



### **Introduction to Network**

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https://www.sjtu.edu.cn

#### **Review: scalable websites**

Caching server Caching server Caching server Scalable websites powered by distributed Caching Caching systems For request handling, data storage Distributed caching Database server Database server **Database Database Users Application #1** user, price user, price generate the page Application server Distributed database Internet Load **Application #2** File server File server File server Balance add the order File: File: 0.00 image Application server Message queue Distributed file system

### **Layers in Network**

#### **Application**

- Can be thought of as a fourth layer
- Not part of the network

#### **End-to-end layer**

Everything else required to provide a comfortable application interface

#### **Network layer**

Forwarding data through intermediate points to the place it is wanted

#### Link layer

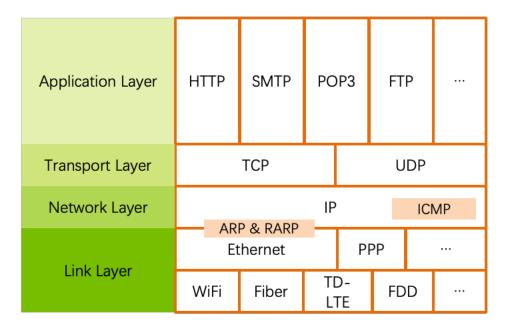
Moving data directly from one point to another

### **OSI, TCP/IP & Protocol Stack**

#### OSI

- The open systems interconnection (OSI) model
- 7-layer architecture

7th Application Layer
6th Presentation Layer
5th Session Layer
4th Transport Layer
3th Network Layer
2nd Link Layer
1st Physical Layer



End-to-end Layer

Network Layer

Link Layer

#### The Internet "Hour Glass" Protocols

#### More people, more useful

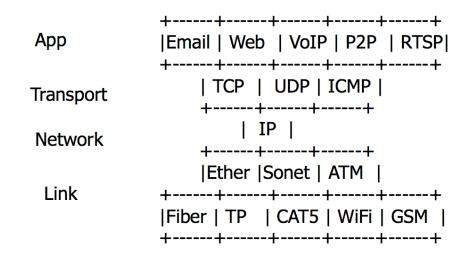
- Value to me = N
- Value to society is N<sup>2</sup>

#### Network, dumb vs. smart

Standardize vs. flexibility

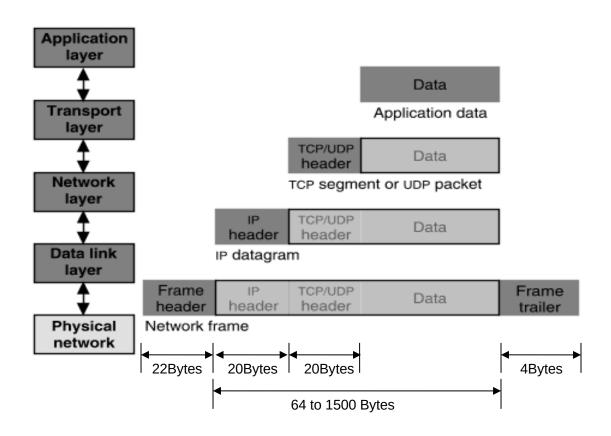
#### Network is a black box

Simplify the system that uses it



"Everything over IP, and IP over everything"

### **Packet Encapsulation**



### **Application Layer**

#### **Entities**

- Client and server
- End-to-end connection

Name space: URL

#### **Protocols**

HTTP, FTP, POP3, SMTP, etc.

#### What to care?

Content of the data: video, text, ...



```
<html>
    <head>
        <title>Google</title>

<script>window.google=.....
        </script>
        </head>
        <body> ... </body>
        </html>
```

### **Transport Layer**

#### **Entities**

- Sender and receiver
- Proxy, firewall, etc.
- End-to-end connection

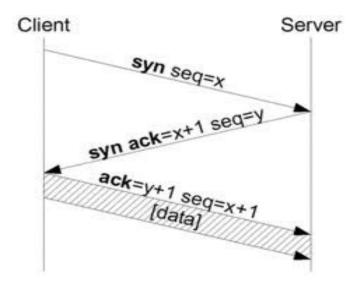
Name space: port number

Protocols: TCP, UDP, etc.

#### What to care?

- TCP: Retransmit packet if lost
- UDP: nothing





### **Packet Format of TCP & UDP**

			TCP Segme	ent	Header	Forma	ıt	
Bit#	0	7	8	15	16	23	24	31
0	Source Port			Destination Port				
32	Sequence Number							
64	Acknowledgment Number							
96	Data Offset	Res	Flags			Windo	w Size	
128	Header and Data Checksum			Urgent Pointer				
160	Options							

UDP Datagram Header Format								
Bit #	0	7	8	15	16	23	24	31
0	Source Port			Destination Port				
32	Length			Header and Data Checksum				

### **Network Layer (the Internet Layer, IP Layer)**

#### **Network entities**

- Gateway, bridge
- Router, etc.

#### Name space

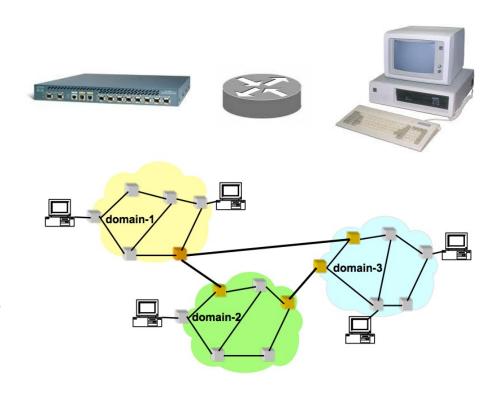
IP address

#### **Protocols**

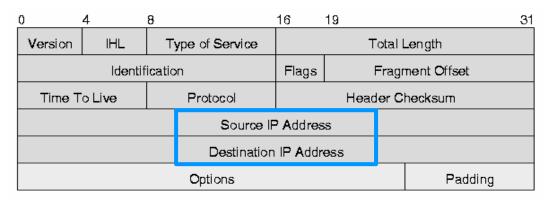
IP, ICMP (ping)

#### What to care?

Next hop decided by route table

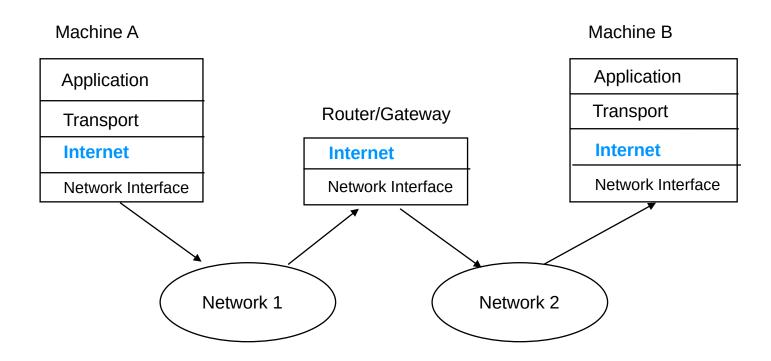


### IP Datagram (Packet, Package)



Header

#### **TCP/IP Architecture**



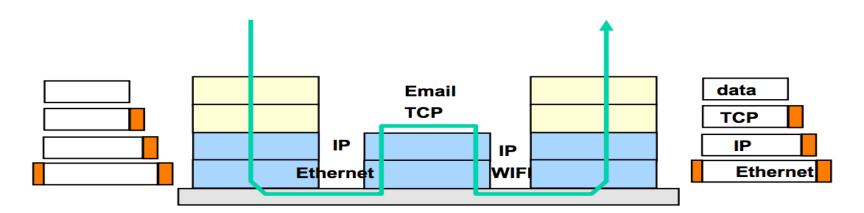
#### **TCP/IP Architecture**

Each layer adds/strips off its own header

Each layer may split up higher-level data

Each layer multiplexes multiple higher layers

Each layer is (mostly) transparent to higher layers



## **Link Layer**

From a node to its physical neighbor

### The Link Layer

#### The bottom-most layer of the three layers

#### Purpose: moving data directly from one physical location to another

- 1. Physical transmission
- 2. Multiplexing the link
- 3. Framing bits & bit sequences
- 4. Detecting transmission errors
- 5. Providing a useful interface to the up layer

### **Physical Transmission using Shared Clock**

# Example-1: moving a bit from register-1 to register-2 on the same chip

- Run a wire to connect output of reg-1 to input of reg-2
- Wait till reg-1's output has settled & signal has propagated to reg-2
- Reg-2 read input the next clock tick
- Assumption: propagation can be done within one clock

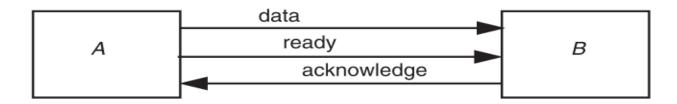
How to send data between two modules without sharing a clock?



### **Physical Transmission without Shared Clock**

#### Three-wire ready/acknowledge protocol

- 1. A places data on data line
- 2. A changes value on the ready line
- B sees the ready line change, reads value on the data line, then changes the acknowledge line



B: when to look at the data line? (ready is set)

A: when to stop holding the bit value on the data line? (ack is set)

#### **Parallel Transmission**

#### **Propagation time Δt**

- It takes more than  $2x\Delta t$  to send one bit
- The max data rate is  $1/(2\Delta t)$

#### **Parallel transmission**

- Use N parallel data lines to achieve  $N/(2\Delta t)$
- E.g., SCSI, printer, etc.

#### **Serial Transmission**

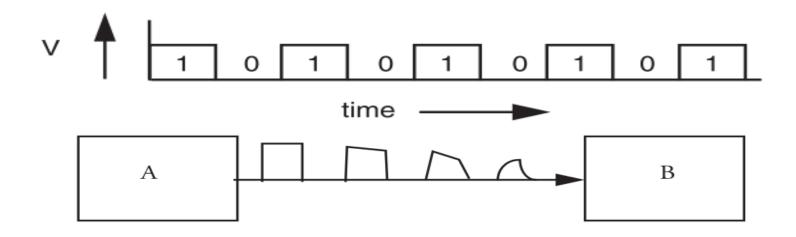
#### Ready/acknowledge protocol

Δt grows significantly, which limits the data rate

#### Serial transmission

- Send a stream of bits down a single line
- Without waiting for any response from the receiver
- Expect the receiver can recover the bits with no additional signal
- Higher rates, longer distance, fewer wires
- E.g., USB, SATA

### **Signal Transmission on Analog Line**



#### It is hard for B to understand the signal

– B doesn't have a copy of A's clock, so when to sample the signal?

### **VCO: Voltage Controlled Oscillator**

How to make two ends agree on the data rate without clock line?

#### The receiver run a VCO at about the same data rate

- VCO's output is multiplied by the voltage of incoming signal
- The product is suitably filtered and sent back to adjust the VCO
- VCO will finally be locked to both the frequency and phase of the arriving signal: phase-locked loop
- Then the VCO becomes a clock source for the receiver.

Problem: if no transition in the stream (e.g., a lot of zero), the phase-locked loop cannot synchronize

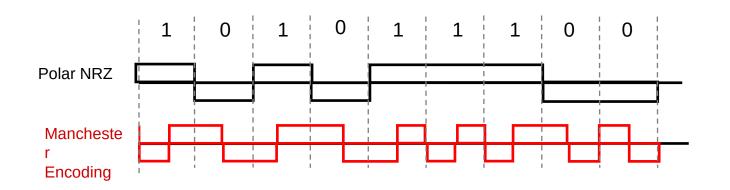


#### **Manchester Code**

Solution: sender encodes the data to ensure transitions

Phase encoding: at least 1 level transition for a bit

- Manchester code: 0 -> 01, 1 -> 10
- Max data rate is only half, but simple enough



#### **How to Share a Connection?**

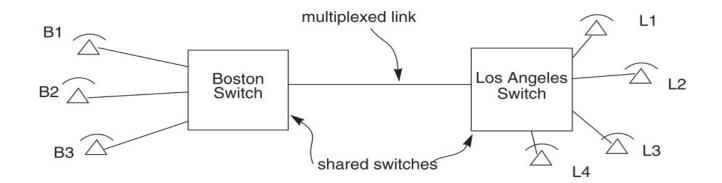
#### **Isochronous communication (telephone communication)**

- Needs prior arrangement between switches
- Connection: set up and tear down
- Stream: continuous bits flows out of a phone

#### **Asynchronous communication (data communication)**

- Message: burst, ill-suited to fixed size and spacing of isochronous frames
- Connectionless, asynchronous

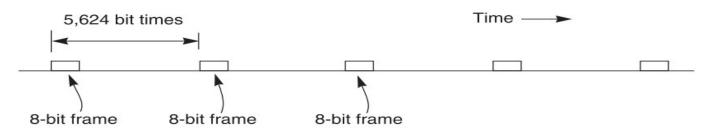
### **Isochronous Multiplexing**



#### **Telephone network**

- Leverage "virtual link" for connection
- "network is busy" when no available time slot

#### **Isochronous - TDM**



64 Kbps each phone, 45 Mbps link

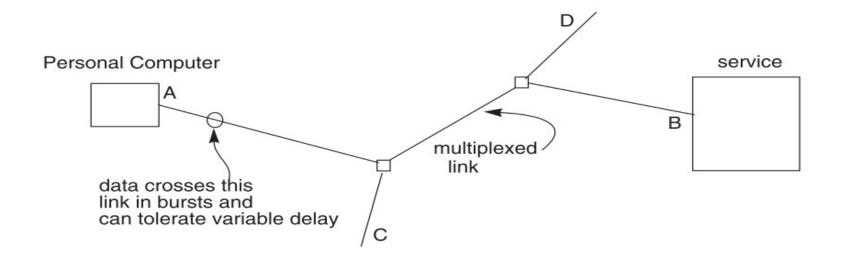
8-bit block (frame), 8000 frames per second

5624 bit times or 125 us

703 simultaneous conversations (what if there is a 704th calling?)

Q: Why the voice is still *continuous*, instead of *fragmented*?

### **Data Communication Network**



Data communication network usually contains burst communication

Different from the telephone network

### Frame and Packet: Asynchronous Link

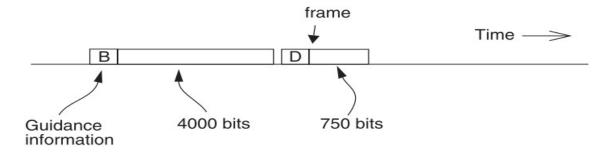
Frame can be of any length, carried at any time that the link is free

Packet: a variable-length frame with its guidance info

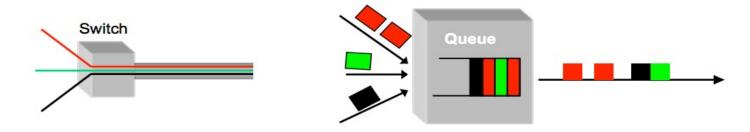
Connectionless transmission: no state maintained

Segment and reassemble

Packet voice: replacing many parts of isochronous network



### **Multiplexing / Demultiplexing**



Multiplex using a queue: switch need memory/buffer

#### Demultiplex using information in packet header

- Header has destination
- Switch has a forwarding table that contains information about which link to use to reach a destination

### **Framing Frames**

#### Where a frame begins and ends

#### **Independent from framing bits**

Some model separates link layer to 2: one for bits and one for frames

#### Simple method

- Choose a pattern of bits, e.g., 7 one-bits in a row, as a frame-separator
- Bit stuffing: if data contains 6 ones in a row, then add an extra bit (0)

### **Error Handling**

#### **Error detection code**

Adding redundancy: e.g., checksum at the end

#### What to do if detect an error

- Error correction code: with enough redundancy
  - Where noise is well understood, e.g., disk
- Ask sender to resend: sender holds frame in buffer
- Let receiver discard the frame
- Blending these techniques

### **Coding: Incremental Redundancy**

#### **Forward error correction**

- Perform coding before storing or transmitting
- Later decode the data without appealing to the creator

#### **Hamming distance**

- Number of 1 in  $A \oplus B$ ,  $\oplus$  is exclusive OR (XOR)
- If H-distance between every legitimate pair is 2
  - 000101, can only detect 1-bit flip
- If H-distance between every legitimate pair is 3
  - Can only correct 1 bit flip
- If H-distance between every legitimate pair is 4
  - Can detect 2-bit flip, correct 1-bit flip

100101 000111

100101 010111

### **Example-1: Simple Parity Check**

#### 2 bits -> 3 bits

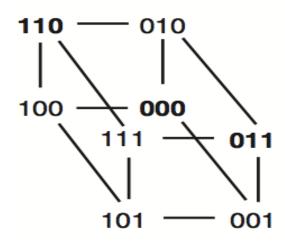
- Detect 1-bit errors
- 8 patterns total

#### Only 4 correct patterns

- -00 -> 000
- 11 -> 11**0**
- -10 -> 101
- -01 -> 011

#### Hamming distance of this code is 2

1-bit flipping will cause incorrect pattern



### **Example-2: 4-bit -> 7-bit**

#### 4 bits -> 7 bits (56 using only extra 7)

- 3 extra bits to distinguish 8 cases
- e.g. 1101 -> 1010101

#### **Correct 1-bit errors**

- 1010101 -> 1010001 : P1 & P4 not match
- 1010101 -> 1110101 : P2 not match

$P_1$	=	$P_7$	$\oplus$	$P_5$	$\oplus$	$P_3$
$P_2$						
$P_4$	=	$P_7$	$\oplus$	$P_6$	$\oplus$	$P_5$

Not Match	Error
None	None
P1	P1
P2	P2
P4	P4
P1 & P2	P3
P1 & P4	P5
P2 & P4	P6
P1 & P2 & P4	P7

# **NETWORK LAYER**

#### **IP: Best-effort Network**

#### 1. Best-effort network

If it cannot dispatch, may discard a packet

#### 2. Guaranteed-delivery network

- Also called store-and-forward network, no discarding data
- Work with complete messages rather than packets
- Use disk for buffering to handle peaks
- Tracks individual message to make sure none are lost

#### In real world

- No absolute guarantee
- Guaranteed-delivery: higher layer; best-effort: lower layer

### **Duplicate Packets and Suppression**

### Discarding packets is common case

Many network protocol includes timeout and resend mechanism

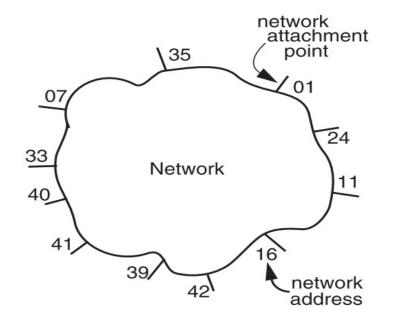
### When a congested forwarder discards a packet

- Client does not receive a response as quickly as originally hoped
- Users may prepared for duplicate requests and responses
- Detecting duplicates may or may not be important

### The Network Layer

#### **Addressing interface**

- Network attachment points
- Network address
- Source & destination



**NETWORK\_SEND** (segment\_buffer, destnation, network\_protocol, end\_layer\_protocol)

**NETWORK\_HANDLE** (packet, network\_protocol)

### Managing the Forwarding Table: Routing

#### **Routing (or path-finding)**

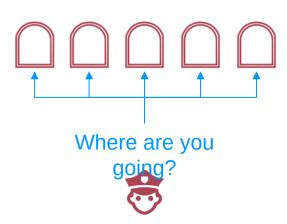
Constructing the tables

#### Impractical by hand

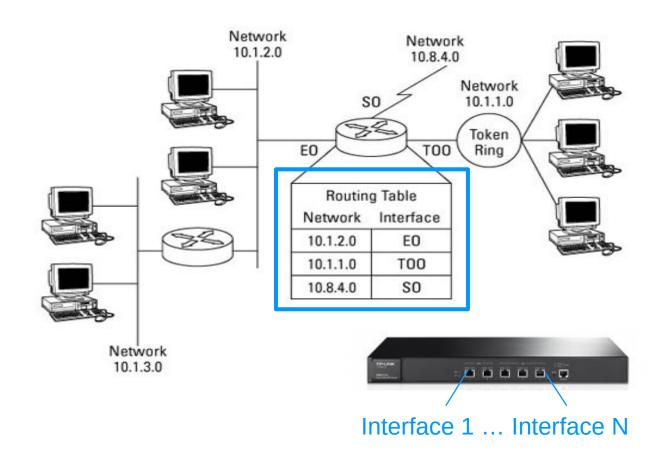
- Determining the best paths requires calculation
- Recalculating the table when links change
- Recalculating the table when link fails
- Adapt according to traffic congestion

#### Static routing vs. adaptive routing

Adaptive routing requires exchange of info



#### **IP Route Table**



### Control-plane VS. Data-plane

#### Control-plane

- Control the data flow by defining rules
- E.g., the routing algorithm

#### **Data-plane**

- Copies data according to the rules
- Performance critical
- E.g., the IP forwarding process