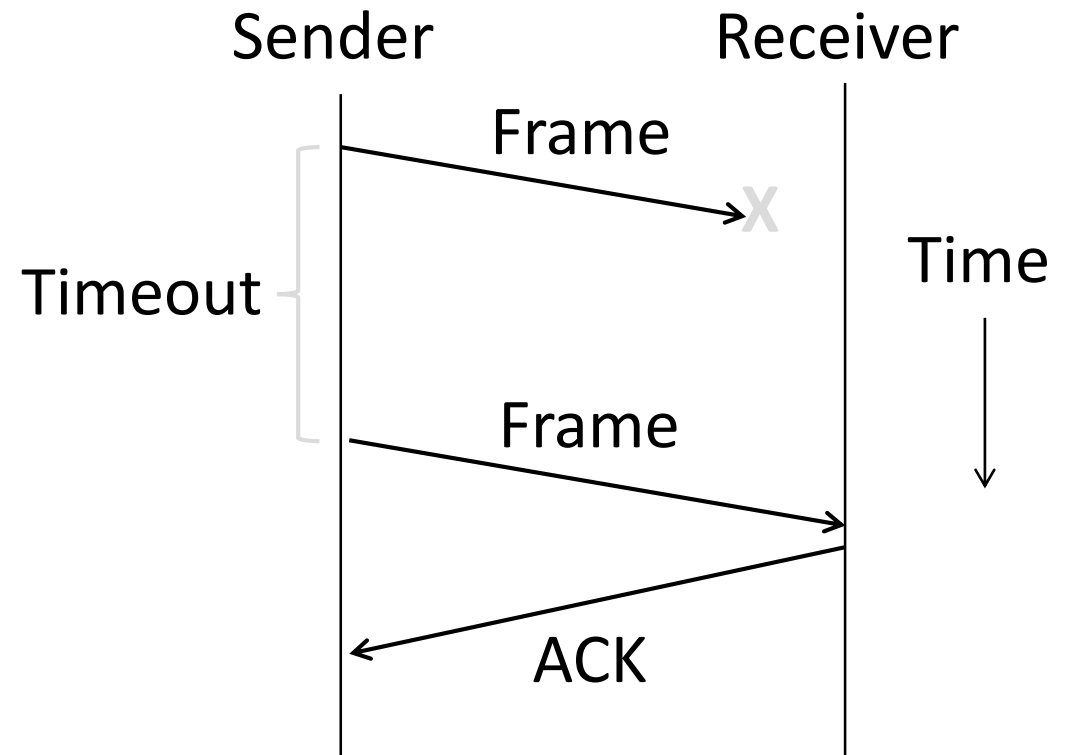
# ARQ (2)

- Normal operation (no loss)



# ARQ (3)

- Loss and retransmission



# So What's Tricky About ARQ?

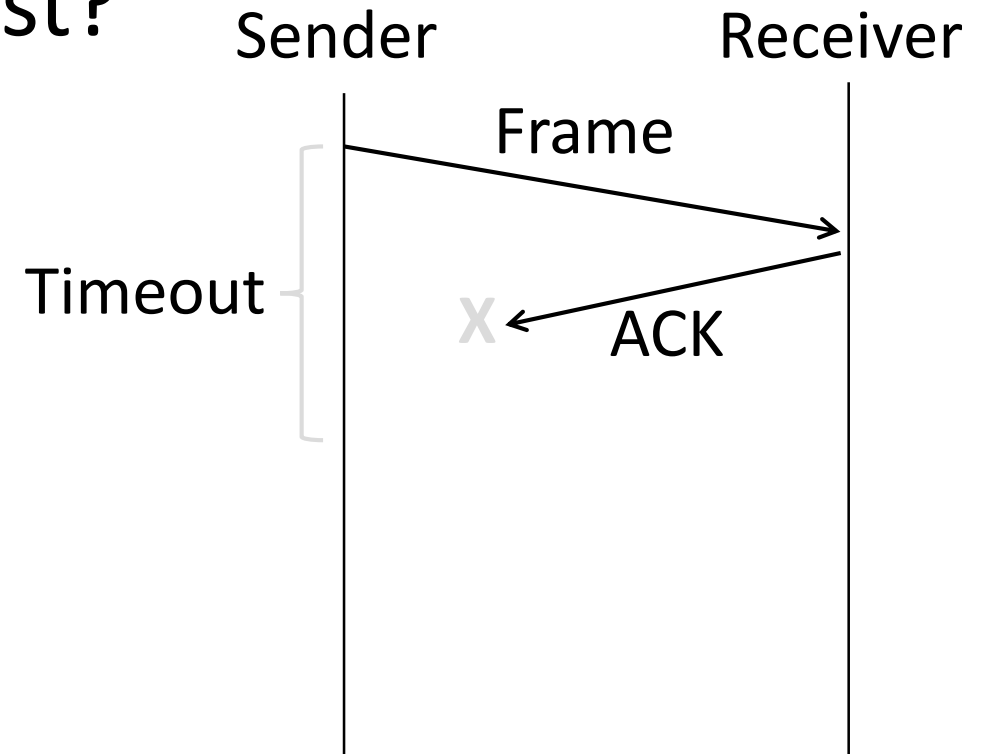
- Two non-trivial issues:
  - How long to set the timeout?
  - How to avoid accepting duplicate frames as new frames
- Want performance in the common case and correctness always

# Timeouts

- Timeout should be:
  - Not too big (link goes idle)
  - Not too small (spurious resend)
- Fairly easy on a LAN
  - Clear worst case, little variation
- Fairly difficult over the Internet
  - Much variation, no obvious bound
  - We'll revisit this with TCP (later)

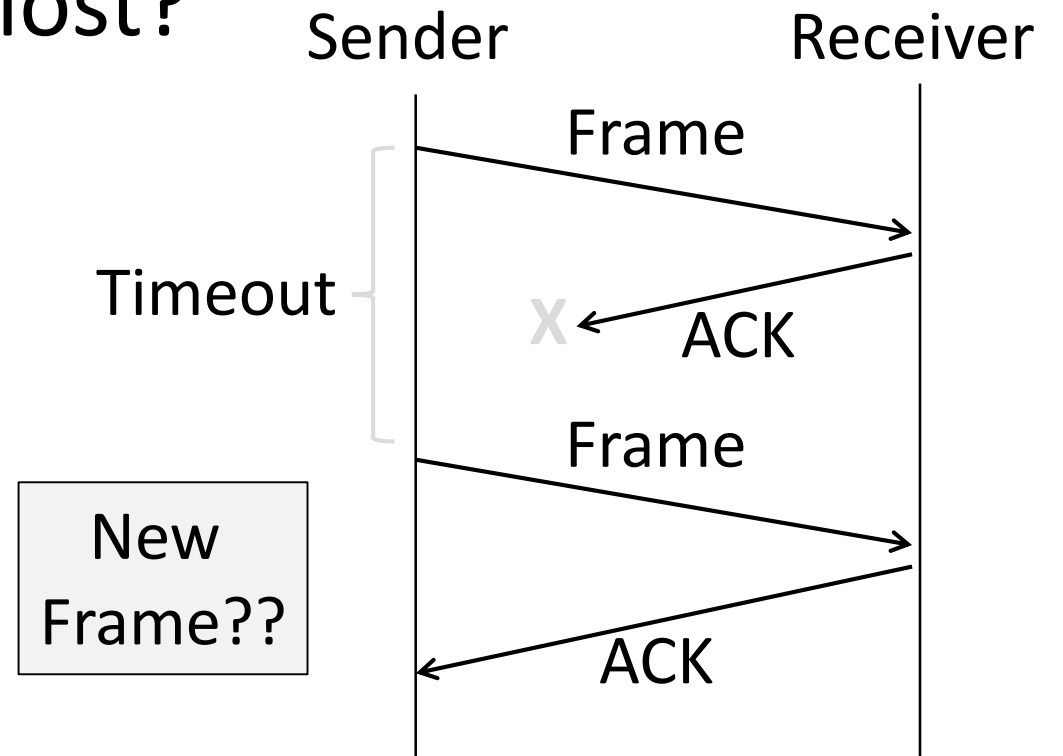
# Duplicates

- What happens if an ACK is lost?



# Duplicates (2)

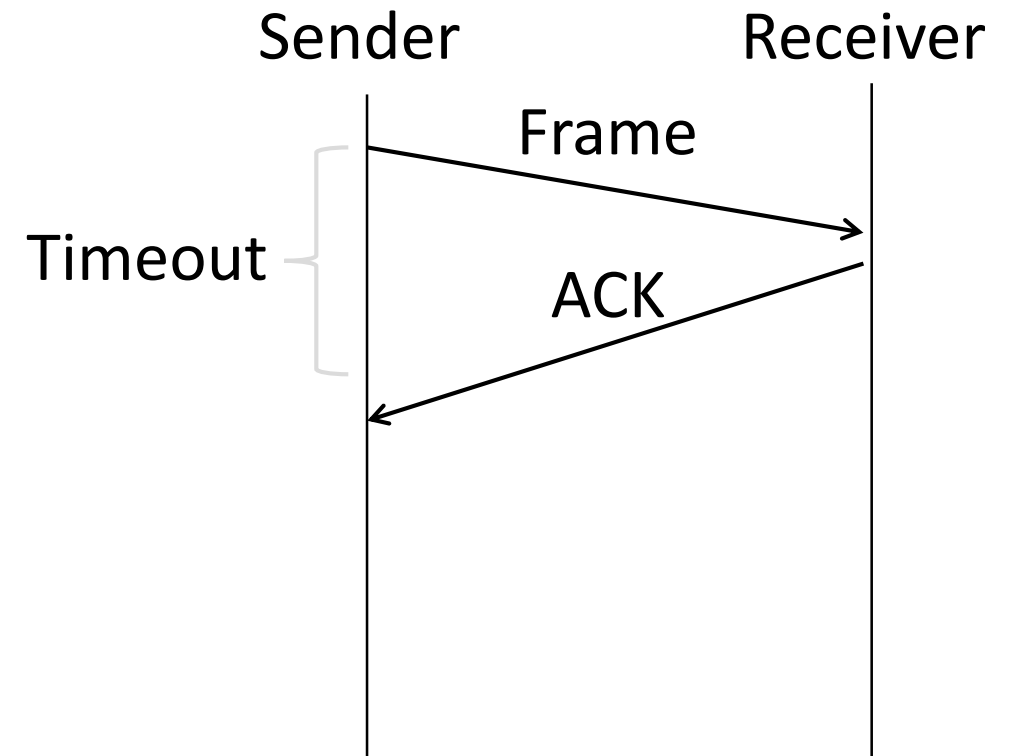
- What happens if an ACK is lost?





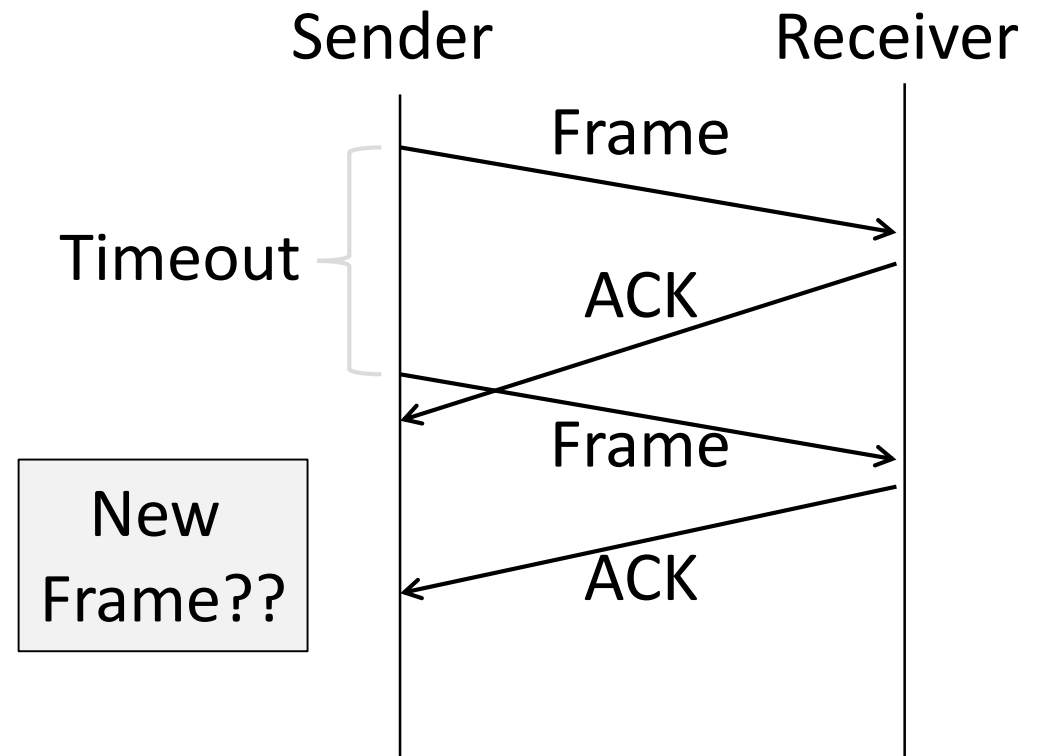
# Duplicates (3)

- Or the timeout is early?



# Duplicates (4)

- Or the timeout is early?

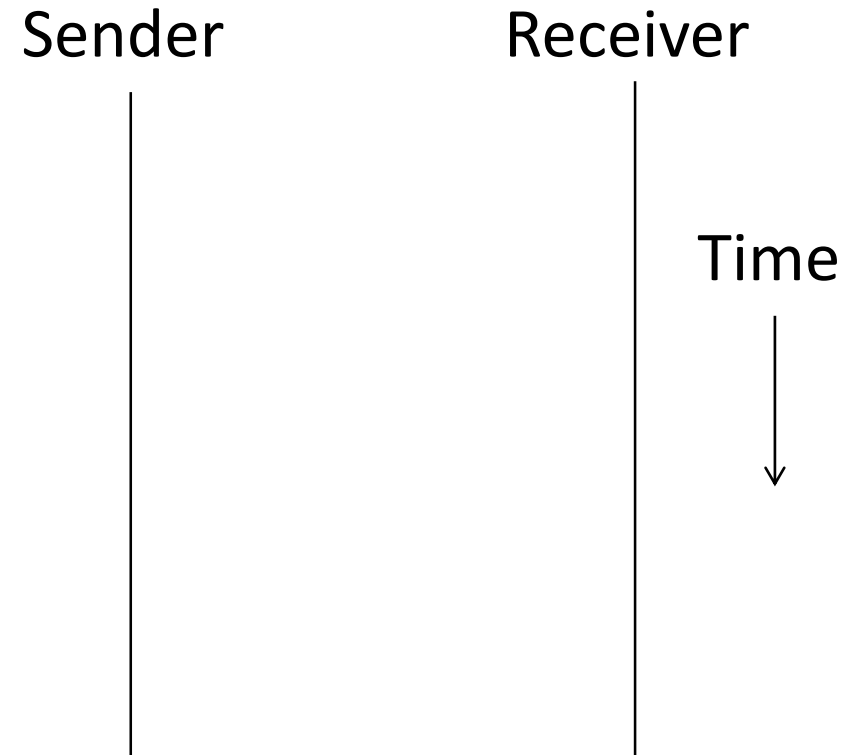


# Sequence Numbers

- Frames and ACKs must both carry sequence numbers for correctness
- To distinguish the current frame from the next one, a single bit (two numbers) is sufficient
  - Called Stop-and-Wait

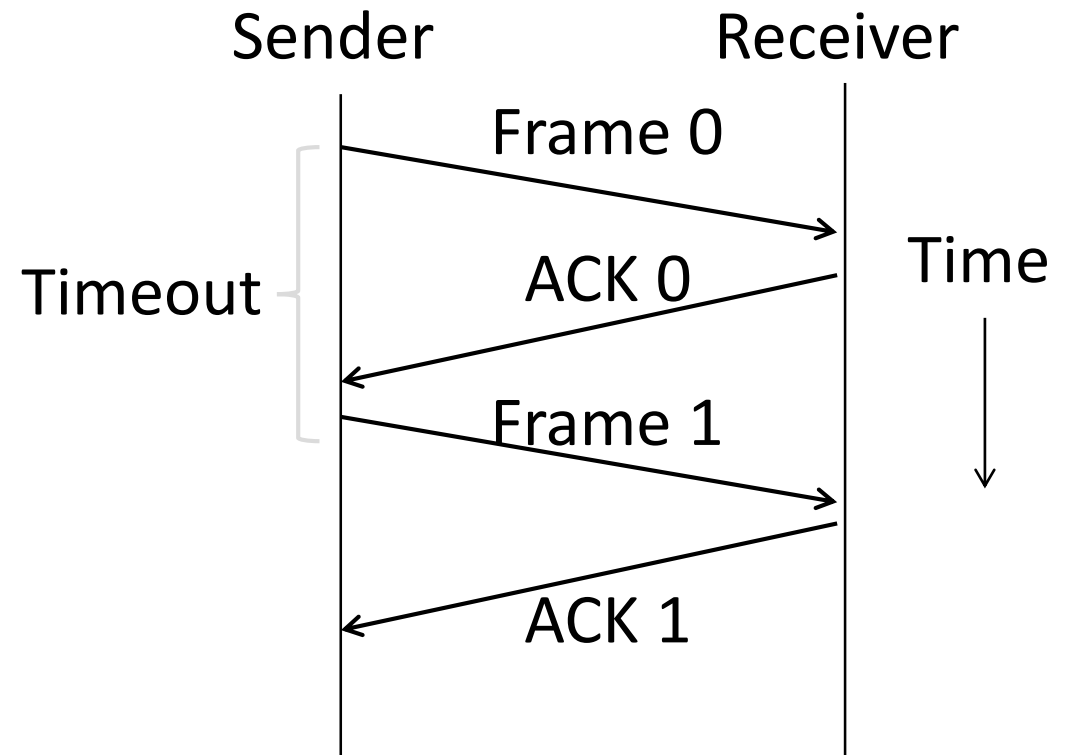
# Stop-and-Wait

- In the normal case:



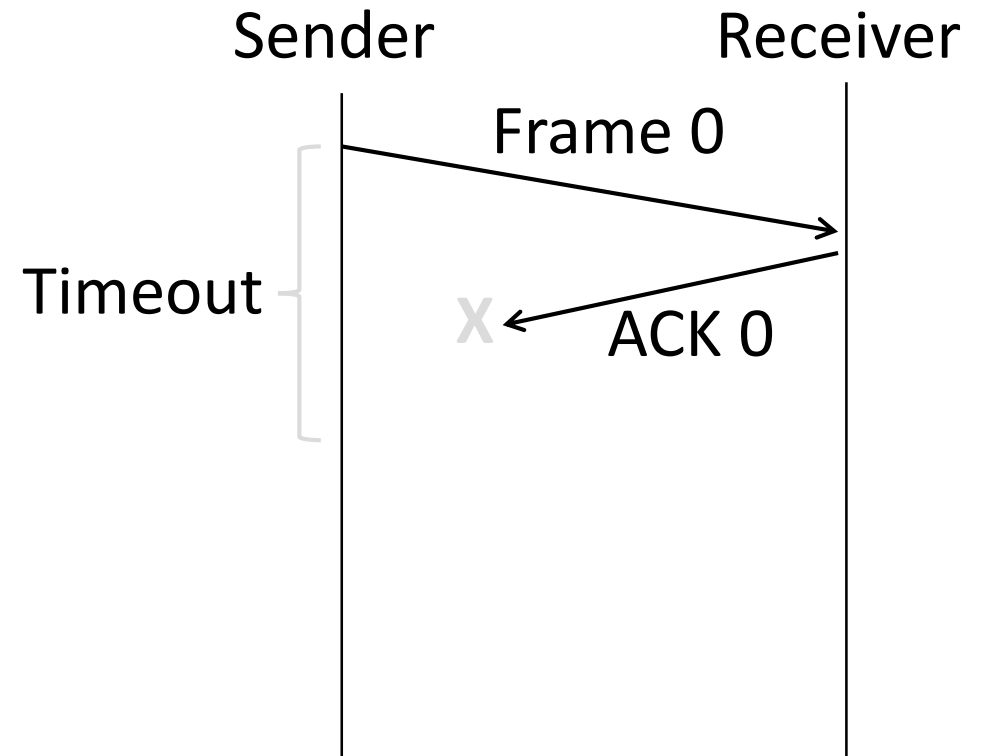
# Stop-and-Wait (2)

- In the normal case:



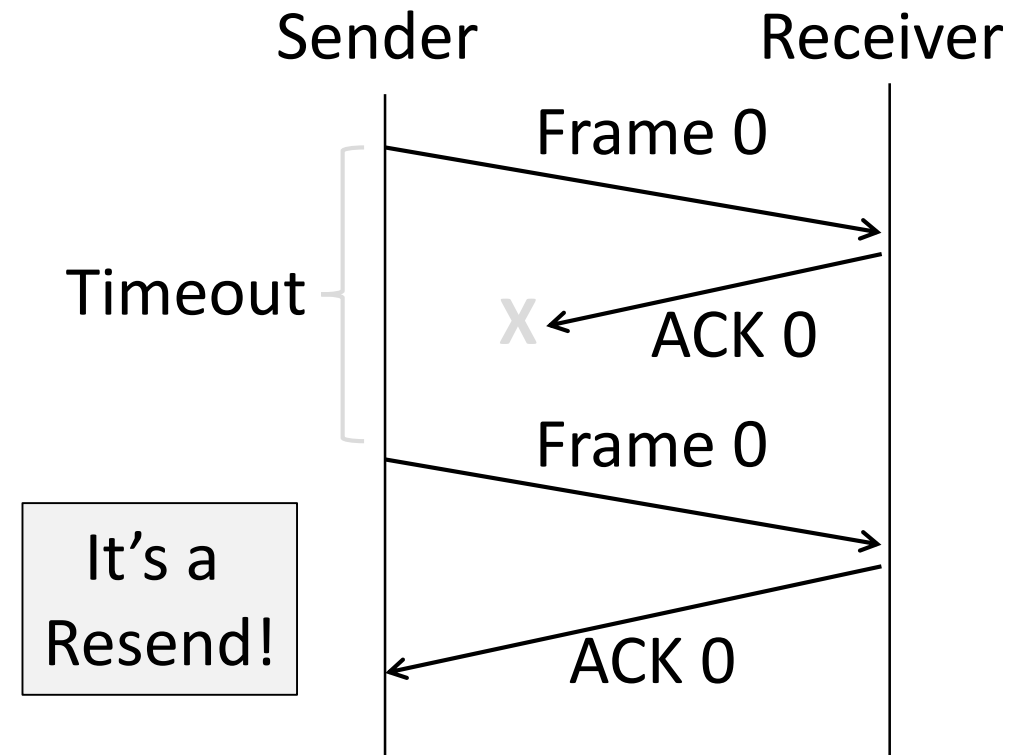
# Stop-and-Wait (3)

- With ACK loss:



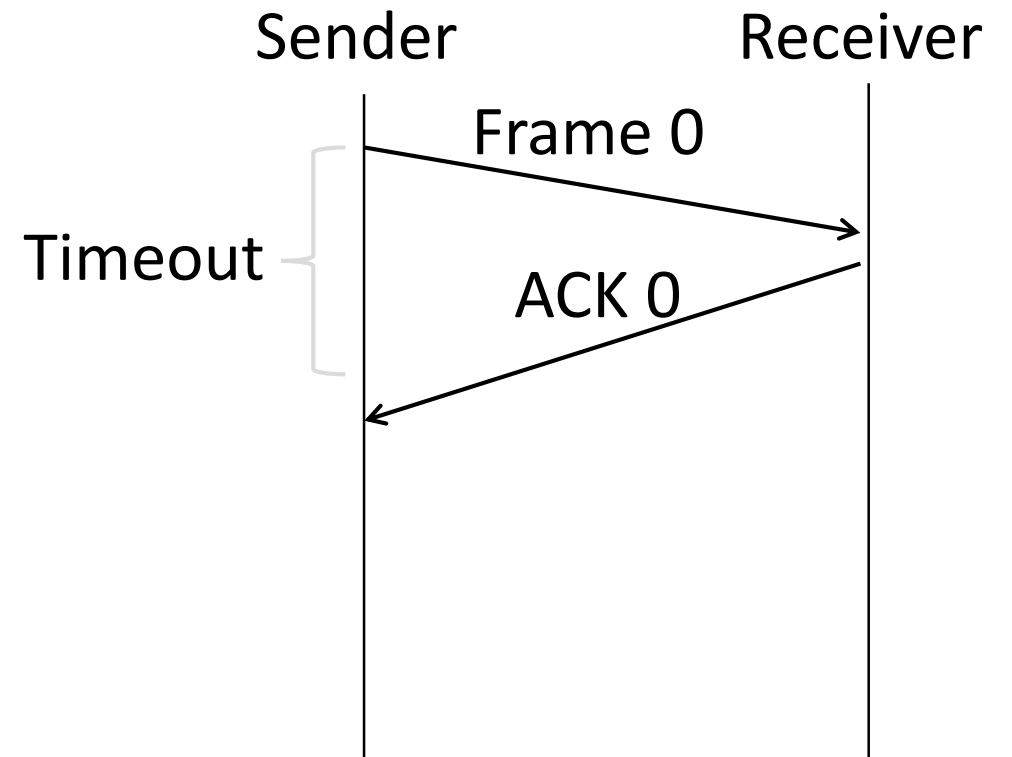
# Stop-and-Wait (4)

- With ACK loss:



# Stop-and-Wait (5)

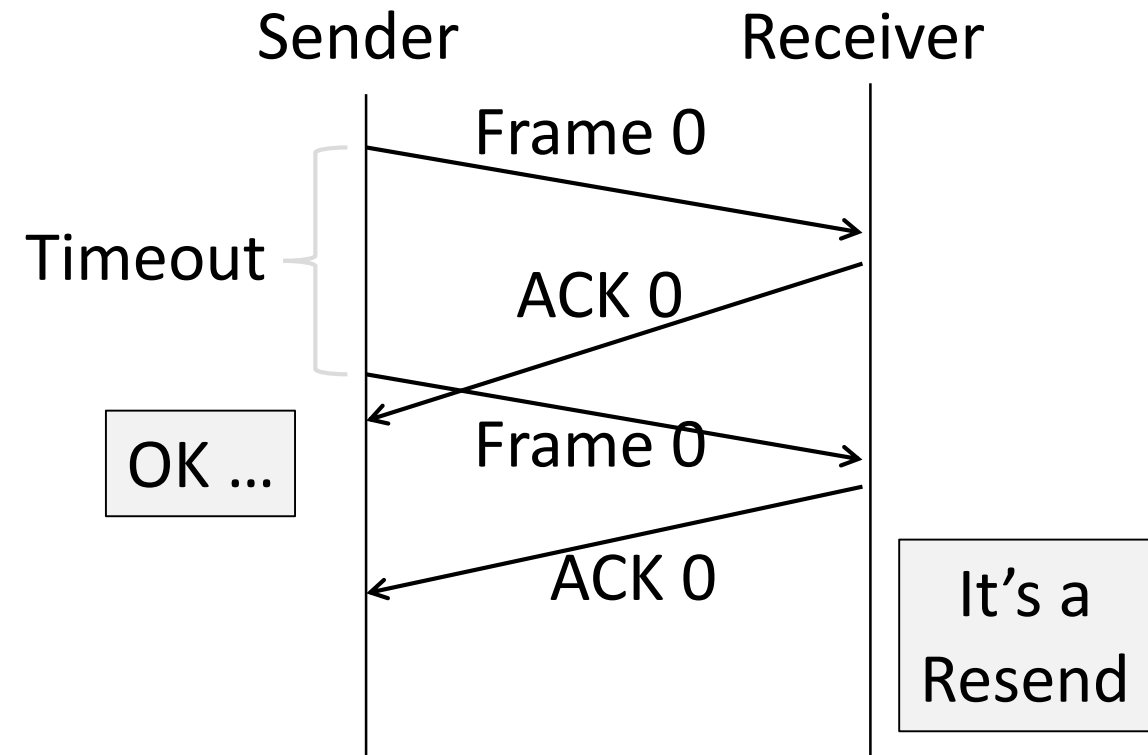
- With early timeout:





# Stop-and-Wait (6)

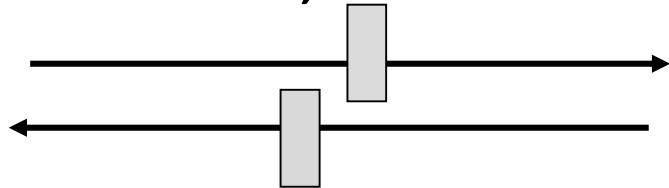
- With early timeout:



# Limitation of Stop-and-Wait

- It allows only a single frame to be outstanding from the sender:

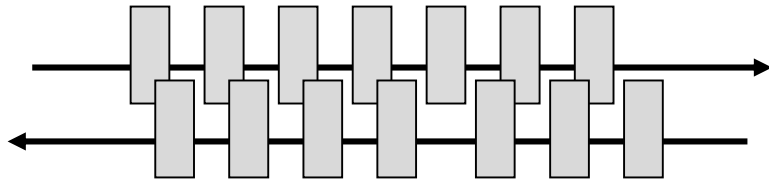
- Good for LAN, not efficient for high BD



- Ex:  $R=1$  Mbps,  $D = 50$  ms
  - How many frames/sec? If  $R=10$  Mbps?

# Sliding Window

- Generalization of stop-and-wait
  - Allows  $W$  frames to be outstanding
  - Can send  $W$  frames per RTT ( $=2D$ )



- Various options for numbering frames/ACKs and handling loss
  - Will look at along with TCP (later)

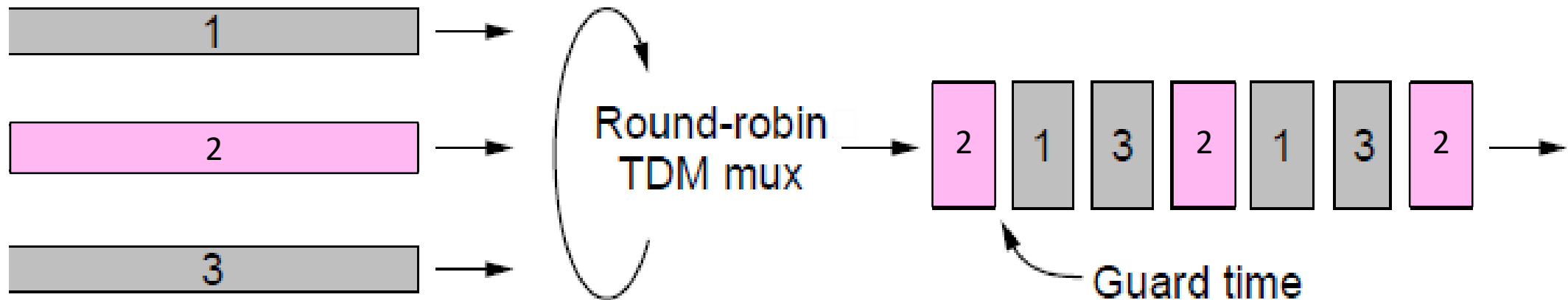
Multiple Access

# Topic

- Multiplexing is the network word for the sharing of a resource
- Classic scenario is sharing a link among different users
  - Time Division Multiplexing (TDM)
  - Frequency Division Multiplexing (FDM)

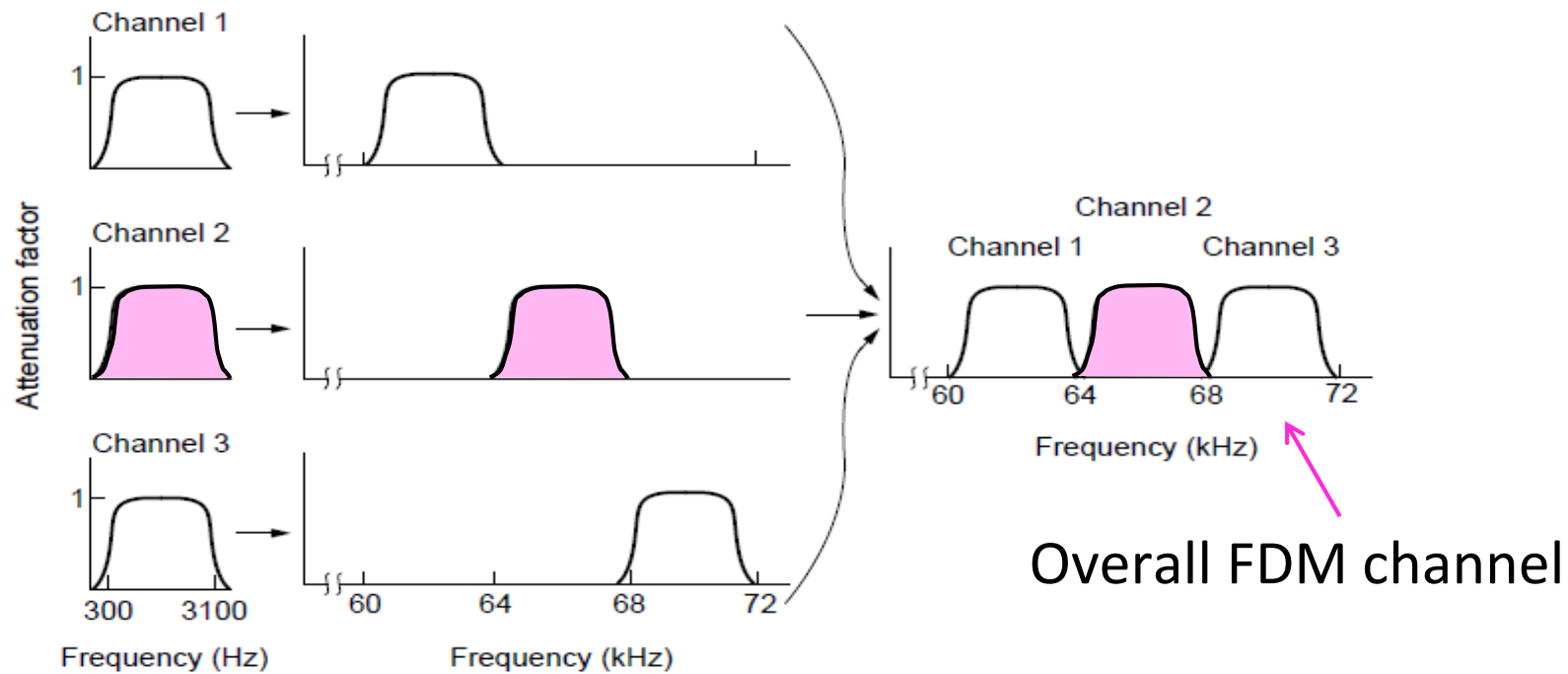
# Time Division Multiplexing (TDM)

- Users take turns on a fixed schedule



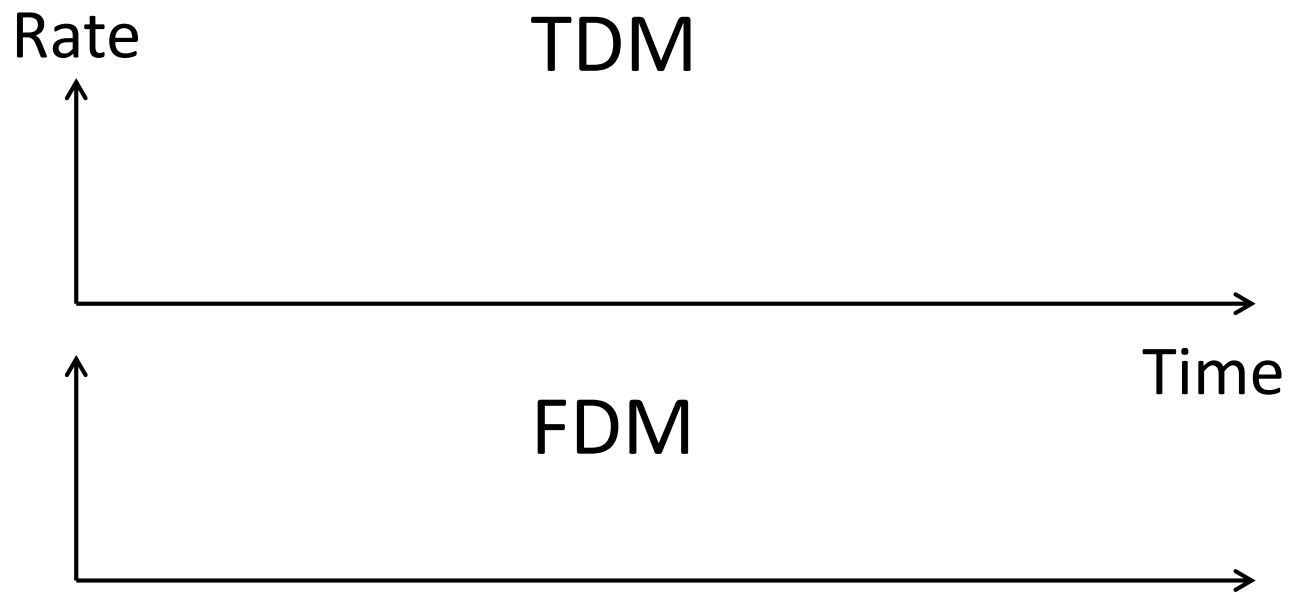
# Frequency Division Multiplexing (FDM)

- Put different users on different frequency bands



# TDM versus FDM

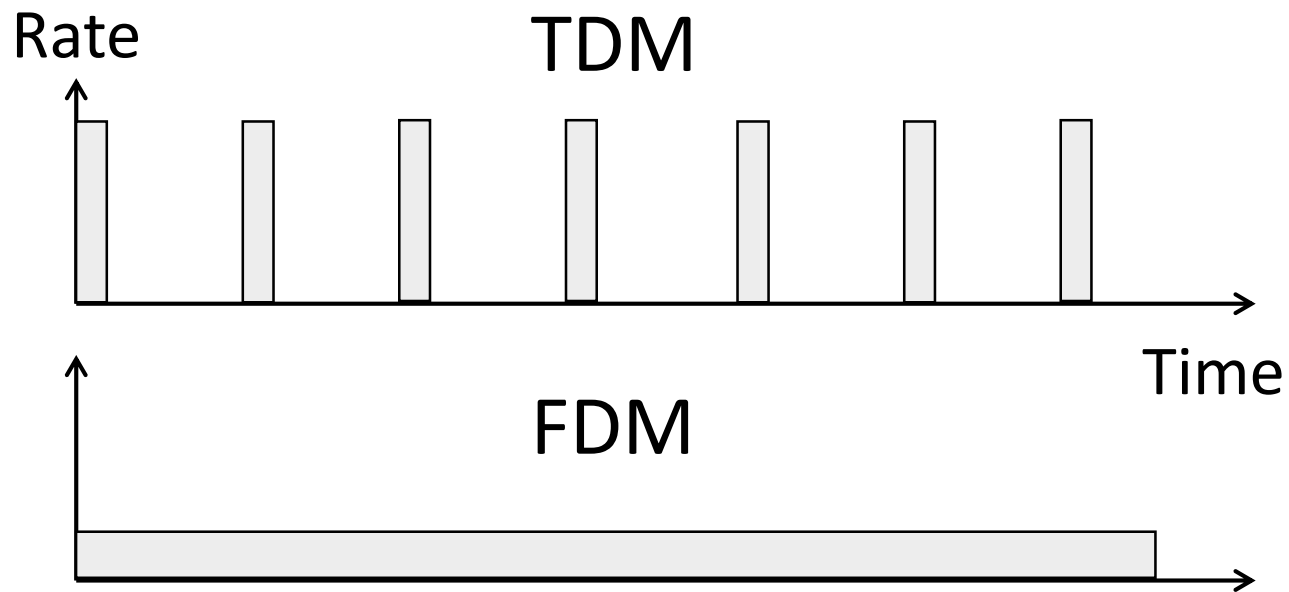
- In TDM a user sends at a high rate a fraction of the time; in FDM, a user sends at a low rate all the time





## TDM versus FDM (2)

- In TDM a user sends at a high rate a fraction of the time; in FDM, a user sends at a low rate all the time

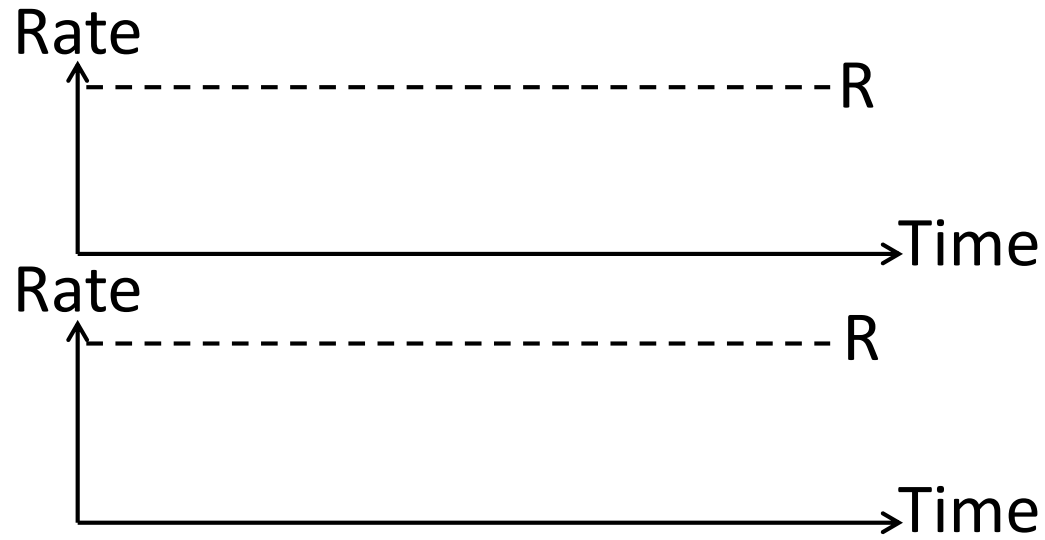


# TDM/FDM Usage

- **Statically divide a resource**
  - Suited for continuous traffic, fixed number of users
- **Widely used in telecommunications**
  - TV and radio stations (FDM)
  - GSM (2G cellular) allocates calls using TDM within FDM

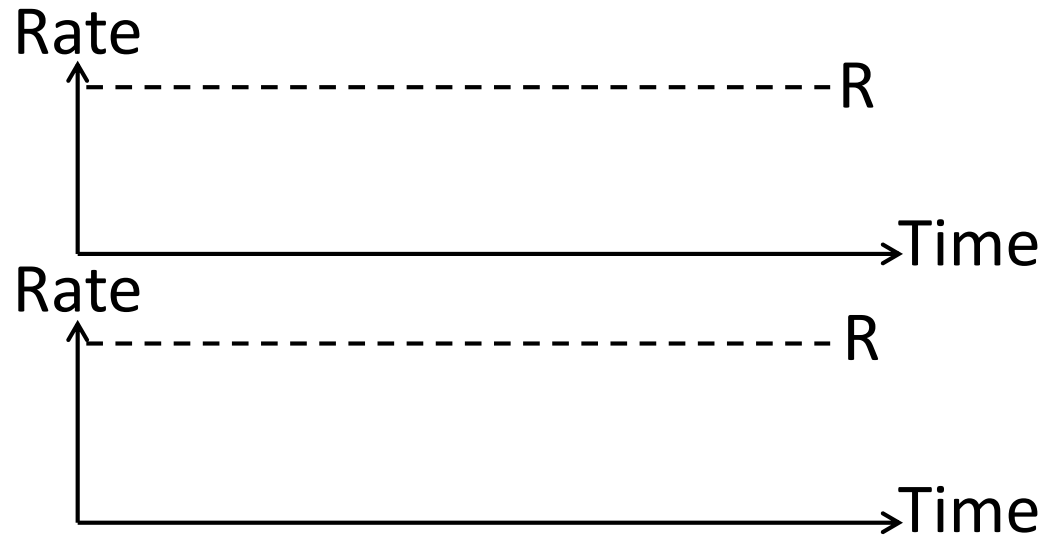
# Multiplexing Network Traffic

- Network traffic is bursty
  - ON/OFF sources
  - Load varies greatly over time



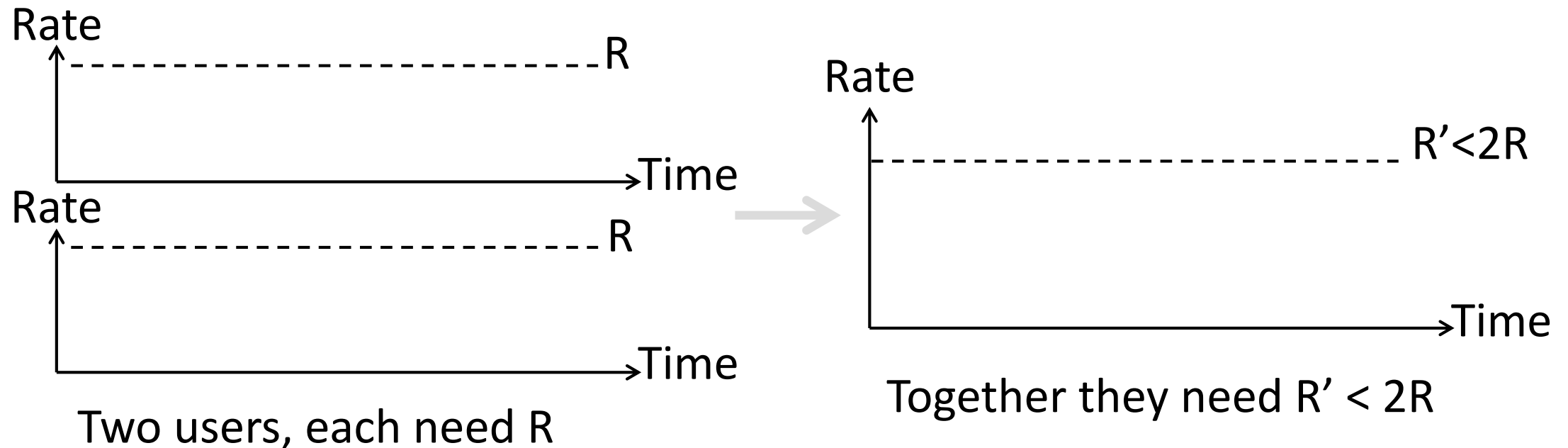
# Multiplexing Network Traffic (2)

- Network traffic is bursty
  - Inefficient to always allocate user their ON needs with TDM/FDM



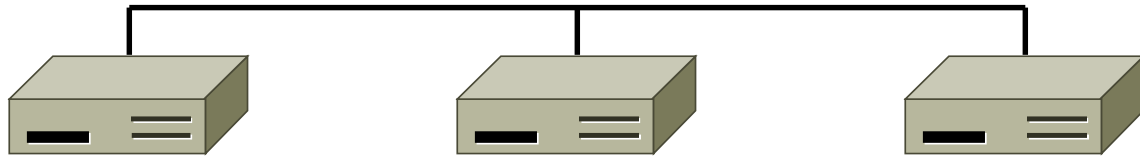
# Multiplexing Network Traffic (3)

- Multiple access schemes multiplex users according to demands – for gains of statistical multiplexing



# Random Access

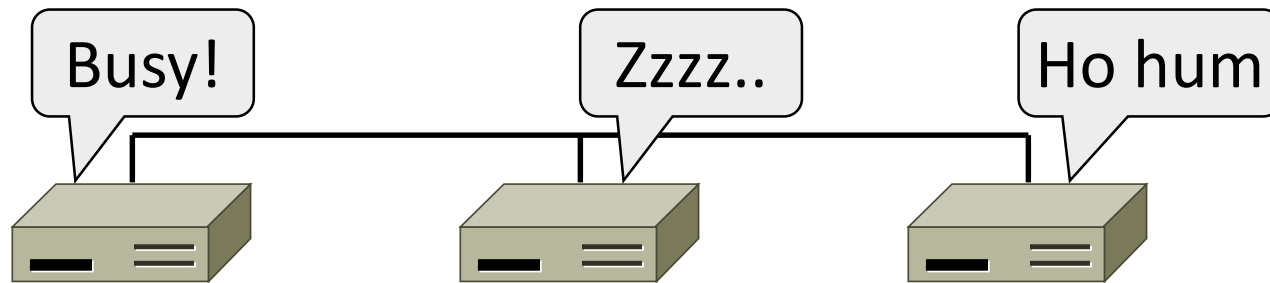
- How do nodes share a single link? Who sends when, e.g., in WiFi?
  - Explore with a simple model



- Assume no-one is in charge
  - Distributed system

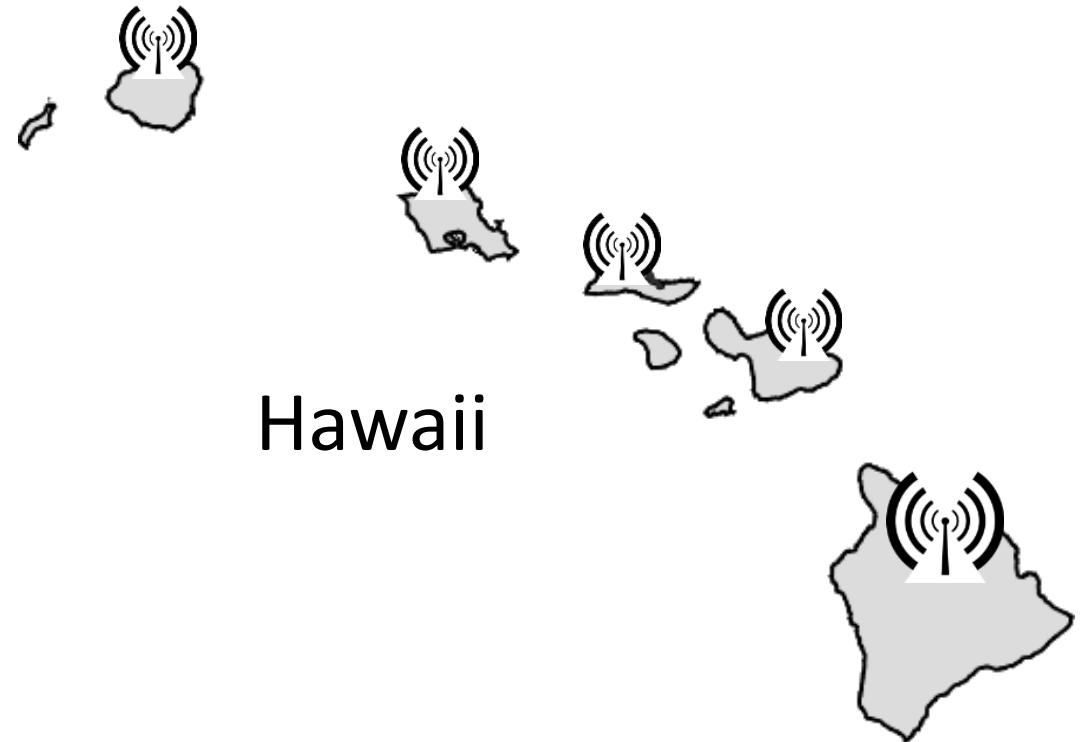
# Random Access (2)

- We will explore random multiple access control (MAC) protocols
  - This is the basis for classic Ethernet
  - Remember: data traffic is bursty



# ALOHA Network

- Seminal computer network connecting the Hawaiian islands in the late 1960s
  - When should nodes send?
  - A new protocol was devised by Norm Abramson ...



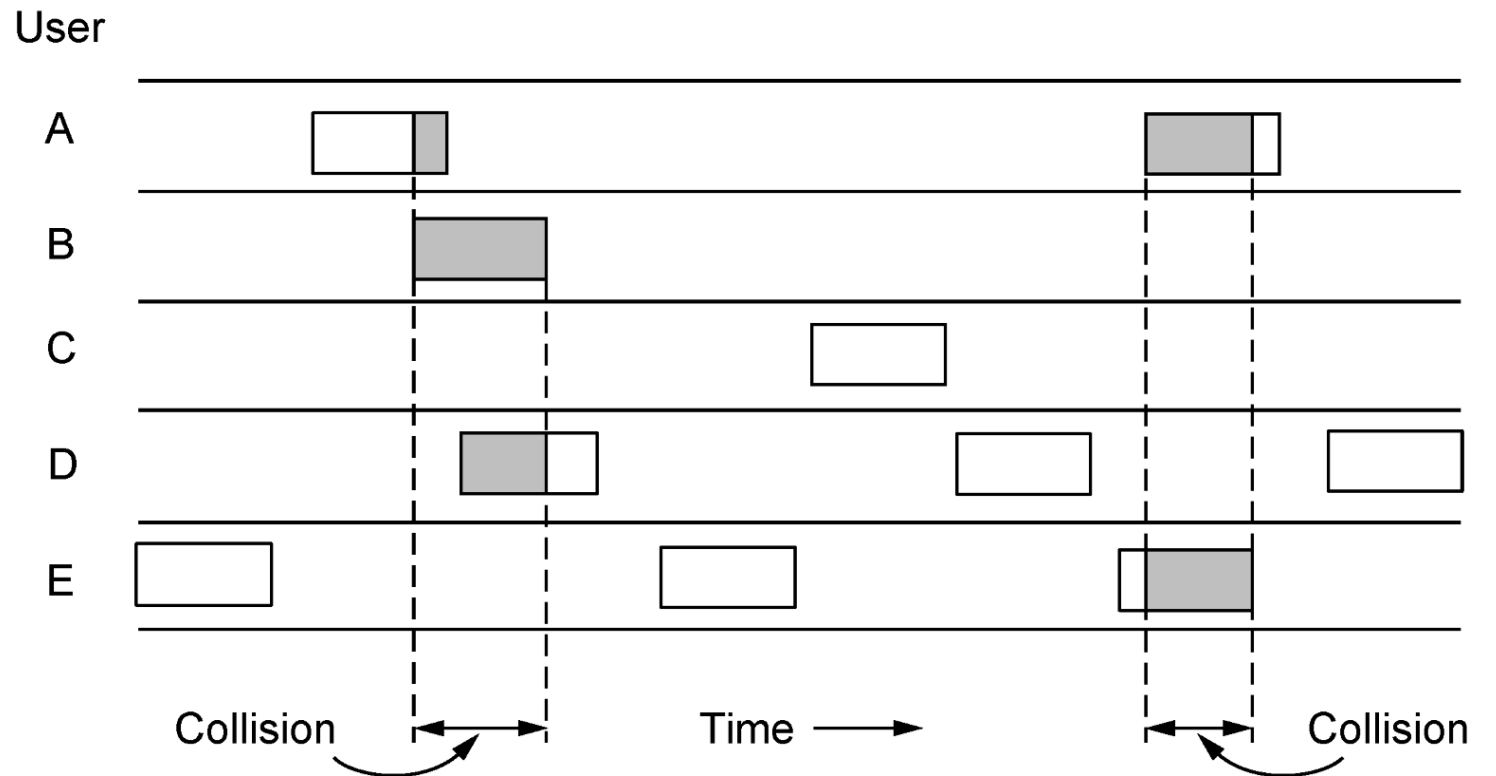


# ALOHA Protocol

- Simple idea:
  - Node just sends when it has traffic.
  - If there was a collision (no ACK received) then wait a random time and resend
- That's it!

# ALOHA Protocol (2)

- Some frames will be lost, but many may get through...
- Good idea?

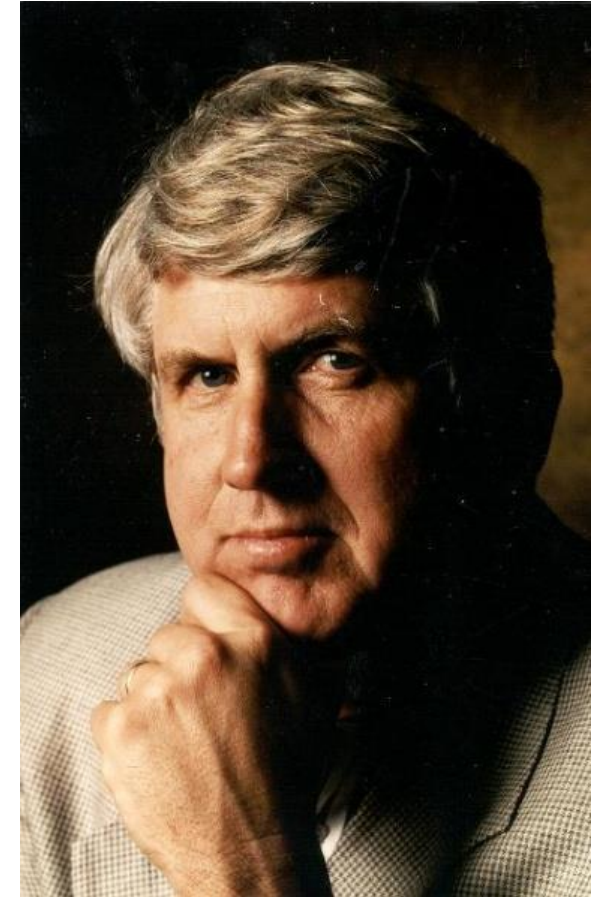
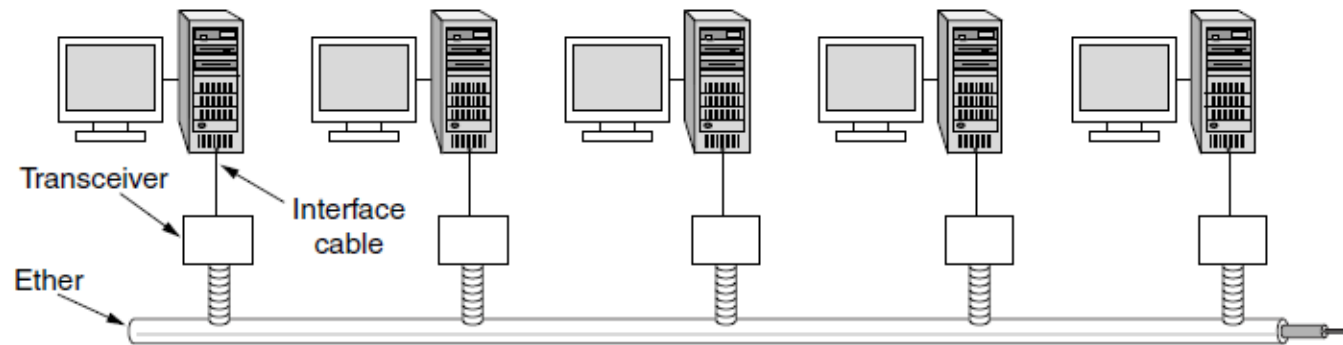


# ALOHA Protocol (3)

- Simple, decentralized protocol that works well under low load!
- Not efficient under high load
  - Analysis shows at most 18% efficiency
  - Improvement: divide time into slots and efficiency goes up to 36%
- We'll look at other improvements

# Classic Ethernet

- ALOHA inspired Bob Metcalfe to invent Ethernet for LANs in 1973
  - Nodes share 10 Mbps coaxial cable
  - Hugely popular in 1980s, 1990s



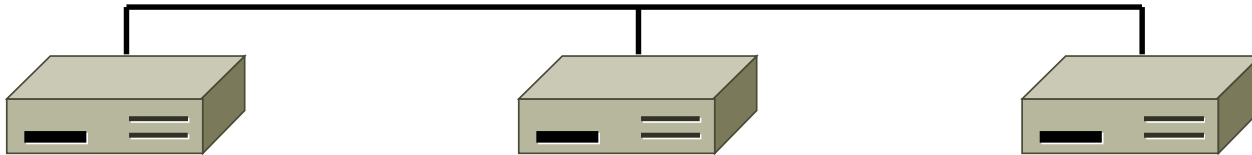
: © 2009 IEEE

# CSMA (Carrier Sense Multiple Access)

- Improve ALOHA by listening for activity before we send (Doh!)
  - Can do easily with wires, not wireless
- So does this eliminate collisions?
  - Why or why not?

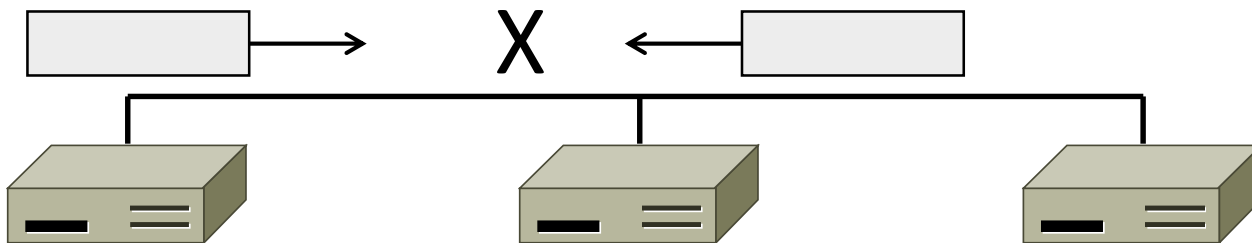
## CSMA (2)

- Still possible to listen and hear nothing when another node is sending because of delay



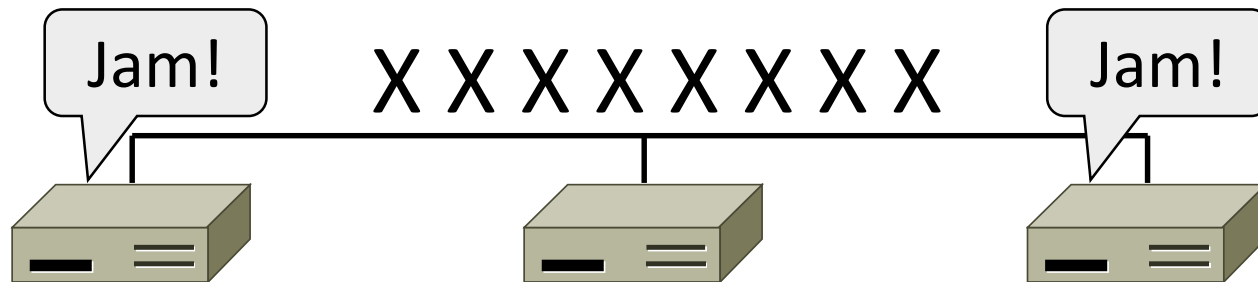
## CSMA (3)

- CSMA is a good defense against collisions only when BD is small



# CSMA/CD (with Collision Detection)

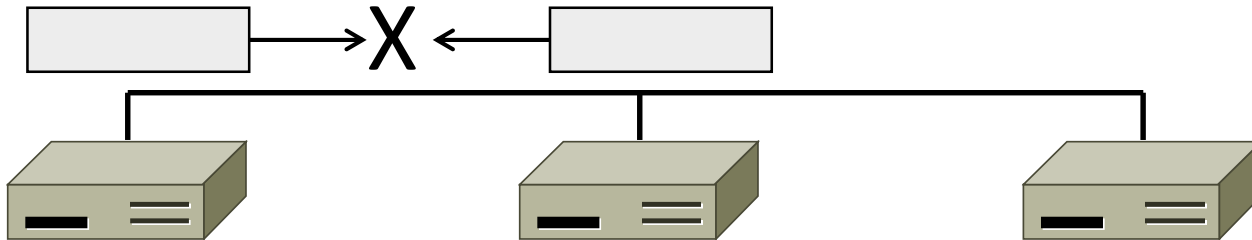
- Can reduce the cost of collisions by detecting them and aborting (Jam) the rest of the frame time
  - Again, we can do this with wires





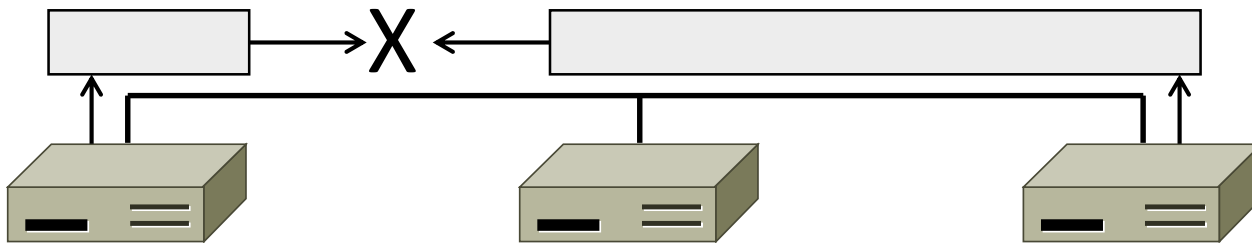
# CSMA/CD Complications

- Everyone who collides needs to know it happened
  - Time window in which a node may hear of a collision is  $2D$  seconds



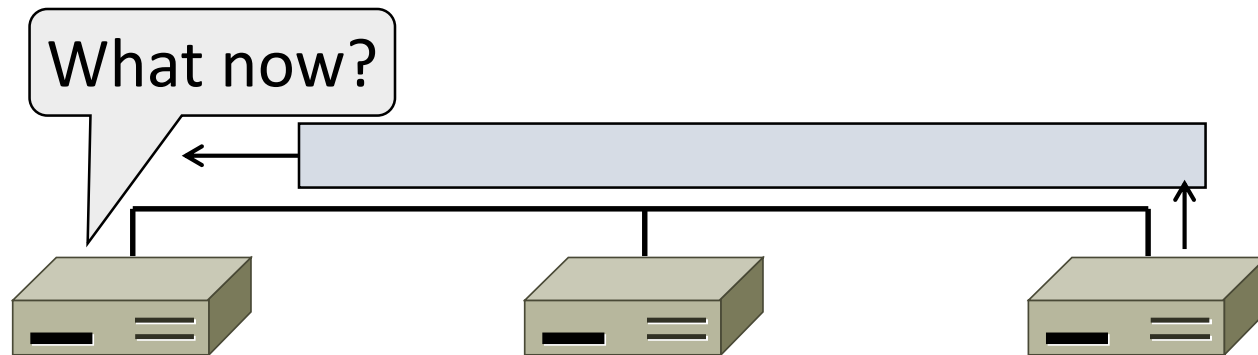
## CSMA/CD Complications (2)

- Impose a minimum frame length of  $2D$  seconds
  - So node can't finish before collision
  - Ethernet minimum frame is 64 bytes



# CSMA “Persistence”

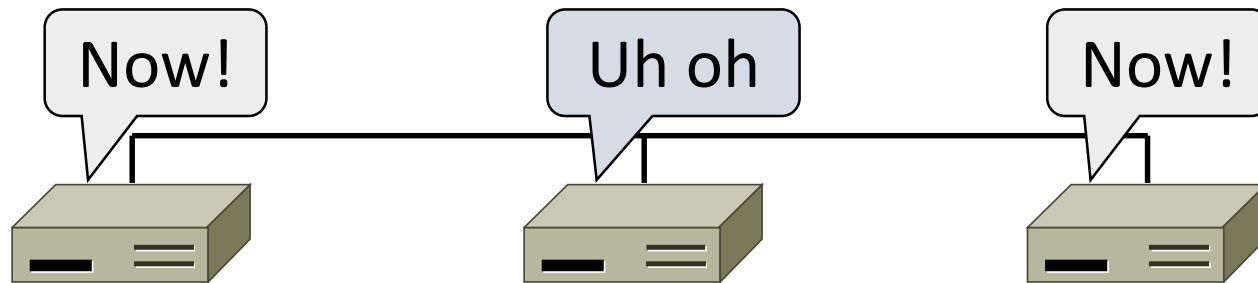
- What should a node do if another node is sending?



- Idea: Wait until it is done, and send

## CSMA “Persistence” (2)

- Problem is that multiple waiting nodes will queue up then collide
  - More load, more of a problem



# CSMA “Persistence” (3)

- Intuition for a better solution
  - If there are  $N$  queued senders, we want each to send next with probability  $1/N$

