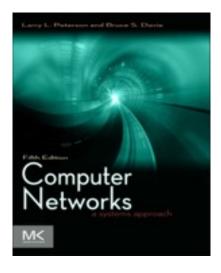


# Computer Networks: A Systems Approach, 5e

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Chapter 2

**Getting Connected** 



#### **Problems**

- In Chapter 1 we saw networks consists of links interconnecting nodes. How to connect two nodes together?
- We also introduced the concept of "cloud" abstractions to represent a network without revealing its internal complexities. How to connect a host to a cloud?



#### **Chapter Outline**

- Perspectives on Connecting nodes
- Encoding
- Framing
- Error Detection
- Reliable Transmission
- Ethernet and Multiple Access Networks
- Wireless Networks



#### **Chapter Goal**

- Exploring different communication medium over which we can send data
- Understanding the issue of encoding bits onto transmission medium so that they can be understood by the receiving end
- Discussing the matter of delineating the sequence of bits transmitted over the link into complete messages that can be delivered to the end node
- Discussing different technique to detect transmission errors and take the appropriate action

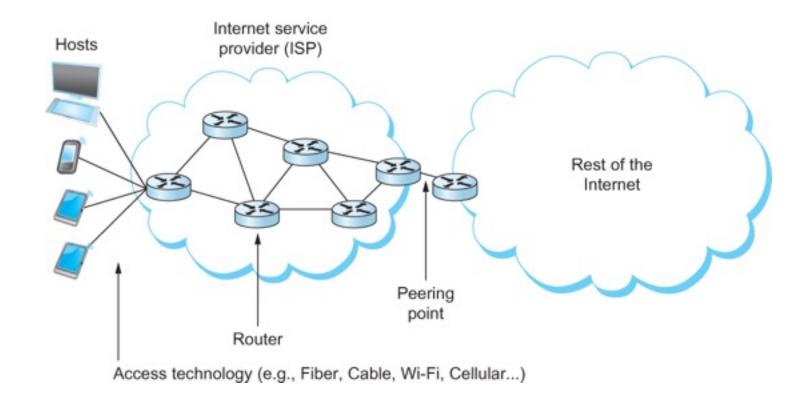


## **Chapter Goal (contd.)**

- Discussing the issue of making the links reliable in spite of transmission problems
- Introducing Media Access Control Problem
- Introducing Carrier Sense Multiple Access (CSMA) networks
- Introducing Wireless Networks with different available technologies and protocol



#### **Perspectives on Connecting**



An end-user's view of the Internet



#### Link Capacity and Shannon-Hartley Theorem

- Gives the upper bound to the capacity of a link in terms of bits per second (bps) as a function of signal-to-noise ratio of the link measured in decibels (dB).
- $C = Blog_2(1+S/N)$ 
  - Where B = 3300 300 = 3000Hz, S is the signal power, N the average noise.
  - The signal to noise ratio (S/N) is measured in decibels is related to  $dB = 10 \times log_{10}(S/N)$ . If there is 30dB of noise then S/N = 1000.
  - Now C =  $3000 \times \log_2(1001) = 30$ kbps.
  - How can we get 56kbps?



- All practical links rely on some sort of electromagnetic radiation propagating through a medium or, in some cases, through free space
- One way to characterize links, then, is by the medium they use
  - Typically copper wire in some form (as in Digital Subscriber Line (DSL) and coaxial cable),
  - Optical fiber (as in both commercial fiber-to-the home services and many long-distance links in the Internet's backbone), or
  - Air/free space (for wireless links)



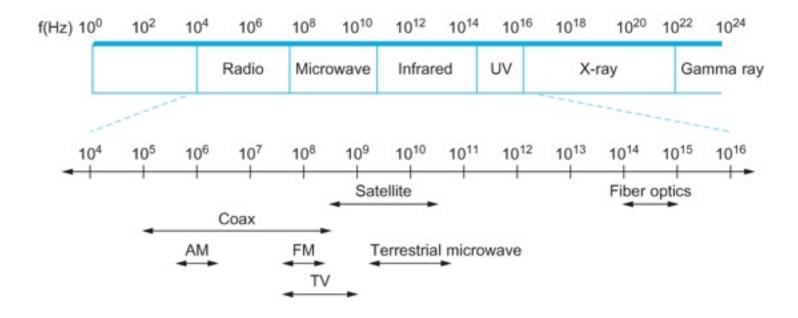
- Another important link characteristic is the *frequency* 
  - Measured in hertz, with which the electromagnetic waves oscillate
- Distance between the adjacent pair of maxima or minima of a wave measured in meters is called wavelength

  • Speed of light divided by frequency gives the wavelength.

  • Frequency on a copper cable range from 300Hz to 3300Hz;

  - Wavelength for 300Hz wave through copper is speed of light on a copper / frequency
  - $2/3 \times 3 \times 10^8 / 300 = 667 \times 10^3$  meters.
- Placing binary data on a signal is called *encoding*.
- Modulation involves modifying the signals in terms of their frequency, amplitude, and phase.





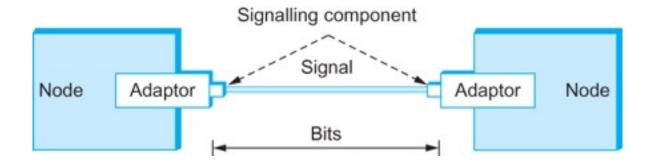
Electromagnetic spectrum



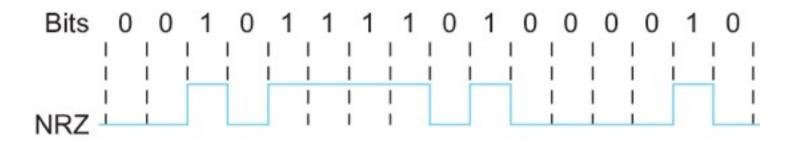
Service	Bandwidth (typical)
Dial-up	28–56 kbps
ISDN	64–128 kbps
DSL	128 kbps-100 Mbps
CATV (cable TV)	1–40 Mbps
FTTH (fibre to the home)	50 Mbps-1 Gbps

Common services available to connect your home





Signals travel between signaling components; bits flow between adaptors



NRZ encoding of a bit stream



- Problem with NRZ
  - Baseline wander
    - The receiver keeps an average of the signals it has seen so far
    - Uses the average to distinguish between low and high signal
    - When a signal is significantly low than the average, it is 0, else it is 1
    - Too many consecutive 0's and 1's cause this average to change, making it difficult to detect



- Problem with NRZ
  - Clock recovery
    - Frequent transition from high to low or vice versa are necessary to enable clock recovery
    - Both the sending and decoding process is driven by a clock
    - Every clock cycle, the sender transmits a bit and the receiver recovers a bit
    - The sender and receiver have to be precisely synchronized



#### NRZI

- Non Return to Zero Inverted
- Sender makes a transition from the current signal to encode 1 and stay at the current signal to encode 0
- Solves for consecutive 1's

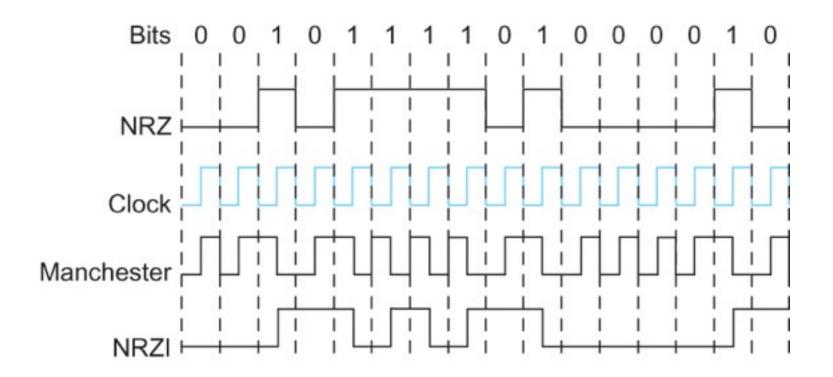


- Manchester encoding
  - Merging the clock with signal by transmitting Ex-OR of the NRZ encoded data and the clock
  - Clock is an internal signal that alternates from low to high, a low/high pair is considered as one clock cycle
  - In Manchester encoding
    - 0: low→ high transition
    - 1: high → low transition



- Problem with Manchester encoding
  - Doubles the rate at which the signal transitions are made on the link
    - Which means the receiver has half of the time to detect each pulse of the signal
  - The rate at which the signal changes is called the link's baud rate
  - In Manchester the bit rate is half the baud rate





Different encoding strategies



#### 4B/5B encoding

- Insert extra bits into bit stream so as to break up the long sequence of 0's and 1's
- Every 4-bits of actual data are encoded in a 5-bit code that is transmitted to the receiver
- 5-bit codes are selected in such a way that each one has no more than one leading 0(zero) and no more than two trailing 0's.
- No pair of 5-bit codes results in more than three consecutive 0's



4B/5B encoding

 $0000 \rightarrow 11110$ 

 $0001 \rightarrow 01001$ 

 $0010 \rightarrow 10100$ 

. .

. .

1111 → 11101

16 left

11111 – when the line is idle

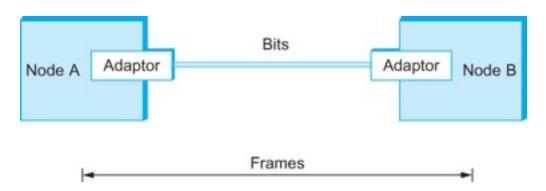
00000 - when the line is dead

00100 - to mean halt

13 left: 7 invalid, 6 for various control signals



- We are focusing on packet-switched networks, which means that blocks of data (called *frames* at this level), not bit streams, are exchanged between nodes.
- It is the network adaptor that enables the nodes to exchange frames.



Bits flow between adaptors, frames between hosts



- When node A wishes to transmit a frame to node
   B, it tells its adaptor to transmit a frame from the
   node's memory. This results in a sequence of bits
   being sent over the link.
- The adaptor on node B then collects together the sequence of bits arriving on the link and deposits the corresponding frame in B's memory.
- Recognizing exactly what set of bits constitute a frame—that is, determining where the frame begins and ends—is the central challenge faced by the adaptor

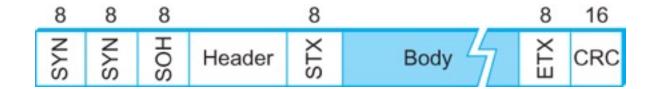


- **Byte-oriented Protocols** 
  - To view each frame as a collection of bytes (characters) rather than bitsBISYNC (Binary Synchronous Communication)
  - Protocol
    - Developed by IBM (late 1960)
  - DDCMP (Digital Data Communication Protocol)
     Used in DECNet



- BISYNC sentinel approach
  - Frames transmitted beginning with leftmost field
  - Beginning of a frame is denoted by sending a special SYN (synchronize) character
  - Data portion of the frame is contained between special sentinel character STX (start of text) and ETX (end of text)
  - SOH : Start of Header
  - DLE : Data Link Escape
  - CRC: Cyclic Redundancy Check



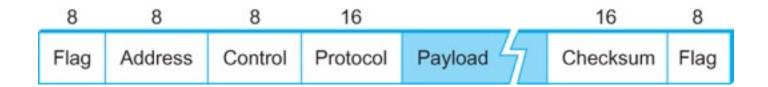


**BISYNC Frame Format** 



- Recent PPP which is commonly run over Internet links uses sentinel approach
  - Special start of text character denoted as Flag
     0 1 1 1 1 1 0
  - Address, control : default numbers
  - Protocol for demux : IP / IPX
  - Payload : negotiated (1500 bytes)
  - Checksum: for error detection



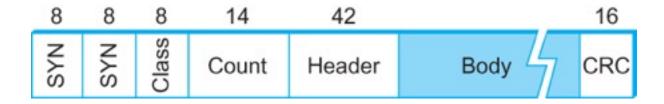


**PPP Frame Format** 



- Byte-counting approach
  - DDCMP
  - count: how many bytes are contained in the frame body
  - If count is corrupted
    - Framing error





**DDCMP Frame Format** 



- Bit-oriented Protocol
  - HDLC: High Level Data Link Control
    - Beginning and Ending Sequences
       0 1 1 1 1 1 1 0



**HDLC Frame Format** 



- HDLC Protocol
  - On the sending side, any time five consecutive 1's have been transmitted from the body of the message (i.e. excluding when the sender is trying to send the distinguished 01111110 sequence)
    - The sender inserts 0 before transmitting the next bit



- HDLC Protocol
  - On the receiving side
    - 5 consecutive 1's
      - Next bit 0 : Stuffed, so discard it
        - 1 : Either End of the frame marker
          Or Error has been introduced in the

#### bitstream

Look at the next bit

If  $0 (011111110) \rightarrow \text{End of the frame marker}$ 

If 1 (011111111 ) → Error, discard the whole frame

The receiver needs to wait

for next

01111110 before it can start receiving again



- Bit errors are introduced into frames
  - Because of electrical interference and thermal noises
- Detecting Error
- Correction Error
- Two approaches when the recipient detects an error
  - Notify the sender that the message was corrupted, so the sender can send again.
    - If the error is rare, then the retransmitted message will be error-free
  - Using some error correct detection and correction algorithm, the receiver reconstructs the message



- Common technique for detecting transmission error
  - CRC (Cyclic Redundancy Check)
    - Used in HDLC, DDCMP, CSMA/CD, Token Ring
  - Other approaches
    - Two Dimensional Parity (BISYNC)
    - Checksum (IP)



- Basic Idea of Error Detection
  - To add redundant information to a frame that can be used to determine if errors have been introduced
  - Imagine (Extreme Case)
    - Transmitting two complete copies of data
      - Identical → No error
      - Differ → Error
      - Poor Scheme ???
        - n bit message, n bit redundant information
        - Error can go undetected
    - In general, we can provide strong error detection technique
      - k redundant bits, n bits message, k << n</li>
      - In Ethernet, a frame carrying up to 12,000 bits of data requires only 32bit CRC



- Extra bits are redundant
  - They add no new information to the message
  - Derived from the original message using some algorithm
  - Both the sender and receiver know the algorithm

Sender

Receiver

m	r

m r

Receiver computes *r* using *m* 

If they match, no error

### **Two-dimensional parity**

- Two-dimensional parity is exactly what the name suggests
- It is based on "simple" (one-dimensional) parity, which usually involves adding one extra bit to a 7bit code to balance the number of 1s in the byte.
   For example,
  - Odd parity sets the eighth bit to 1 if needed to give an odd number of 1s in the byte, and
  - Even parity sets the eighth bit to 1 if needed to give an even number of 1s in the byte

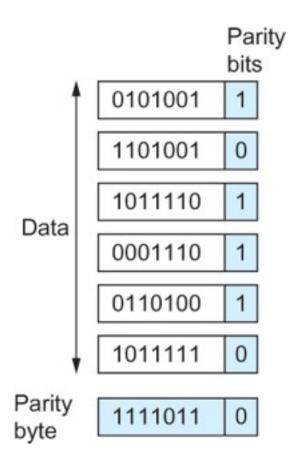


### **Two-dimensional parity**

- Two-dimensional parity does a similar calculation for each bit position across each of the bytes contained in the frame
- This results in an extra parity byte for the entire frame, in addition to a parity bit for each byte
- Two-dimensional parity catches all 1-, 2-, and 3bit errors and most 4-bit errors



### **Two-dimensional parity**



**Two Dimensional Parity** 



- Not used at the link level
- Add up all the words that are transmitted and then transmit the result of that sum
  - The result is called the checksum
- The receiver performs the same calculation on the received data and compares the result with the received checksum
- If any transmitted data, including the checksum itself, is corrupted, then the results will not match, so the receiver knows that an error occurred



- Consider the data being checksummed as a sequence of 16-bit integers.
- Add them together using 16-bit ones complement arithmetic (explained next slide) and then take the ones complement of the result.
- That 16-bit number is the checksum



- In ones complement arithmetic, a negative integer –x is represented as the complement of x;
  - Each bit of x is inverted.
- When adding numbers in ones complement arithmetic, a carryout from the most significant bit needs to be added to the result.



- Consider, for example, the addition of -5 and -3 in ones complement arithmetic on 4-bit integers
   +5 is 0101, so -5 is 1010; +3 is 0011, so -3 is 1100
- If we add 1010 and 1100 ignoring the carry, we get 0110
- In ones complement arithmetic, the fact that this operation caused a carry from the most significant bit causes us to increment the result, giving 0111, which is the ones complement representation of −8 (obtained by inverting the bits in 1000), as we would expect



- Reduce the number of extra bits and maximize protection
- Given a bit string 110001 we can associate a polynomial on a single variable x for it.

```
1.x^5+1.x^4+0.x^3+0.x^2+0.x^1+1.x^0 = x^5+x^4+1 and the degree is 5.
```

A k-bit frame has a maximum degree of k-1

 Let M(x) be a message polynomial and C(x) be a generator polynomial.



- Let M(x)/C(x) leave a remainder of 0.
- When M(x) is sent and M'(x) is received we have M'(x) = M(x)+E(x)
- The receiver computes M'(x)/C(x) and if the remainder is nonzero, then an error has occurred.
- The only thing the sender and the receiver should know is C(x).



#### Polynomial Arithmetic Modulo 2

- Any polynomial B(x) can be divided by a divisor polynomial C(x) if B(x) is of higher degree than C(x).
- Any polynomial B(x) can be divided once by a divisor polynomial C(x) if B(x) is of the same degree as C(x).
- The remainder obtained when B(x) is divided by C(x) is obtained by subtracting C(x) from B(x).
- To subtract C(x) from B(x), we simply perform the exclusive-OR (XOR) operation on each pair of matching coefficients.

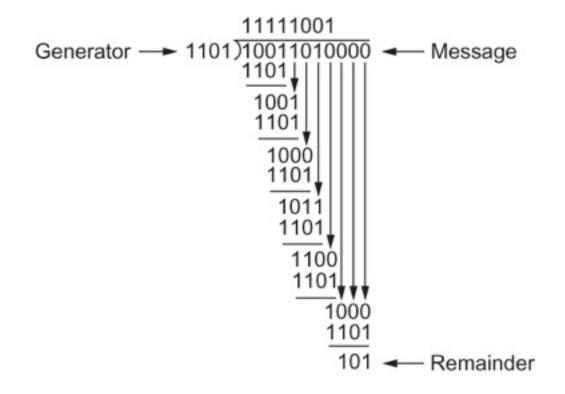


- Let M(x) be a frame with m bits and let the generator polynomial have less than m bits say equal to r.
- Let r be the degree of C(x). Append r zero bits to the low-order end of the frame, so it now contains m+r bits and corresponds to the polynomial x<sup>r</sup>M(x).



- Divide the bit string corresponding to x<sup>r</sup>M(x) by the bit string corresponding to C(x) using modulo 2 division.
- Subtract the remainder (which is always r or fewer bits) from the string corresponding to x<sup>r</sup>M(x) using modulo 2 subtraction (addition and subtraction are the same in modulo 2).
- The result is the checksummed frame to be transmitted. Call it polynomial M'(x).





CRC Calculation using Polynomial Long Division



- Properties of Generator Polynomial
  - Let P(x) represent what the sender sent and P(x) + E(x) is the received string. A 1 in E(x) represents that in the corresponding position in P(x) the message the bit is flipped.
  - We know that P(x)/C(x) leaves a remainder of 0, but if E(x)/C(x) leaves a remainder of 0, then either E(x) = 0 or C(x) is factor of E(x).
  - When C(x) is a factor of E(x) we have problem; errors go unnoticed.
  - If there is a single bit error then  $E(x) = x^i$ , where *i* determines the bit in error. If C(x) contains two or more terms it will never divide E(x), so all single bit errors will be detected.



- Properties of Generator Polynomial
  - In general, it is possible to prove that the following types of errors can be detected by a C(x) with the stated properties
    - All single-bit errors, as long as the x<sup>k</sup> and x<sup>0</sup> terms have nonzero coefficients.
    - All double-bit errors, as long as C(x) has a factor with at least three terms.
    - Any odd number of errors, as long as C(x) contains the factor (x+1).
    - Any "burst" error (i.e., sequence of consecutive error bits) for which the length of the burst is less than k bits. (Most burst errors of larger than k bits can also be detected.)



- Six generator polynomials that have become international standards are:
  - CRC-8 =  $x^8 + x^2 + x + 1$
  - CRC-10 =  $x^{10}+x^9+x^5+x^4+x+1$
  - CRC-12 =  $x^{12}+x^{11}+x^3+x^2+x+1$
  - CRC-16 =  $x^{16}+x^{15}+x^2+1$
  - CRC-CCITT =  $x^{16}+x^{12}+x^{5}+1$
  - CRC-32 =  $x^{32}+x^{26}+x^{23}+x^{22}+x^{16}+x^{12}+x^{11}+x^{10}+x^{8}+x^{7}+x^{5}+x^{4}+x^{2}+x+1$



#### **Reliable Transmission**

- CRC is used to detect errors.
- Some error codes are strong enough to correct errors.
- The overhead is typically too high.
- Corrupt frames must be discarded.
- A link-level protocol that wants to deliver frames reliably must recover from these discarded frames.
- This is accomplished using a combination of two fundamental mechanisms
  - Acknowledgements and Timeouts



#### **Reliable Transmission**

- An acknowledgement (ACK for short) is a small control frame that a protocol sends back to its peer saying that it has received the earlier frame.
  - A control frame is a frame with header only (no data).
- The receipt of an acknowledgement indicates to the sender of the original frame that its frame was successfully delivered.



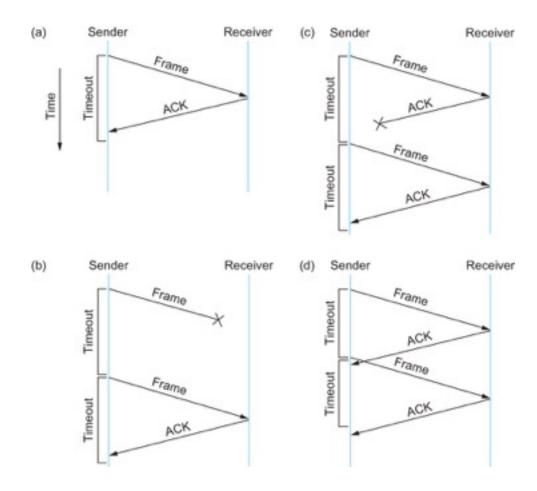
#### **Reliable Transmission**

- If the sender does not receive an acknowledgment after a reasonable amount of time, then it retransmits the original frame.
- The action of waiting a reasonable amount of time is called a *timeout*.
- The general strategy of using acknowledgements and timeouts to implement reliable delivery is sometimes called Automatic Repeat reQuest (ARQ).



- Idea of stop-and-wait protocol is straightforward
  - After transmitting one frame, the sender waits for an acknowledgement before transmitting the next frame.
  - If the acknowledgement does not arrive after a certain period of time, the sender times out and retransmits the original frame





Timeline showing four different scenarios for the stop-and-wait algorithm.

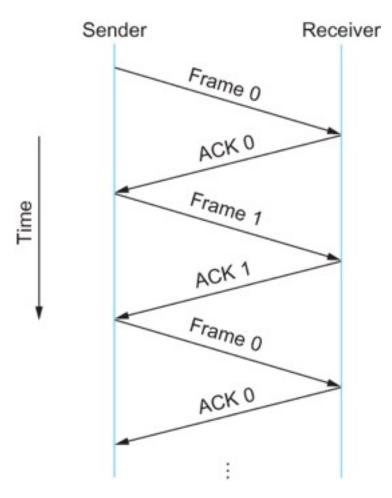
(a) The ACK is received before the timer expires; (b) the original frame is lost; (c) the

ACK is lost; (d) the timeout fires too soon



- If the acknowledgment is lost or delayed in arriving
  - The sender times out and retransmits the original frame, but the receiver will think that it is the next frame since it has correctly received and acknowledged the first frame
  - As a result, duplicate copies of frames will be delivered
- How to solve this
  - Use 1 bit sequence number (0 or 1)
  - When the sender retransmits frame 0, the receiver can determine that it is seeing a second copy of frame 0 rather than the first copy of frame 1 and therefore can ignore it (the receiver still acknowledges it, in case the first acknowledgement was lost)



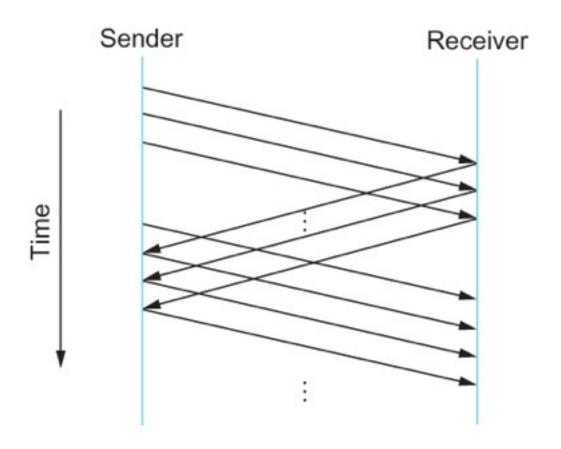


Timeline for stop-and-wait with 1-bit sequence number



- The sender has only one outstanding frame on the link at a time
  - This may be far below the link's capacity
- Consider a 1.5 Mbps link with a 45 ms RTT
  - The link has a delay × bandwidth product of 67.5 Kb or approximately 8 KB
  - Since the sender can send only one frame per RTT and assuming a frame size of 1 KB
  - Maximum Sending rate
    - Bits per frame  $\div$  Time per frame =  $1024 \times 8 \div 0.045 = 182$  Kbps Or about one-eighth of the link's capacity
  - To use the link fully, then sender should transmit up to eight frames before having to wait for an acknowledgement





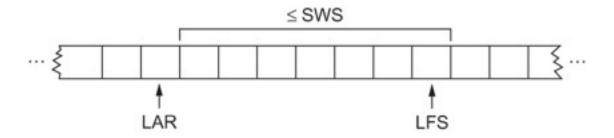
**Timeline for Sliding Window Protocol** 



- Sender assigns a sequence number denoted as SeqNum to each frame.
  - Assume it can grow infinitely large
- Sender maintains three variables
  - Sending Window Size (SWS)
    - Upper bound on the number of outstanding (unacknowledged) frames that the sender can transmit
  - Last Acknowledgement Received (LAR)
    - Sequence number of the last acknowledgement received
  - Last Frame Sent (LFS)
    - Sequence number of the last frame sent



 Sender also maintains the following invariant LFS – LAR ≤ SWS



Sliding Window on Sender



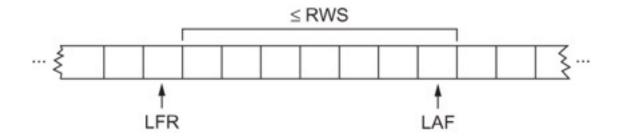
- When an acknowledgement arrives
  - the sender moves LAR to right, thereby allowing the sender to transmit another frame
- Also the sender associates a timer with each frame it transmits
  - It retransmits the frame if the timer expires before the ACK is received
- Note that the sender has to be willing to buffer up to SWS frames
  - WHY?



- Receiver maintains three variables
  - Receiving Window Size (RWS)
    - Upper bound on the number of out-of-order frames that the receiver is willing to accept
  - Largest Acceptable Frame (LAF)
    - Sequence number of the largest acceptable frame
  - Last Frame Received (LFR)
    - Sequence number of the last frame received



Receiver also maintains the following invariant
 LAF – LFR ≤ RWS



Sliding Window on Receiver



- When a frame with sequence number SeqNum arrives, what does the receiver do?
  - If SeqNum ≤ LFR or SeqNum > LAF
    - Discard it (the frame is outside the receiver window)
  - If LFR < SeqNum ≤ LAF</li>
    - Accept it
    - Now the receiver needs to decide whether or not to send an ACK



- Let SeqNumToAck
  - Denote the largest sequence number not yet acknowledged, such that all frames with sequence number less than or equal to SeqNumToAck have been received
- The receiver acknowledges the receipt of SeqNumToAck even if high-numbered packets have been received
  - This acknowledgement is said to be cumulative.
- The receiver then sets
  - LFR = SeqNumToAck and adjusts
  - LAF = LFR + RWS



For example, suppose LFR = 5 and RWS = 4

(i.e. the last ACK that the receiver sent was for seq. no. 5)

 $\Rightarrow$  LAF = 9

If frames 7 and 8 arrive, they will be buffered because they are within the receiver window

But no ACK will be sent since frame 6 is yet to arrive Frames 7 and 8 are out of order

Frame 6 arrives (it is late because it was lost first time and had to be retransmitted)

Now Receiver Acknowledges Frame 8 and bumps LFR to 8 and LAF to 12



#### **Issues with Sliding Window Protocol**

- When timeout occurs, the amount of data in transit decreases
  - Since the sender is unable to advance its window
- When the packet loss occurs, this scheme is no longer keeping the pipe full
  - The longer it takes to notice that a packet loss has occurred, the more severe the problem becomes
- How to improve this
  - Negative Acknowledgement (NAK)
  - Additional Acknowledgement
  - Selective Acknowledgement



#### <u>Issues with Sliding Window Protocol</u>

- Negative Acknowledgement (NAK)
  - Receiver sends NAK for frame 6 when frame 7 arrive (in the previous example)
    - However this is unnecessary since sender's timeout mechanism will be sufficient to catch the situation
- Additional Acknowledgement
  - Receiver sends additional ACK for frame 5 when frame 7 arrives

    Sender uses duplicate ACK as a clue for frame loss
- Selective Acknowledgement
  - Receiver will acknowledge exactly those frames it has received, rather than the highest number frames
    Receiver will acknowledge frames 7 and 8
    Sender knows frame 6 is lost

    - Sender can keep the pipe full (additional complexity)



#### **Issues with Sliding Window Protocol**

#### How to select the window size

- SWS is easy to compute
  - Delay × Bandwidth
- RWS can be anything
  - Two common setting

```
• RWS = 1
```

No buffer at the receiver for frames that

arrive out of

order

RWS = SWS

The receiver can buffer frames that the

sender

transmits

It does not make any sense to keep RWS > SWS

WHY?



- Finite Sequence Number
  - Frame sequence number is specified in the header field
    - Finite size
      - 3 bit: eight possible sequence number: 0, 1, 2, 3, 4, 5, 6, 7
    - It is necessary to wrap around



- How to distinguish between different incarnations of the same sequence number?
  - Number of possible sequence number must be larger than the number of outstanding frames allowed
    - Stop and Wait: One outstanding frame
      - 2 distinct sequence number (0 and 1)
    - Let MaxSeqNum be the number of available sequence numbers
    - SWS + 1 ≤ MaxSeqNum
      - Is this sufficient?



```
SWS + 1 ≤ MaxSeqNum
```

- Is this sufficient?
- Depends on RWS
- If RWS = 1, then sufficient
- If RWS = SWS, then not good enough
- For example, we have eight sequence numbers

```
0, 1, 2, 3, 4, 5, 6, 7
RWS = SWS = 7
Sender sends 0, 1, ..., 6
Receiver receives 0, 1, ..., 6
Receiver acknowledges 0, 1, ..., 6
ACK (0, 1, ..., 6) are lost
Sender retransmits 0, 1, ..., 6
Receiver is expecting 7, 0, ..., 5
```



To avoid this, If RWS = SWS

SWS < (MaxSeqNum + 1)/2



- Serves three different roles
  - Reliable
  - Preserve the order
    - Each frame has a sequence number
    - The receiver makes sure that it does not pass a frame up to the next higher-level protocol until it has already passed up all frames with a smaller sequence number
  - Frame control
    - Receiver is able to throttle the sender
      - Keeps the sender from overrunning the receiver
        - From transmitting more data than the receiver is able to process



- Most successful local area networking technology of last 20 years.
- Developed in the mid-1970s by researchers at the Xerox Palo Alto Research Centers (PARC).
- Uses CSMA/CD technology
  - Carrier Sense Multiple Access with Collision Detection.
  - A set of nodes send and receive frames over a shared link.
  - Carrier sense means that all nodes can distinguish between an idle and a busy link.
  - Collision detection means that a node listens as it transmits and can therefore detect when a frame it is transmitting has collided with a frame transmitted by another node.

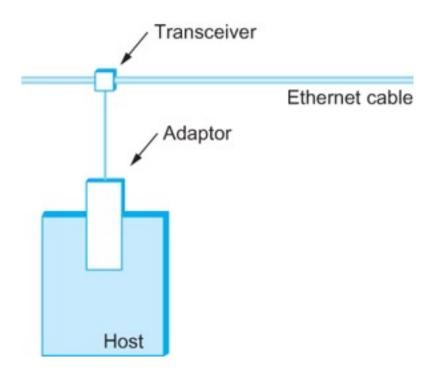


- Uses ALOHA (packet radio network) as the root protocol
  - Developed at the University of Hawaii to support communication across the Hawaiian Islands.
  - For ALOHA the medium was atmosphere, for Ethernet the medium is a coax cable.
- DEC and Intel joined Xerox to define a 10-Mbps Ethernet standard in 1978.
- This standard formed the basis for IEEE standard 802.3
- More recently 802.3 has been extended to include a 100-Mbps version called Fast Ethernet and a 1000-Mbps version called Gigabit Ethernet.



- An Ethernet segment is implemented on a coaxial cable of up to 500 m.
  - This cable is similar to the type used for cable TV except that it typically has an impedance of 50 ohms instead of cable TV's 75 ohms.
- Hosts connect to an Ethernet segment by tapping into it.
- A transceiver (a small device directly attached to the tap)
  detects when the line is idle and drives signal when the host is
  transmitting.
- The transceiver also receives incoming signal.
- The transceiver is connected to an Ethernet adaptor which is plugged into the host.
- The protocol is implemented on the adaptor.



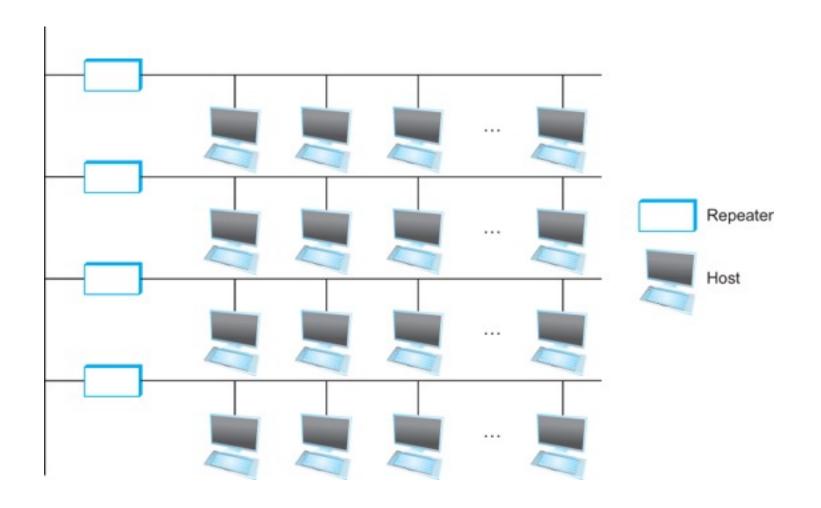


Ethernet transceiver and adaptor



- Multiple Ethernet segments can be joined together by repeaters.
- A repeater is a device that forwards digital signals.
- No more than four repeaters may be positioned between any pair of hosts.
  - An Ethernet has a total reach of only 2500 m.





**Ethernet repeater** 



- Any signal placed on the Ethernet by a host is broadcast over the entire network
  - Signal is propagated in both directions.
  - Repeaters forward the signal on all outgoing segments.
  - Terminators attached to the end of each segment absorb the signal.
- Ethernet uses Manchester encoding scheme.

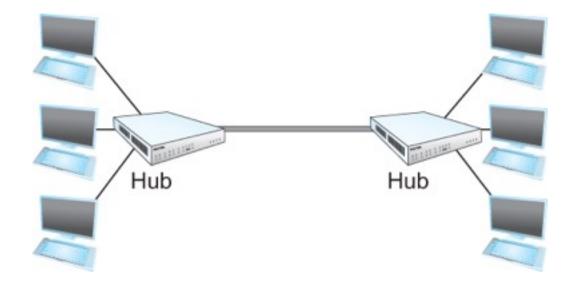


- New Technologies in Ethernet
  - Instead of using coax cable, an Ethernet can be constructed from a thinner cable known as 10Base2 (the original was 10Base5)
    - 10 means the network operates at 10 Mbps
    - Base means the cable is used in a baseband system
    - 2 means that a given segment can be no longer than 200 m



- New Technologies in Ethernet
  - Another cable technology is 10BaseT
    - T stands for twisted pair
    - Limited to 100 m in length
  - With 10BaseT, the common configuration is to have several point to point segments coming out of a multiway repeater, called *Hub*





**Ethernet Hub** 

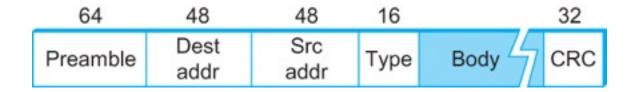


### **Access Protocol for Ethernet**

- The algorithm is commonly called Ethernet's Media Access Control (MAC).
  - It is implemented in Hardware on the network adaptor.
- Frame format
  - Preamble (64bit): allows the receiver to synchronize with the signal (sequence of alternating 0s and 1s).
  - Host and Destination Address (48bit each).
  - Packet type (16bit): acts as demux key to identify the higher level protocol.
  - Data (up to 1500 bytes)
    - Minimally a frame must contain at least 46 bytes of data.
    - Frame must be long enough to detect collision.
  - CRC (32bit)



### **Ethernet Frame**



**Ethernet Frame Format** 



- Each host on an Ethernet (in fact, every Ethernet host in the world) has a unique Ethernet Address.
- The address belongs to the adaptor, not the host.
  - It is usually burnt into ROM.
- Ethernet addresses are typically printed in a human readable format
  - As a sequence of six numbers separated by colons.
  - Each number corresponds to 1 byte of the 6 byte address and is given by a pair of hexadecimal digits, one for each of the 4-bit nibbles in the byte
  - Leading 0s are dropped.
  - For example, 8:0:2b:e4:b1:2 is

- To ensure that every adaptor gets a unique address, each manufacturer of Ethernet devices is allocated a different prefix that must be prepended to the address on every adaptor they build
  - AMD has been assigned the 24bit prefix 8:0:20



- Each frame transmitted on an Ethernet is received by every adaptor connected to that Ethernet.
- Each adaptor recognizes those frames addressed to its address and passes only those frames on to the host.
- In addition, to unicast address, an Ethernet address consisting of all 1s is treated as a broadcast address.
  - All adaptors pass frames addressed to the broadcast address up to the host.
- Similarly, an address that has the first bit set to 1 but is not the broadcast address is called a multicast address.
  - A given host can program its adaptor to accept some set of multicast addresses.



- To summarize, an Ethernet adaptor receives all frames and accepts
  - Frames addressed to its own address
  - Frames addressed to the broadcast address
  - Frames addressed to a multicast addressed if it has been instructed



- When the adaptor has a frame to send and the line is idle, it transmits the frame immediately.
  - The upper bound of 1500 bytes in the message means that the adaptor can occupy the line for a fixed length of time.
- When the adaptor has a frame to send and the line is busy, it waits for the line to go idle and then transmits immediately.
- The Ethernet is said to be 1-persistent protocol because an adaptor with a frame to send transmits with probability 1 whenever a busy line goes idle.



- Since there is no centralized control it is possible for two (or more) adaptors to begin transmitting at the same time,
  - Either because both found the line to be idle,
  - Or, both had been waiting for a busy line to become idle.
- When this happens, the two (or more) frames are said to be collide on the network.



- Since Ethernet supports collision detection, each sender is able to determine that a collision is in progress.
- At the moment an adaptor detects that its frame is colliding with another, it first makes sure to transmit a 32bit jamming sequence and then stops transmission.
  - Thus, a transmitter will minimally send 96 bits in the case of collision
    - 64-bit preamble + 32-bit jamming sequence



- One way that an adaptor will send only 96 bit (called a runt frame) is if the two hosts are close to each other.
- Had they been farther apart,
  - They would have had to transmit longer, and thus send more bits, before detecting the collision.



- The worst case scenario happens when the two hosts are at opposite ends of the Ethernet.
- To know for sure that the frame its just sent did not collide with another frame, the transmitter may need to send as many as 512 bits.
  - Every Ethernet frame must be at least 512 bits (64 bytes) long.
    - 14 bytes of header + 46 bytes of data + 4 bytes of CRC

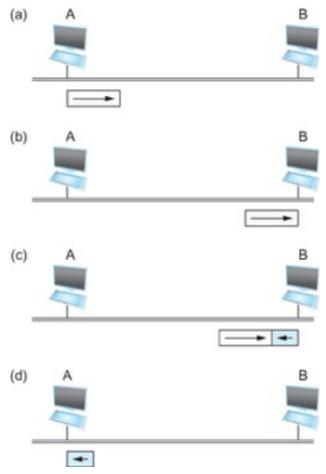


- Why 512 bits?
  - Why is its length limited to 2500 m?
- The farther apart two nodes are, the longer it takes for a frame sent by one to reach the other, and the network is vulnerable to collision during this time



- A begins transmitting a frame at time t
- d denotes the one link latency
- The first bit of A's frame arrives at B at time t + d
- Suppose an instant before host A's frame arrives, host B begins to transmit its own frame
- B's frame will immediately collide with A's frame and this collision will be detected by host B
- Host B will send the 32-bit jamming sequence
- Host A will not know that the collision occurred until B's frame reaches it, which will happen at t + 2 \* d
- Host A must continue to transmit until this time in order to detect the collision
  - Host A must transmit for 2 \* d to be sure that it detects all possible collisions





Worst-case scenario: (a) A sends a frame at time t; (b) A's frame arrives at B at time t + d; (c) B begins transmitting at time t + d and collides with A's frame; (d) B's runt (32-bit) frame arrives at A at time t + 2d.



- Consider that a maximally configured Ethernet is 2500 m long, and there may be up to four repeaters between any two hosts, the round trip delay has been determined to be 51.2 µs
  - Which on 10 Mbps Ethernet corresponds to 512 bits
- The other way to look at this situation,
  - We need to limit the Ethernet's maximum latency to a fairly small value (51.2 μs) for the access algorithm to work
    - Hence the maximum length for the Ethernet is on the order of 2500 m.

- Once an adaptor has detected a collision, and stopped its transmission, it waits a certain amount of time and tries again.
- Each time the adaptor tries to transmit but fails, it doubles the amount of time it waits before trying again.
- This strategy of doubling the delay interval between each retransmission attempt is known as Exponential Backoff.



- The adaptor first delays either 0 or 51.2 μs, selected at random.
- If this effort fails, it then waits 0, 51.2, 102.4, 153.6 μs (selected randomly) before trying again;
  - This is k \* 51.2 for k = 0, 1, 2, 3
- After the third collision, it waits k \* 51.2 for  $k = 0...2^3 1$  (again selected at random).
- In general, the algorithm randomly selects a k between 0 and  $2^n 1$  and waits for  $k * 51.2 \mu s$ , where n is the number of collisions experienced so far.



### **Experience with Ethernet**

- Ethernets work best under lightly loaded conditions.
  - Under heavy loads, too much of the network's capacity is wasted by collisions.
- Most Ethernets are used in a conservative way.
  - Have fewer than 200 hosts connected to them which is far fewer than the maximum of 1024.
- Most Ethernets are far shorter than 2500m with a roundtrip delay of closer to 5 μs than 51.2 μs.
- Ethernets are easy to administer and maintain.
  - There are no switches that can fail and no routing and configuration tables that have to be kept up-to-date.
  - It is easy to add a new host to the network.
  - It is inexpensive.
    - Cable is cheap, and only other cost is the network adaptor on each host.

#### **Wireless Links**

- Wireless links transmit electromagnetic signals
  - Radio, microwave, infrared
- Wireless links all share the same "wire" (so to speak)
  - The challenge is to share it efficiently without unduly interfering with each other
  - Most of this sharing is accomplished by dividing the "wire" along the dimensions of frequency and space
- Exclusive use of a particular frequency in a particular geographic area may be allocated to an individual entity such as a corporation



#### **Wireless Links**

- These allocations are determined by government agencies such as FCC (Federal Communications Commission) in USA
- Specific bands (frequency) ranges are allocated to certain uses.
  - Some bands are reserved for government use
  - Other bands are reserved for uses such as AM radio, FM radio, televisions, satellite communications, and cell phones
  - Specific frequencies within these bands are then allocated to individual organizations for use within certain geographical areas.
  - Finally, there are several frequency bands set aside for "license exempt" usage
    - Bands in which a license is not needed



#### **Wireless Links**

- Devices that use license-exempt frequencies are still subject to certain restrictions
  - The first is a limit on transmission power
  - This limits the range of signal, making it less likely to interfere with another signal
    - For example, a cordless phone might have a range of about 100 feet.

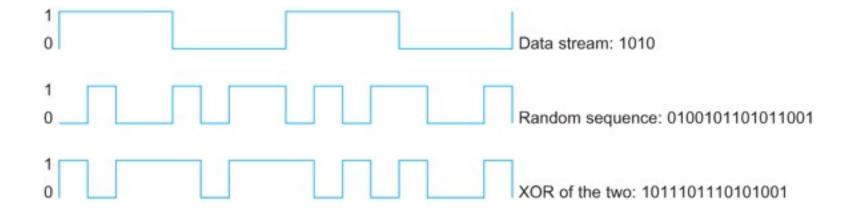


- The second restriction requires the use of Spread Spectrum technique
  - Idea is to spread the signal over a wider frequency band
    - So as to minimize the impact of interference from other devices
    - Originally designed for military use
  - Frequency hopping
    - Transmitting signal over a random sequence of frequencies
      - First transmitting at one frequency, then a second, then a third...
      - The sequence of frequencies is not truly random, instead computed algorithmically by a pseudorandom number generator
      - The receiver uses the same algorithm as the sender, initializes it with the same seed, and is
        - Able to hop frequencies in sync with the transmitter to correctly receive the frame



- A second spread spectrum technique called *Direct* sequence
  - Represents each bit in the frame by multiple bits in the transmitted signal.
  - For each bit the sender wants to transmit
    - It actually sends the exclusive OR of that bit and n random bits
  - The sequence of random bits is generated by a pseudorandom number generator known to both the sender and the receiver.
  - The transmitted values, known as an *n*-bit chipping code,
     spread the signal across a frequency band that is *n* times wider





Example 4-bit chipping sequence



- Wireless technologies differ in a variety of dimensions
  - How much bandwidth they provide
  - How far apart the communication nodes can be
- Four prominent wireless technologies
  - Bluetooth
  - Wi-Fi (more formally known as 802.11)
  - WiMAX (802.16)
  - 3G cellular wireless

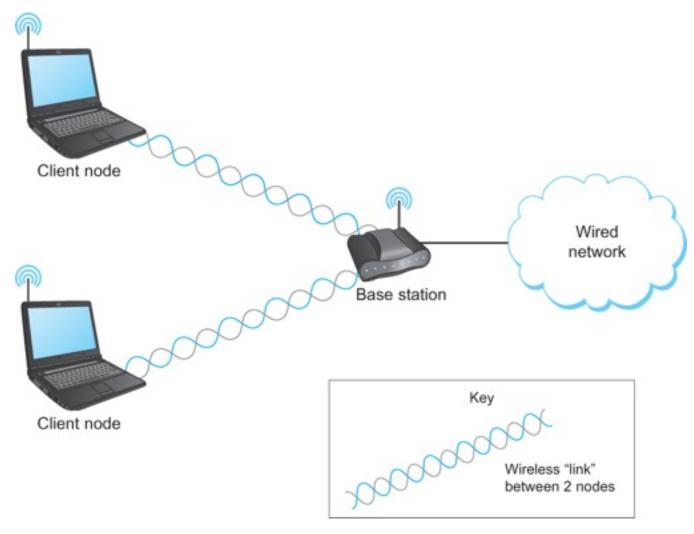


	Bluetooth (802.15.1)	Wi-Fi (802.11)	3G Cellular
Typical link length	10 m	100 m	Tens of kilometers
Typical data rate	2 Mbps (shared)	54 Mbps (shared)	Hundreds of kbps (per connection)
Typical use	Link a peripheral to a computer	Link a computer to a wired base	Link a mobile phone to a wired tower
Wired technology analogy	USB	Ethernet	DSL

Overview of leading wireless technologies

- Mostly widely used wireless links today are usually asymmetric
  - Two end-points are usually different kinds of nodes
    - One end-point usually has no mobility, but has wired connection to the Internet (known as base station)
    - The node at the other end of the link is often mobile





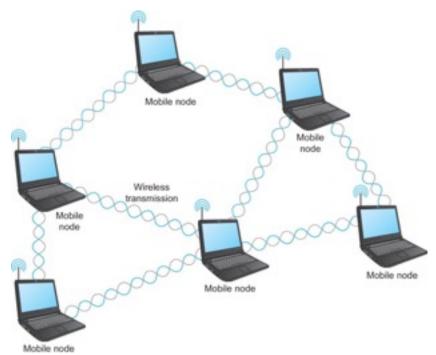
A wireless network using a base station



- Wireless communication supports point-to-multipoint communication
- Communication between non-base (client) nodes is routed via the base station
- Three levels of mobility for clients
  - No mobility: the receiver must be in a fix location to receive a directional transmission from the base station (initial version of WiMAX)
  - Mobility is within the range of a base (Bluetooth)
  - Mobility between bases (Cell phones and Wi-Fi)



- Mesh or Ad-hoc network
  - Nodes are peers
  - Messages may be forwarded via a chain of peer nodes



A wireless ad-hoc or mesh network



## **IEEE 802.11**

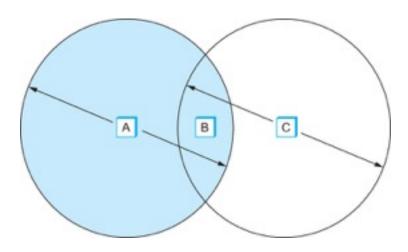
- Also known as Wi-Fi
- Like its Ethernet and token ring siblings, 802.11 is designed for use in a limited geographical area (homes, office buildings, campuses)
  - Primary challenge is to mediate access to a shared communication medium – in this case, signals propagating through space
- 802.11 supports additional features
  - power management and
  - security mechanisms



## **IEEE 802.11**

- Original 802.11 standard defined two radio-based physical layer standard
  - One using the frequency hopping
    - Over 79 1-MHz-wide frequency bandwidths
  - Second using direct sequence
    - Using 11-bit chipping sequence
  - Both standards run in the 2.4-GHz and provide up to 2 Mbps
- Then physical layer standard 802.11b was added
  - Using a variant of direct sequence 802.11b provides up to 11 Mbps
  - Uses license-exempt 2.4-GHz band
- Then came 802.11a which delivers up to 54 Mbps using OFDM
  - 802.11a runs on license-exempt 5-GHz band
- Most recent standard is 802.11g which is backward compatible with 802.11b
  - Uses 2.4 GHz band, OFDM and delivers up to 54 Mbps

- Consider the situation in the following figure where each of four nodes is able to send and receive signals that reach just the nodes to its immediate left and right
  - For example, B can exchange frames with A and C, but it cannot reach D
  - C can reach B and D but not A

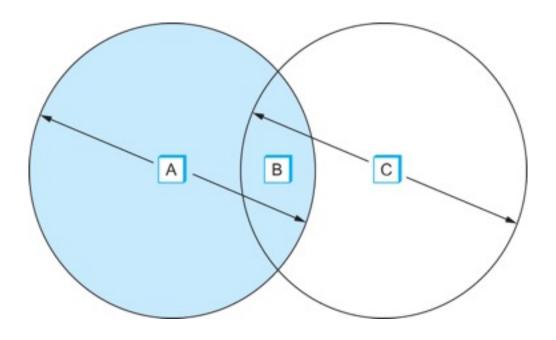


Example of a wireless network



- Suppose both A and C want to communicate with B and so they each send it a frame.
  - A and C are unaware of each other since their signals do not carry that far
  - These two frames collide with each other at B
    - But unlike an Ethernet, neither A nor C is aware of this collision
  - A and C are said to hidden nodes with respect to each other



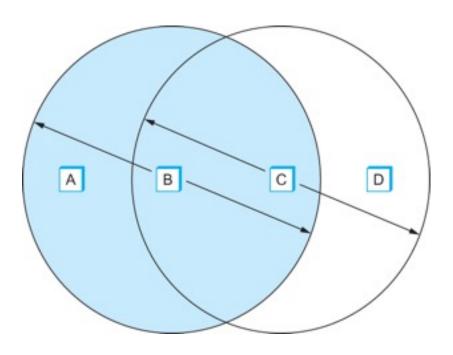


The "Hidden Node" Problem. Although A and C are hidden from each other, their signals can collide at B. (B's reach is not shown.)



- Another problem called exposed node problem occurs
  - Suppose B is sending to A. Node C is aware of this communication because it hears B's transmission.
  - It would be a mistake for C to conclude that it cannot transmit to anyone just because it can hear B's transmission.
  - Suppose C wants to transmit to node D.
  - This is not a problem since C's transmission to D will not interfere with A's ability to receive from B.





Exposed Node Problem. Although B and C are exposed to each other's signals, there is no interference if B transmits to A while C transmits to D. (A and D's reaches are not shown.)



- 802.11 addresses these two problems with an algorithm called Multiple Access with Collision Avoidance (MACA).
- Key Idea
  - Sender and receiver exchange control frames with each other before the sender actually transmits any data.
  - This exchange informs all nearby nodes that a transmission is about to begin
  - Sender transmits a Request to Send (RTS) frame to the receiver.
    - The RTS frame includes a field that indicates how long the sender wants to hold the medium
      - Length of the data frame to be transmitted
  - Receiver replies with a Clear to Send (CTS) frame
    - This frame echoes this length field back to the sender



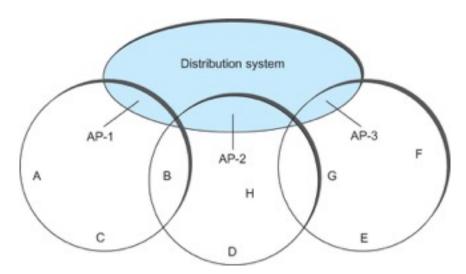
- Any node that sees the CTS frame knows that
  - it is close to the receiver, therefore
  - cannot transmit for the period of time it takes to send a frame of the specified length
- Any node that sees the RTS frame but not the CTS frame
  - is not close enough to the receiver to interfere with it, and
  - so is free to transmit



- Using ACK in MACA
  - Proposed in MACAW: MACA for Wireless LANs
- Receiver sends an ACK to the sender after successfully receiving a frame
- All nodes must wait for this ACK before trying to transmit
- If two or more nodes detect an idle link and try to transmit an RTS frame at the same time
  - Their RTS frame will collide with each other
- 802.11 does not support collision detection
  - So the senders realize the collision has happened when they do not receive the CTS frame after a period of time
  - In this case, they each wait a random amount of time before trying again.
  - The amount of time a given node delays is defined by the same exponential backoff algorithm used on the Ethernet.

- 802.11 is suitable for an ad-hoc configuration of nodes that may or may not be able to communicate with all other nodes.
- Nodes are free to move around
- The set of directly reachable nodes may change over time
- To deal with this mobility and partial connectivity,
  - 802.11 defines additional structures on a set of nodes
  - Instead of all nodes being created equal,
    - some nodes are allowed to roam
    - some are connected to a wired network infrastructure
      - they are called *Access Points* (AP) and they are connected to each other by a so-called *distribution system*

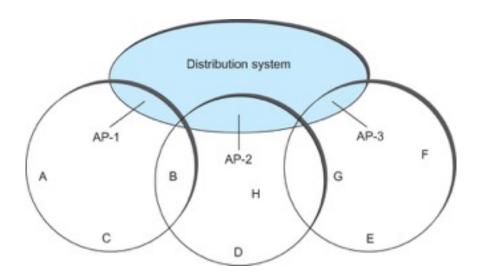
- Following figure illustrates a distribution system that connects three access points, each of which services the nodes in the same region
- Each of these regions is analogous to a cell in a cellular phone system with the APIs playing the same role as a base station
- The distribution network runs at layer 2 of the ISO architecture



Access points connected to a distribution network



- Although two nodes can communicate directly with each other if they are within reach of each other, the idea behind this configuration is
  - Each nodes associates itself with one access point
  - For node A to communicate with node E, A first sends a frame to its AP-1 which forwards the frame across the distribution system to AP-3, which finally transmits the frame to E

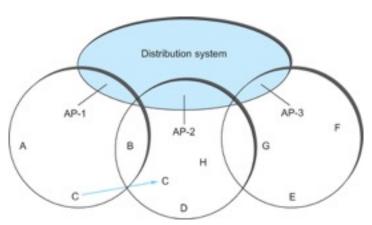


Access points connected to a distribution network



- How do the nodes select their access points
- How does it work when nodes move from one cell to another
- The technique for selecting an AP is called scanning
  - The node sends a *Probe* frame
  - All APs within reach reply with a Probe Response frame
  - The node selects one of the access points and sends that AP an Association Request frame
  - The AP replies with an Association Response frame
- A node engages this protocol whenever
  - it joins the network, as well as
  - when it becomes unhappy with its current AP
    - This might happen, for example, because the signal from its current AP has weakened due to the node moving away from it
    - Whenever a node acquires a new AP, the new AP notifies the old AP of the change via the distribution system

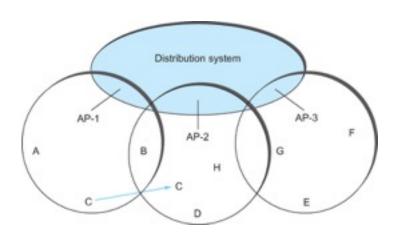
- Consider the situation shown in the following figure when node C moves from the cell serviced by AP-1 to the cell serviced by AP-2.
- As it moves, it sends *Probe* frames, which eventually result in *Probe Responses* from AP-2.
- At some point, C prefers AP-2 over AP-1, and so it associates itself with that access point.
  - This is called active scanning since the node is actively searching for an access point



**Node Mobility** 



- APs also periodically send a *Beacon* frame that advertises the capabilities of the access point; these include the transmission rate supported by the AP
  - This is called passive scanning
  - A node can change to this AP based on the *Beacon* frame simply by sending it an *Association Request* frame back to the access point.



**Node Mobility** 



### **IEEE 802.11 – Frame Format**

- Source and Destinations addresses: each 48 bits
- Data: up to 2312 bytes
- CRC: 32 bit
- Control field: 16 bits
  - Contains three subfields (of interest)
    - 6 bit Type field: indicates whether the frame is an RTS or CTS frame or being used by the scanning algorithm
    - A pair of 1 bit fields : called ToDS and FromDS



Frame Format



### **IEEE 802.11 – Frame Format**

- Frame contains four addresses
- How these addresses are interpreted depends on the settings of the
   ToDS and FromDS bits in the frame's Control field
- This is to account for the possibility that the frame had to be forwarded across the distribution system which would mean that,
  - the original sender is not necessarily the same as the most recent transmitting node
- Same is true for the destination address
- Simplest case
  - When one node is sending directly to another, both the DS bits are 0,
     Addr1 identifies the target node, and Addr2 identifies the source node



### **IEEE 802.11 – Frame Format**

- Most complex case
  - Both DS bits are set to 1
    - Indicates that the message went from a wireless node onto the distribution system, and then from the distribution system to another wireless node
  - With both bits set,
    - Addr1 identifies the ultimate destination,
    - Addr2 identifies the immediate sender (the one that forwarded the frame from the distribution system to the ultimate destination)
    - Addr3 identifies the intermediate destination (the one that accepted the frame from a wireless node and forwarded across the distribution system)
    - Addr4 identifies the original source
- Addr1: E, Addr2: AP-3, Addr3: AP-1, Addr4: A



## **Bluetooth**

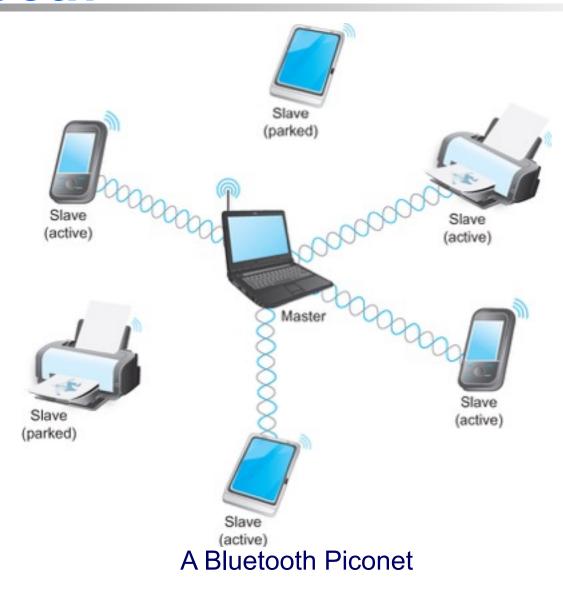
- Used for very short range communication between mobile phones, PDAs, notebook computers and other personal or peripheral devices
- Operates in the license-exempt band at 2.45 GHz
- Has a range of only 10 m
- Communication devices typically belong to one individual or group
  - Sometimes categorized as Personal Area Network (PAN)
- Version 2.0 provides speeds up to 2.1 Mbps
- Power consumption is low



## **Bluetooth**

- Bluetooth is specified by an industry consortium called the Bluetooth Special Interest Group
- It specifies an entire suite of protocols, going beyond the link layer to define application protocols, which it calls profiles, for a range of applications
  - There is a profile for synchronizing a PDA with personal computer
  - Another profile gives a mobile computer access to a wired LAN
- The basic Bluetooth network configuration is called a piconet
  - Consists of a master device and up to seven slave devices
  - Any communication is between the master and a slave
  - The slaves do not communicate directly with each other
  - A slave can be parked: set to an inactive, low-power state

# **Bluetooth**





# **ZigBee**

- ZigBee is a new technology that competes with Bluetooth
- Devised by the ZigBee alliance and standardized as IEEE 802.15.4
- It is designed for situations where the bandwidth requirements are low and power consumption must be very low to give very long battery life
- It is also intended to be simpler and cheaper than Bluetooth, making it financially feasible to incorporate in cheaper devices such as a wall switch that wirelessly communicates with a ceiling-mounted fan



# **Summary**

- We introduced the many and varied type of links that are used to connect users to existing networks, and to construct large networks from scratch.
- We looked at the five key issues that must be addressed so that two or more nodes connected by some medium can exchange messages with each other
  - Encoding
  - Framing
  - Error Detecting
  - Reliability
  - Multiple Access Links
    - Ethernet
    - Wireless 802.11, Bluetooth

