Computer Networks

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Protocols and Layers

- <u>Protocols</u> and <u>layering</u> is the main structuring method used to divide up network functionality
 - Each instance of a protocol talks virtually to its <u>peer</u> using the protocol
 - Each instance of a protocol uses only the services of the lower layer

Protocols and Layers (3)

Protocols are horizontal, layers are vertical



Protocols and Layers (4)

Set of protocols in use is called a protocol stack



Protocols and Layers (6)

- Protocols you've probably heard of:
 - TCP, IP, 802.11, Ethernet, HTTP, SSL,
 DNS, ... and many more
- An example protocol stack
 - Used by a web browser on a host that is wirelessly connected to the Internet

(Browser
	HTTP
	ТСР
	IP
	802.11

Encapsulation

- <u>Encapsulation</u> is the mechanism used to effect protocol layering
 - Lower layer wraps higher layer content, adding its own information to make a new message for delivery
 - Like sending a letter in an envelope; postal service doesn't look inside

Encapsulation (3)

- Message "on the wire" begins to look like an onion
 - Lower layers are outermost



Encapsulation (4)



Advantage of Layering

• Information hiding and reuse



Advantage of Layering (2)

Information hiding and reuse



Advantage of Layering (3)

• Using information hiding to connect different systems



Advantage of Layering (4)

• Using information hiding to connect different systems



Disadvantage of Layering

• ??



Internet Reference Model

- A four layer model based on experience; omits some OSI layers and uses IP as the network layer.
 - 4 Application3 Transport
 - 2 Internet
 - 1 Link

- Programs that use network service
 - Provides end-to-end data delivery
 - Send packets over multiple networks
 - Send frames over a link

Internet Reference Model (3)

- IP is the "narrow waist" of the Internet
 - Supports many different links below and apps above



Layer-based Names (2)

• For devices in the network:



Layer-based Names (3)

• For devices in the network:

Proxy or middlebox or gateway



But they all look like this!



Scope of the Physical Layer

- Concerns how signals are used to transfer message bits over a link
 - Wires etc. carry analog signals
 - We want to send digital bits



Simple Link Model

- We'll end with an abstraction of a physical channel
 - <u>Rate</u> (or bandwidth, capacity, speed) in bits/second
 - Delay in seconds, related to length



- Other important properties:
 - Whether the channel is broadcast, and its error rate

Message Latency

- Latency is the delay to send a message over a link
 - Transmission delay: time to put M-bit message "on the wire"

- <u>Propagation delay</u>: time for bits to propagate across the wire

Combining the two terms we have:

Message Latency (2)

- <u>Latency</u> is the delay to send a message over a link
 - Transmission delay: time to put M-bit message "on the wire"

T-delay = M (bits) / Rate (bits/sec) = M/R seconds

- <u>Propagation delay</u>: time for bits to propagate across the wire

P-delay = Length / speed of signals = Length / ²/₃c = D seconds

- Combining the two terms we have: L = M/R + D

Metric Units

• The main prefixes we use:

Prefix	Exp.	prefix	exp.
K(ilo)	10 ³	m(illi)	10 ⁻³
M(ega)	10 ⁶	µ(micro)	10 ⁻⁶
G(iga)	10 ⁹	n(ano)	10 ⁻⁹

- Use powers of 10 for rates, 2 for storage
 - 1 Mbps = 1,000,000 bps, 1 KB = 2¹⁰ bytes
- "B" is for bytes, "b" is for bits

Latency Examples (2)

• "Dialup" with a telephone modem:

D = 5 ms, R = 56 kbps, M = 1250 bytes

- $L = 5 \text{ ms} + (1250 \text{ x8})/(56 \text{ x} 10^3) \text{ sec} = 184 \text{ ms}!$
- Broadband cross-country link:
 - D = 50 ms, R = 10 Mbps, M = 1250 bytes
 - $L = 50 \text{ ms} + (1250 \text{ x8}) / (10 \text{ x} 10^6) \text{ sec} = 51 \text{ ms}$
- A long link or a slow rate means high latency
 - Often, one delay component dominates

Bandwidth-Delay Product

• Messages take space on the wire!

• The amount of data in flight is the bandwidth-delay (BD) product

 $BD = R \times D$

- Measure in bits, or in messages
- Small for LANs, big for "long fat" pipes

Bandwidth-Delay Example (2)

- Fiber at home, cross-country R=40 Mbps, D=50 ms BD = 40 x 10^{6} x 50 x 10^{-3} bits = 2000 Kbit = 250 KB
- That's quite a lot of data "in the network"!

0101010010

Frequency Representation

 A signal over time can be represented by its frequency components (called Fourier analysis)



Effect of Less Bandwidth

• Fewer frequencies (=less bandwidth) degrades signal



Signals over a Wire (2)

• Example:

2: Attenuation:

Sent signal

• 3: Bandwidth:

4: Noise:

Signals over Wireless

- Signals transmitted on a carrier frequency, like fiber
- Travel at speed of light, spread out and attenuate faster than 1/dist²
- Multiple signals on the same frequency interfere at a receiver

Signals over Wireless (5)

- Various other effects too!
 - Wireless propagation is complex, depends on environment
- Some key effects are highly frequency dependent,
 - E.g., <u>multipath</u> at microwave frequencies

Wireless Multipath

- Signals bounce off objects and take multiple paths
 - Some frequencies attenuated at receiver, varies with location
 - Messes up signal; handled with sophisticated methods



Wireless

- Sender radiates signal over a region
 - In many directions, unlike a wire, to potentially many receivers
 - Nearby signals (same freq.) <u>interfere</u> at a receiver; need to coordinate use



UNITED

STATES FREQUENCY ALLOCATIONS THE RADIO SPECTRUM



ACTIVITY CODE



NON-BOVERNMENT EXCLUSIVE











Wireless (2)

 Microwave, e.g., 3G, and unlicensed (ISM) frequencies, e.g., WiFi, are widely used for computer networking



Topic

- We've talked about signals representing bits. How, exactly?
 - This is the topic of modulation



A Simple Modulation

- Let a high voltage (+V) represent a 1, and low voltage (-V) represent a 0
 - This is called NRZ (Non-Return to Zero)
A Simple Modulation (2)

- Let a high voltage (+V) represent a 1, and low voltage (-V) represent a 0
 - This is called NRZ (Non-Return to Zero)



Modulation



Topic

- How rapidly can we send information over a link?
 - <u>Nyquist</u> limit (~1924) »
 - <u>Shannon</u> capacity (1948) »
- Practical systems are devised to approach these limits



Key Channel Properties

- The bandwidth (B), signal strength (S), and noise strength (N)
 - B limits the rate of transitions
 - S and N limit how many signal levels we can distinguish

Nyquist Limit

• The maximum <u>symbol</u> rate is 2B

101010101010101010101

Thus if there are V signal levels, ignoring noise, the maximum bit rate is: R = 2B log₂V bits/sec

Claude Shannon (1916-2001)

- Father of information theory
 - "A Mathematical Theory of Communication", 1948
- Fundamental contributions to digital computers, security, and communications

Electromechanical mouse that "solves" mazes!



Credit: Courtesy MIT Museum

Shannon Capacity

- How many levels we can distinguish depends on S/N
 - Or SNR, the Signal-to-Noise Ratio
 - Note noise is random, hence some errors
- SNR given on a log-scale in deciBels:

$$-SNR_{dB} = 10log_{10}(S/N)$$



Shannon Capacity (2)

 Shannon limit is for capacity (C), the maximum information carrying rate of the channel:

 $C = B \log_2(1 + S/(BN)) bits/sec$

Wired/Wireless Perspective

- Wires, and Fiber
 - − Engineer link to have requisite SNR and B
 →Can fix data rate
- Wireless
 - Given B, but SNR varies greatly, e.g., up to 60 dB!
 →Can't design for worst case, must adapt data rate

Wired/Wireless Perspective (2)

- Wires, and Fiber Engineer SNR for data rate
 - − Engineer link to have requisite SNR and B
 →Can fix data rate
- Wireless
 Adapt data rate to SNR
 - Given B, but SNR varies greatly, e.g., up to 60 dB!
 →Can't design for worst case, must adapt data rate
 - Norst case, must adapt data rate

Putting it all together – DSL

- DSL (Digital Subscriber Line) is widely used for broadband; many variants offer 10s of Mbps
 - Reuses twisted pair telephone line to the home; it has up to ~2 MHz of bandwidth but uses only the lowest ~4 kHz







DSL (2)

- DSL uses passband modulation (called OFDM)
 - Separate bands for upstream and downstream (larger)
 - Modulation varies both amplitude and phase (called QAM)
 - High SNR, up to 15 bits/symbol, low SNR only 1 bit/symbol



Topic

- Some bits will be received in error due to noise. What can we do?
 - Detect errors with codes »
 - Correct errors with codes »
 - Retransmit lost frames Later
- Reliability is a concern that cuts across the layers – we'll see it again

Problem – Noise may flip received bits



Approach – Add Redundancy

- Error detection codes
 - Add <u>check bits</u> to the message bits to let some errors be detected
- Error correction codes
 - Add more <u>check bits</u> to let some errors be corrected
- Key issue is now to structure the code to detect many errors with few check bits and modest computation

Motivating Example

- A simple code to handle errors:
 - Send two copies! Error if different.

- How good is this code?
 - How many errors can it detect/correct?
 - How many errors will make it fail?

Motivating Example (2)

- We want to handle more errors with less overhead
 - Will look at better codes; they are applied mathematics
 - But, they can't handle all errors
 - And they focus on accidental errors (will look at secure hashes later)

Using Error Codes

• Codeword consists of D data plus R check bits (=systematic block code)

Data bits Check bits D R=fn(D) →

• Sender:

 Compute R check bits based on the D data bits; send the codeword of D+R bits

Using Error Codes (2)

- Receiver:
 - Receive D+R bits with unknown errors
 - Recompute R check bits based on the D data bits; error if R doesn't match R'



Intuition for Error Codes

• For D data bits, R check bits:



 Randomly chosen codeword is unlikely to be correct; overhead is low

R.W. Hamming (1915-1998)

- Much early work on codes:
 - "Error Detecting and Error Correcting Codes", BSTJ, 1950
- See also:
 - "You and Your Research", 1986



Source: IEEE GHN, © 2009 IEEE

Hamming Distance

 Distance is the number of bit flips needed to change D₁ to D₂

 <u>Hamming distance</u> of a code is the minimum distance between any pair of codewords

Hamming Distance (2)

- Error detection:
 - For a code of distance d+1, up to d errors will always be detected

Hamming Distance (3)

- Error correction:
 - For a code of distance 2d+1, up to d errors can always be corrected by mapping to the closest codeword

Topic

- Some bits may be received in error due to noise. How do we detect this?
 - Parity »
 - Checksums »
 - CRCs »
- Detection will let us fix the error, for example, by retransmission (later).

Simple Error Detection – Parity Bit

- Take D data bits, add 1 check bit that is the sum of the D bits
 - Sum is modulo 2 or XOR

Parity Bit (2)

- How well does parity work?
 - What is the distance of the code?

 How many errors will it detect/ correct?

• What about larger errors?

Checksums

- Idea: sum up data in N-bit words
 - Widely used in, e.g., TCP/IP/UDP

1500 bytes	16 bits
------------	---------

• Stronger protection than parity

Internet Checksum

- Sum is defined in 1s complement arithmetic (must add back carries)
 - And it's the negative sum
- "The checksum field is the 16 bit one's complement of the one's complement sum of all 16 bit words ..." RFC 791

Internet Checksum (2)

Sending:

- 1. Arrange data in 16-bit words
- 2. Put zero in checksum position, add
- 3. Add any carryover back to get 16 bits

4. Negate (complement) to get sum

0001 f203 f4f5 f6f7

Internet Checksum (3)

Sending:

Arrange data in 16-bit words
 Put zero in checksum position, add

3. Add any carryover back to get 16 bits

4. Negate (complement) to get sum



Internet Checksum (4)

Receiving:

Arrange data in 16-bit words
 Checksum will be non-zero, add

0001 f203 f4f5 f6f7 + 220d

3. Add any carryover back to get 16 bits

4. Negate the result and check it is 0

Internet Checksum (5)

Receiving:

Arrange data in 16-bit words
 Checksum will be non-zero, add

3. Add any carryover back to get 16 bits

4. Negate the result and check it is 0

+	0001 f203 f4f5 f6f7 220d	
2	fffd	
+	fffd 2	
	ffff	
	0000	

Internet Checksum (6)

- How well does the checksum work?
 - What is the distance of the code?
 - How many errors will it detect/ correct?

• What about larger errors?

Cyclic Redundancy Check (CRC)

- Even stronger protection
 - Given n data bits, generate k check
 bits such that the n+k bits are evenly
 divisible by a generator C
- Example with numbers:
 - n = 302, k = one digit, C = 3

CRCs (2)

- The catch:
 - It's based on mathematics of finite fields, in which "numbers" represent polynomials

- e.g, 10011010 is
$$x^7 + x^4 + x^3 + x^1$$

- What this means:
 - We work with binary values and operate using modulo 2 arithmetic
CRCs (3)

- Send Procedure:
- 1. Extend the n data bits with k zeros
- 2. Divide by the generator value C
- 3. Keep remainder, ignore quotient
- 4. Adjust k check bits by remainder
- Receive Procedure:
- 1. Divide and check for zero remainder

CRCs (4)

Check bits: $C(x)=x^{4}+x^{1}+1$ C = 10011k = 4



CRCs (6)

- Protection depend on generator
 - Standard CRC-32 is 10000010
 01100000 10001110 110110111
- Properties:
 - HD=4, detects up to triple bit errors
 - Also odd number of errors
 - And bursts of up to k bits in error
 - Not vulnerable to systematic errors like checksums

Error Detection in Practice

- CRCs are widely used on links
 - Ethernet, 802.11, ADSL, Cable ...
- Checksum used in Internet
 IP, TCP, UDP ... but it is weak
- Parity

Is little used

Topic

- Two strategies to handle errors:
- Detect errors and retransmit frame (Automatic Repeat reQuest, ARQ)
- 2. Correct errors with an error correcting code

Context on Reliability

• Where in the stack should we place reliability functions?

Application Transport Network Link Physical



Context on Reliability (2)

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



ARQ

- 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 <u>Stop-and-Wait</u>

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 <u>RTT</u> (=2D)



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