

# CS 352: Internet Technology- Midterm 07/19

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## Exam Content

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The exam has been partitioned into 6 sections totaling 114 pts:

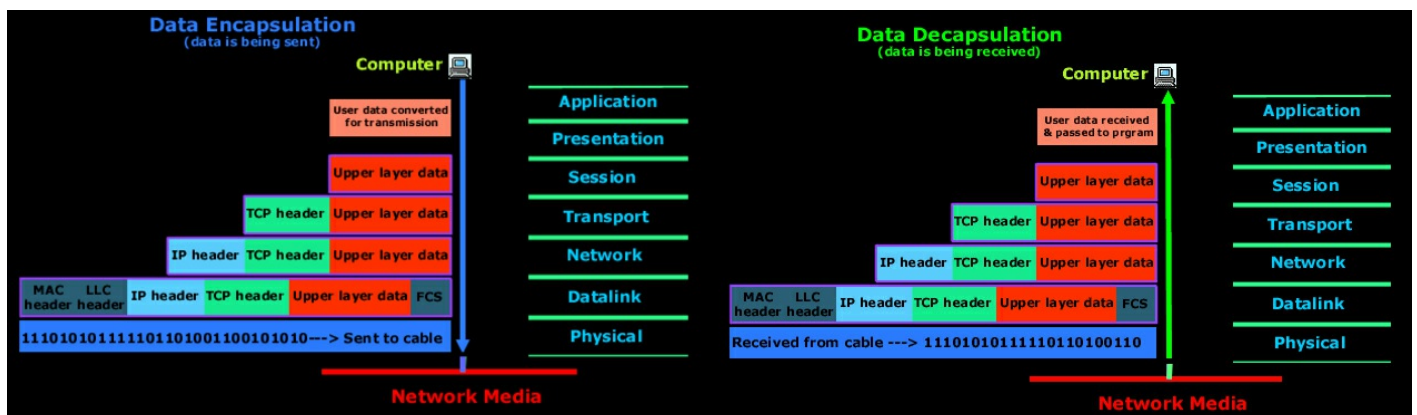
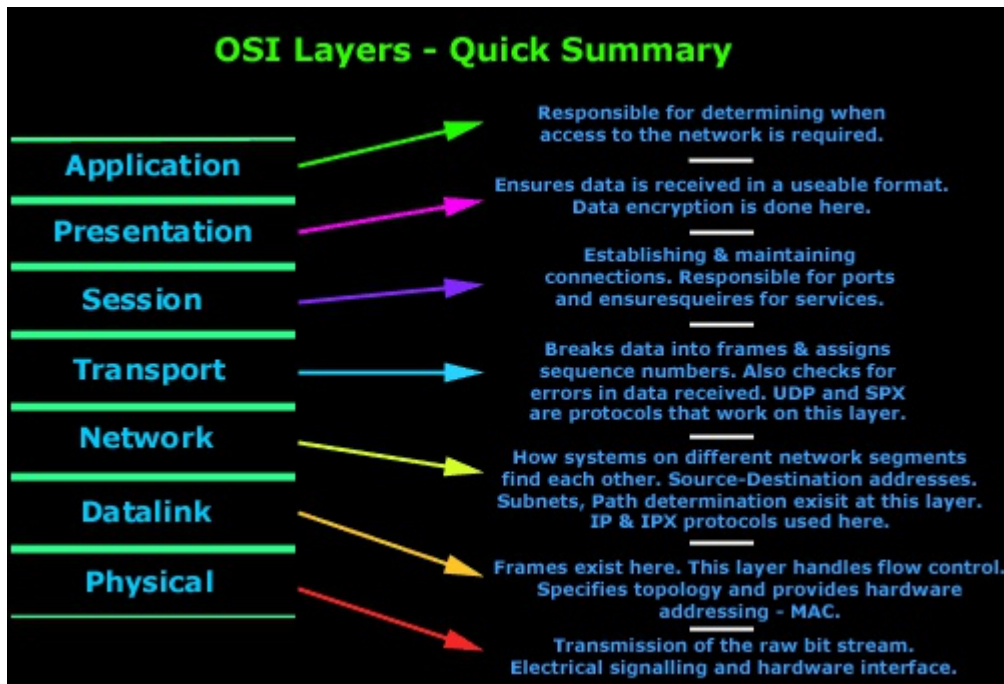
1. **Networking** ~ 24 pts
2. **TCP Details** ~ 20 pts
3. **Delay and Transport** ~ 20 pts
4. **Transport Calc** ~ 20 pts
5. **HTTP Protocol** ~ 30 pts
6. **Extra** ~ 3 pts

## Terms

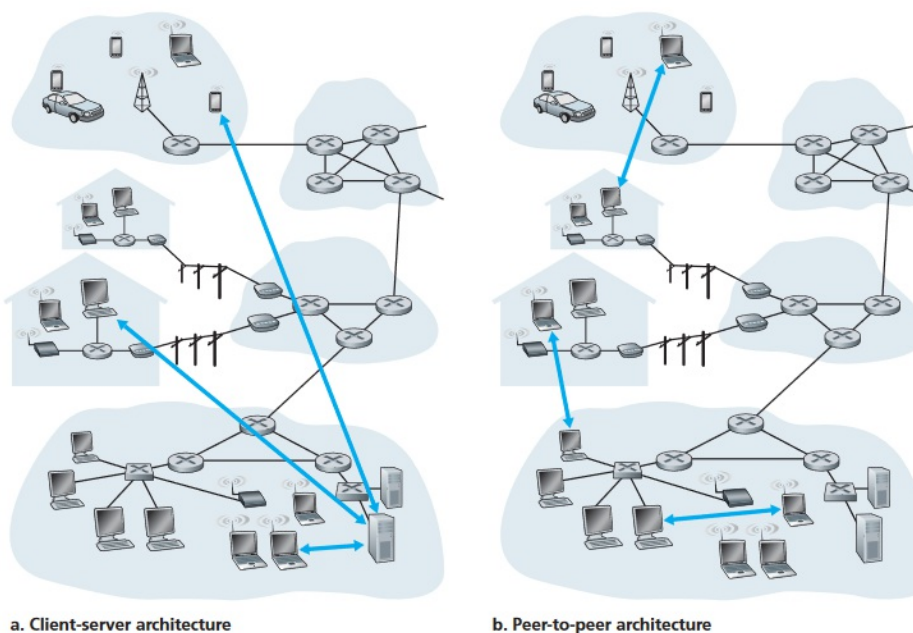
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- **Network**: Collection of interconnected *machines*
- **Internetwork**: Collection of interconnected *networks*
- **Host**: Machine running user application
- **Channel**: *Logical* line of communication
- **Medium**: Physical process used (e.g. fiber optics, copper wire, satellite link)
- **Network Core**: The mesh of packet switches and links that interconnect the Internet's end systems
- **EDGE Network**: A network which provides for the main or CORE network; companies and organizations with a default route
- **CORE Network**: ISP tiers with the biggest ISPs in tier 1 and smaller, regional ISPs in tiers 2, 3
- **Protocol**: Defines the format and the order of messages exchanged between two or more communicating entities, as well as the actions taken on the transmission and/or receipt of a message or other event
- **Packets**: Smaller chunks of data broken off from longer "messages"
- **Packet Switching**: Group data transmitted over the network into packets which then get transmitted (e.g. routers, link-layer switches)
- **Store-And-Forward**: Packet switch must receive the entire packet to transmit the first bit of packet onto outbound link
- **Output Buffer**: An *output queue* that stores the packets that the router is about to send into a link
- **Queueing Delays**: Variable delays that depend on the level of congestion in the network and occurs when an incoming packet finds the link busy with transmission of another packet
- **Processing Delay**: The time required to examine the packet's header and determine where to direct the packet
- **Transmission Delay**: Amount of time required to push/transmit all of the packet's bits into the link
- **Propagation Delay**: The time required to propagate from the beginning to router B
- **Packet Loss**: The arriving packet or an already-queued packet may be dropped if an arriving packet finds that the buffer is completely full with other packets waiting for transmission
- **Forwarding Table**: Each router has one of these that maps destination address (or portions of it) to that router's outbound links
- **Circuit Switching**: A circuit-based network that guarantees a constant transmission rate between a sender and receiver. For these purposes, a dedicated *end-to-end* connection exists
- **Frequency-Division Multiplexing**: The frequency spectrum of a link is divided up among the connections across a link; the link dedicates a frequency band to each connection for the duration of the connection.
- **Time-Division Multiplexing**: Time is divided into frames of fixed duration, and each frame is divided into a fixed number of time slots. When network establishes a connection across a link, the network dedicates one time slot for the sole use of that connection in every frame to this connection.
- **Statistical Multiplexing**:
- **Encapsulation**: The process of passing information through the OSI stack, usually by appending header and payload.
- **Client-Server Architecture**: An always-on *host* accepts service requests from incoming *clients*, but the clients do not directly communicate with each other
- **P2P Architecture**: The application exploits direct communication between intermittently connected hosts, or *peers*, without passing through a dedicated server
- **Socket**: The interface between your application and the network; one of the endpoints in a two-way communication link between two programs running on the network
- **Round-Trip Time (RTT)**: Amount of time for a small packet to travel to the server and back;  $\text{Response Time} = 2\text{RTT} + \text{FTT}$
- **Maximum Segment Size (MSS)**: The maximum amount of data that can be grabbed and placed in a segment

# ISO-OSI Seven-Layer Stack



## Client-Server vs. P2P



# HTTP 1.0

## Response Codes

| Status Code | Response                   | Meaning      |
|-------------|----------------------------|--------------|
| 200         | OK                         | Success      |
| 304         | Not Modified               | Redirection  |
| 400         | Bad Request                | Client error |
| 403         | Forbidden                  | Client error |
| 404         | Not Found                  | Client error |
| 408         | Request Timeout            | Client error |
| 418         | I'm a Teapot               | For Fun      |
| 500         | Internal Server Error      | Server error |
| 501         | Not Implemented            | Server error |
| 503         | Service Unavailable        | Server error |
| 505         | HTTP Version Not Supported | Server error |

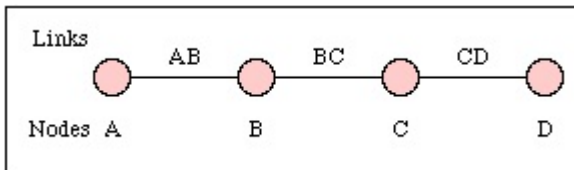
## MIME Types

HTTP 1.0 also supports multi-parts and messages

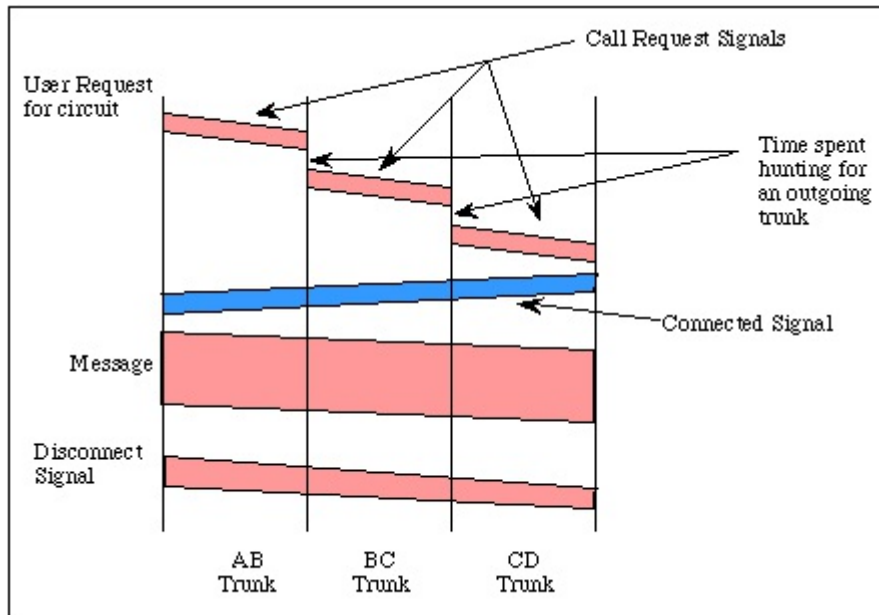
| Extension   | Kind of Document                 | MIME Type                |
|-------------|----------------------------------|--------------------------|
| .txt        | default text file                | text/plain               |
| .htm, .html | HTML                             | text/html                |
| .*          | default for all other file types | application/octet-stream |
| .gif        | GIF                              | image/gif                |
| .jpg        | JPEG Images                      | image/jpeg               |
| .*          | Basic Audio Files                | audio/basic              |
| .mpeg       | MPEG Video                       | video/mpeg               |

## Delay and Transport (with Calculations)

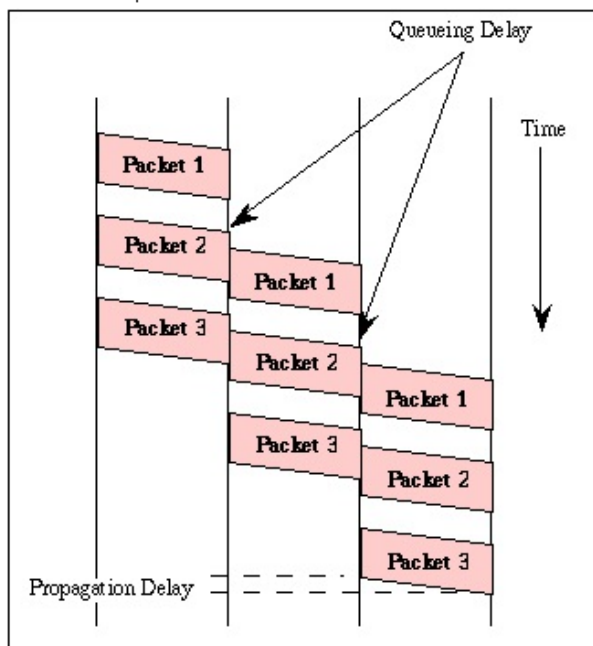
- Processing, Queueing, Transmission, and Propagation Delays are as defined in the definitions section
- Network:  $d_{\text{nodal}} = d_{\text{proc}} + d_{\text{queue}} + d_{\text{trans}} + d_{\text{prop}}$
- Transport



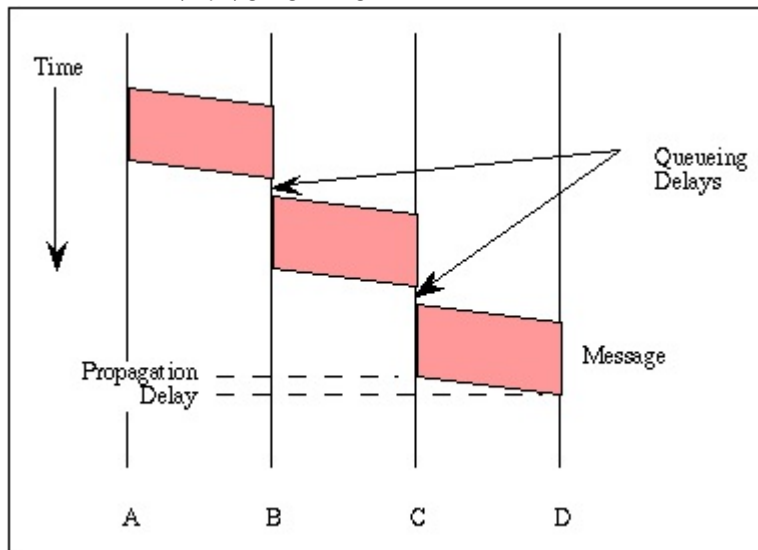
- Circuit Switching:  $t = C + M/B$ 
  - C: connection setup time (secs)
  - M: message size (bits)
  - B: bandwidth (bits/sec)



- Packet Switching:  $t = t_0 * (P + S)$ 
  - h: header size
  - d: packet data size (payload)
  - $t_0$ : time it takes to send one packet =  $(h+d)/B$
  - B: bandwidth (bits/sec)
  - S: no. of stations (hops) going through
  - P: number of packets to send =  $\lceil M/d \rceil$



- Message Switching:  $t = t_0 * (1 + S)$ 
  - h: header size
  - $t_0$ : time it takes to send one message =  $(h+M)/B$
  - M: message size (bits)
  - B: bandwidth (bits/sec)
  - S: no. of stations (hops) going through



## Useful Comparisons

### TCP vs. UDP

| PRO/CON                     | TCP | UDP |
|-----------------------------|-----|-----|
| connectionless              | -   | X   |
| faster                      | -   | X   |
| more reliable               | X   | -   |
| provides ports              | -   | X   |
| more functionality          | X   | -   |
| point-to-point              | X   | -   |
| flow/congestion control     | X   | -   |
| light weight/less header    | -   | X   |
| in-order packet delivery    | X   | -   |
| higher thruput <sup>+</sup> | -   | X   |

<sup>+</sup> UDP packets are easier to process, esp. the source

## HTTP vs. FTP vs. SMTP

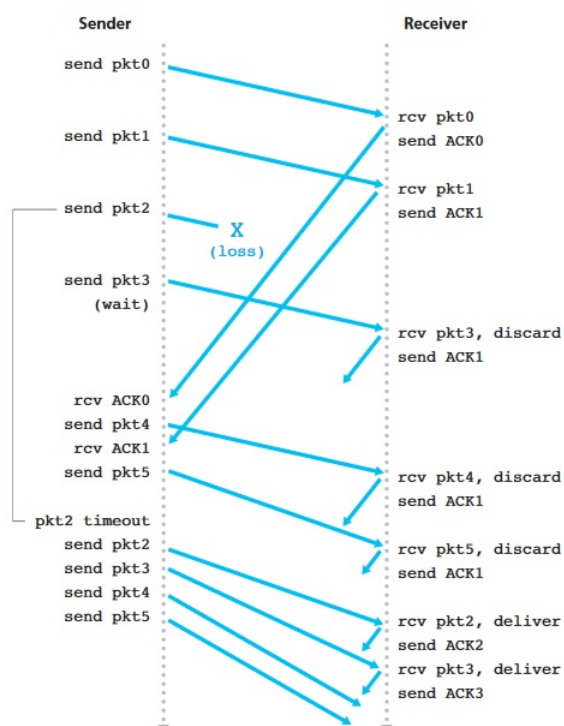
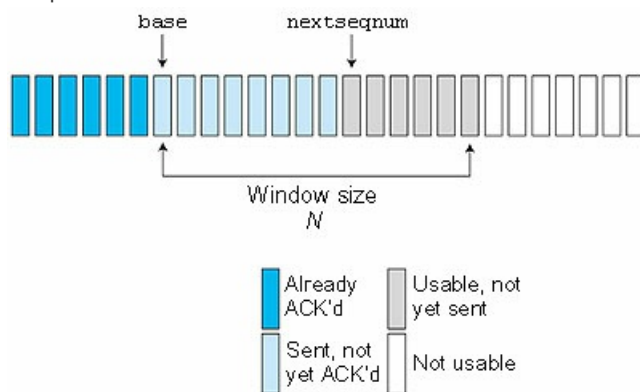
| FEATURE                   | HTTP | FTP | SMTP |
|---------------------------|------|-----|------|
| request/response protocol | X    | X   | X    |
| connection semantics      | -    | X   | X    |

## Circuit vs. Packet vs. Message Switching

- **Header Overhead:** Circuit (best) < Packet < Message (worst)
- **Transmission Delay:**
  - **Short, Bursty Messages:** Packet (best) < Message < Circuit (worst)
  - **Long, Continuous Messages:** Circuit (best) < Message < Packet (worst)
- Circuit vs. Packet
  - Still (TDM, FDM) vs. Statistical Multiplexing
  - Less vs. More Users

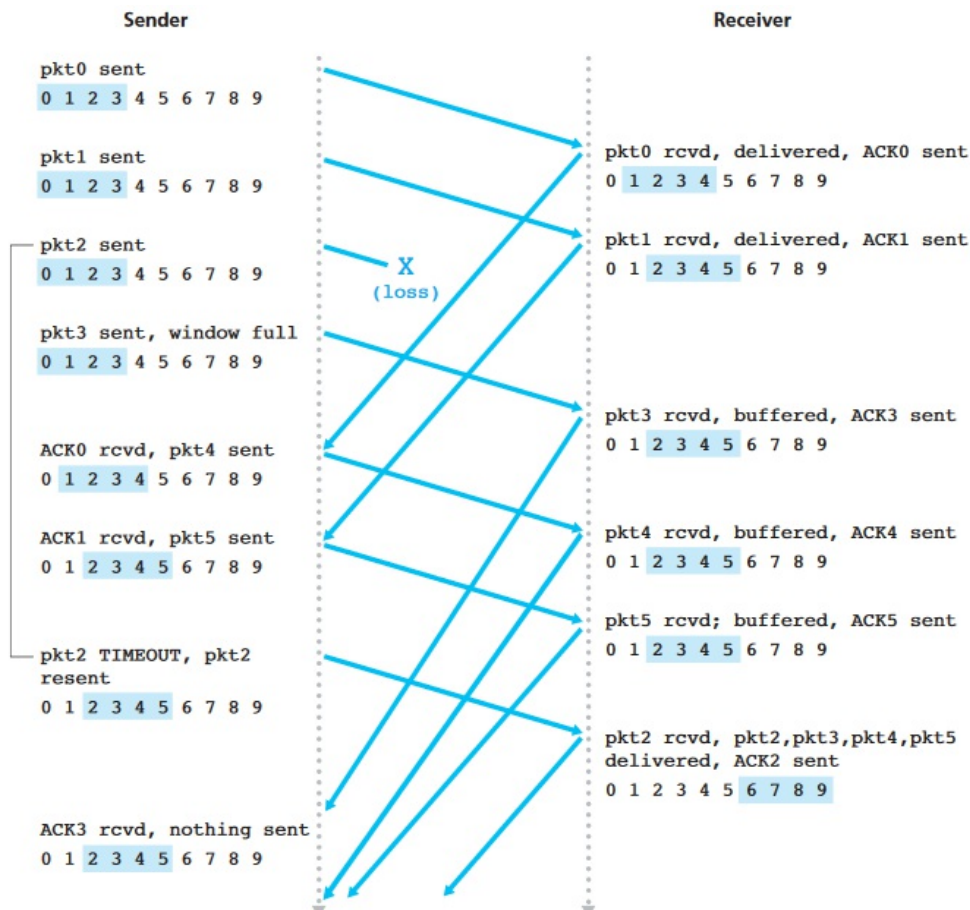
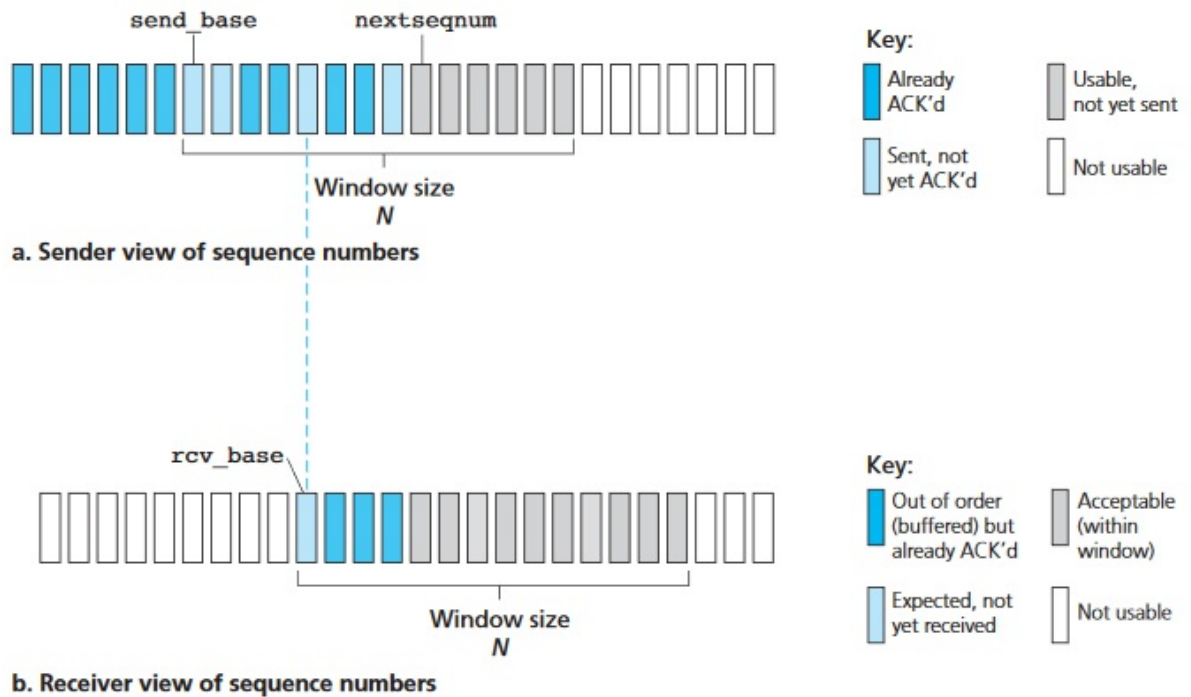
## Go-Back-N vs. Selective Repeat Protocols

- The *Go-Back-N (GBN)*
  - Allows the sender to transmit multiple available packets without waiting for an acknowledgement but is constrained to have a maximum of  $N$  unacknowledged packets in the pipeline
  - $N$  represents the window size



- The *Selective Repeat (SR)*

- Avoids unnecessary retransmissions by having the sender retransmit only those that it suspects were lost or corrupted at the receiver
- A window size of  $N$  will again be used to limit unacknowledged packets in the pipeline
- However, unlike GBN, the sender will have already received ACKs for some of the packets in the window
- The receiver will acknowledge a correctly received packet regardless of whether it is in order or not
- The receiver *re-acknowledges* already-received packets with certain sequence *below* the window number
- No synchronization between sender and receiver



## Flow Control vs. Congestion Control

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- Flow Control
  - How much the user can handle
  - Makes sure sender does not overload receiver by sending only enough data to accommodate receiver
  - Allows for more efficient bi-directional communication
  - Initiated by sender
- Congestion Control
  - How much the network can handle
  - Makes sure that the network can handle packet load
  - Reduces amount of sent packets to avoid overflowing its buffer/queue
  - Initiated by router

## Exponential Math

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Tip:  $2^5 = 32$  &  $2^{10} = 1024$

| Unit | Value    |
|------|----------|
| KB   | $2^{10}$ |
| MB   | $2^{20}$ |
| GB   | $2^{30}$ |
| TB   | $2^{40}$ |
| PB   | $2^{50}$ |