

Computer Networks

LAB 1

Some common commands used in Linux and Windows:

Linux Command	Windows Command	Usage / Effect
ifconfig	ipconfig	to find ip address of the computer
hostname	hostname	to display host name
nmap	nmap	To scan what hosts are available on a network and what ports they have open.
nslookup	nslookup	to list variety of info about DNS and the computers that have joined the domain
ping	ping	to check if a host can be accessed (by IP or name)
traceroute	tracert	to trace route from a host through internet router to a destination. Useful to discover why a network cannot get access to internet, and internet routing problems.
netstat	netstat	to print status of network ports, routing tables and more

/all

ipconfig: MAC, IP, gateway, DHCP

hostname: hostname

netstart -ano: ports

Ping Hostname: IP connectivity ↙ incorrect IP config
 ↙ Physical connectivity issue

net view Hostname: shared resources

arp -a: available hosts

tracert URL : path

LAB 2

```
: #1: How to get Host name using IP address in python
```

```
import socket
hostname=socket.gethostname()
ip_address=socket.gethostbyname(hostname)
print("Hostname : {hostname}")
print("IP Address : {ip_address}")

Hostname : DESKTOP-19
IP Address : 192.168.56.1
```

```
: #4: How to find service name, given port number & protocol
```

```
import socket
def find_service_name():
    protocol='tcp'
    for port in [80 , 25]:
        print ("Port: %s => service name: %s" %(port,socket.getservbyport(port,protocolname)))
    print ("Port: %s => service name: %s" %(53,socket.getservbyport(53,'udp')))

find_service_name()

Port: 80 => service name: http
Port: 25 => service name: smtp
Port: 53 => service name: domain
```

```
: #5: Port scanner
from socket import *
```

```
import time
startTime = time.time()

target = input("Enter the host to be scanned: ")
t_IP= gethostname(target)
print ('starting scan on host: ', t_IP)

for i in range (50,500):
    s=socket(AF_INET,SOCK_STREAM)

    conn = s.connect_ex((t_IP, i)) # Use connect_ex to check for connection success

    if (conn == 0):
        print('Port %d: OPEN' % (i))

    s.close()

print('Time taken:', time.time() - startTime)

Enter the host to be scanned: localhost
starting scan on host: 127.0.0.1
```

```
#6: Simple Client Server connecting program
```

```
#Server:
```

```
import socket # Import socket module

s = socket.socket() # Create a socket object
print('Socket Created')
s.bind(('localhost', 1604)) # Bind to the port

s.listen(5) # Now wait for client connection.
print('waiting for connection')

while True:
    c, addr = s.accept() # Establish connection with client.
    print ('Got connection from', addr)
    c.send(bytes ('Thank you for connecting','utf-8'))
    c.close() # Close the connection
```

```
Socket Created
waiting for connection
Got connection from ('127.0.0.1', 62518)
```

```
#Client:
```

```
import socket # Import socket module

s = socket.socket() # Create a socket object
s.connect(('localhost', 1604))
print(s.recv(1024).decode())
s.close()
```

```
Thank you for connecting
```

```
#Design a simple Client Server chat application
```

```
#Server:
import socket
def start_server():
    host = 'localhost'
    port = 1555

    server_socket = socket.socket(socket.AF_INET, socket.SOCK_STREAM)
    server_socket.bind((host, port))
    server_socket.listen(5)

    print("Server listening on (%s):(%s)" % (host, port))

    while True:
        client_socket, addr = server_socket.accept()
        print("Connection from (%s)" % str(addr))

        # Communication Loop
        while True:
            message = client_socket.recv(1024).decode('utf-8')
            if not message:
                break # Break the loop if no message received
            print("Client: (%s)" % message)

            reply = input("Server (you): ")
            client_socket.send(reply.encode('utf-8'))

        client_socket.close()
        print("Connection closed")
```

```
start_server()
```

```
Server listening on localhost:1555
Connection from ('127.0.0.1', 62749)
Client: hello
Server (you): hi hows you
Client: i am thak gaya
Server (you): mee too
```

```
#Client
```

```
import socket
def start_client():
    host = 'localhost'
    port = 1555

    client_socket = socket.socket(socket.AF_INET, socket.SOCK_STREAM)
    client_socket.connect((host, port))

    # Communication Loop
    while True:
        message = input("Client (you): ")
        client_socket.send(message.encode('utf-8'))

        reply = client_socket.recv(1024).decode('utf-8')
        print("Server: (%s)" % reply)

    client_socket.close()
```

```
start_client()
```

```
Client (you): hello
Server: hi hows you
Client (you): i am thak gaya
Server: mee too
```

```
Client (you): 
```

LAB 3

network addresses for each pair:

IP Address: 188.10.18.2
Subnet Mask: 255.255.0.0
Network Address: 188.10.0.0

BC

IP Address: 10.10.48.80
Subnet Mask: 255.255.255.0
Network Address: 10.10.48.0

s

IP Address: 192.149.149.191
Subnet Mask: 255.255.255.0
Network Address: 192.149.24.0

IP Address: 150.203.23.19
Subnet Mask: 255.255.0.0
Network Address: 150.203.0.0

IP Address: 10.10.10.10
Subnet Mask: 255.0.0.0
Network Address: 10.0.0.0

IP Address: 186.13.23.110
Subnet Mask: 255.255.255.0
Network Address: 186.13.23.0

IP Address: 223.69.230.250
Subnet Mask: 255.255.0.0
Network Address: 223.69.0.0

IP Address: 200.120.135.15
Subnet Mask: 255.255.255.0
Network Address: 200.120.135.0

188.10.255.255

s

188.10.18.2

10.10.48.80

192.149.149.191

150.203.23.19

10.10.10.10

186.13.23.110

223.69.230.250

200.120.135.15

10.0.0.0

186.13.23.0

223.69.0.0

200.120.135.0

10.0.0.0

186.13.23.110

223.69.230.250

200.120.135.15

10.0.0.0

186.13.23.0

223.69.0.0

200.120.135.0

10.0.0.0

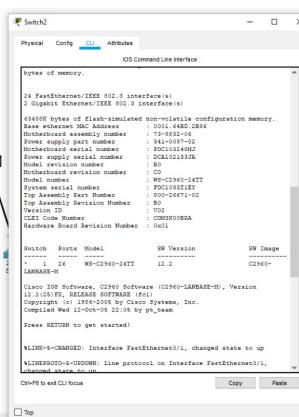
USER EXEC MODE: show interfaces

Privilege

:show running-config

Global Config

:interface GigabitEthernet 0/0



Privileged EXEC Mode: show running-config

switch#

route

IP Address Classes

Class A	1 – 127	(Network 127 is reserved for loopback and internal testing)
Class B	128 – 191	Leading bit pattern 10 10000000.00000000.00000000.00000000 Network . Host . Host . Host
Class C	192 – 223	Leading bit pattern 110 11000000.00000000.00000000.00000000 Network . Network . Network . Host
Class D	224 – 239	(Reserved for multicast)
Class E	240 – 255	(Reserved for experimental, used for research)

Private Address Space

Class A	10.0.0.0 to 10.255.255.255
Class B	172.16.0.0 to 172.31.255.255
Class C	192.168.0.0 to 192.168.255.255

Default Subnet Masks

Class A	255.0.0.0
Class B	255.255.0.0
Class C	255.255.255.0

Routers: chooses shortest path, and forwards, routes packet to other network → Packet switching

Switch: intelligent broadcast data overtime

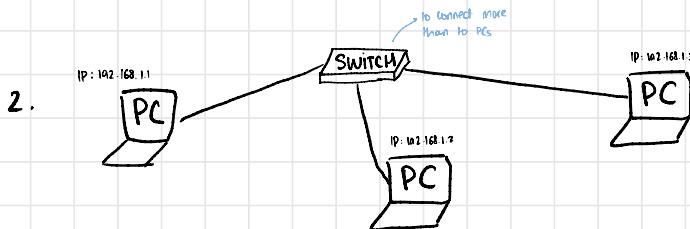
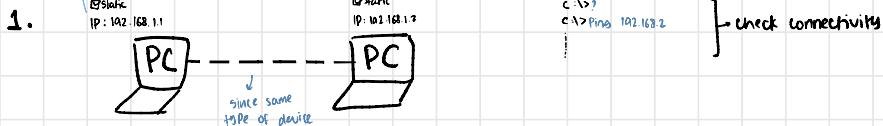
Hub: broadcast data

NIC: installed in computer, so computer can be connected to network

straight wire: connect different types of devices

crossover wire: connect same types of devices

Lab 1



IPv4: Network Part + Host Part

192.168.1.3 /24

32: $\frac{8}{NP} + \frac{24}{HP}$

same network

192.168.1.2 /24

192.168.1.55 /24

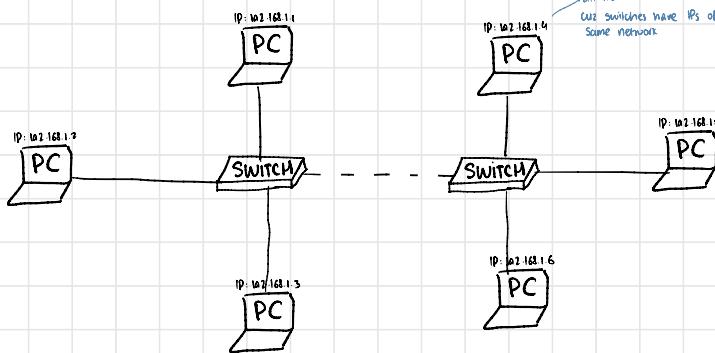
192.168.1.99 /24

192.168.1.9 /24

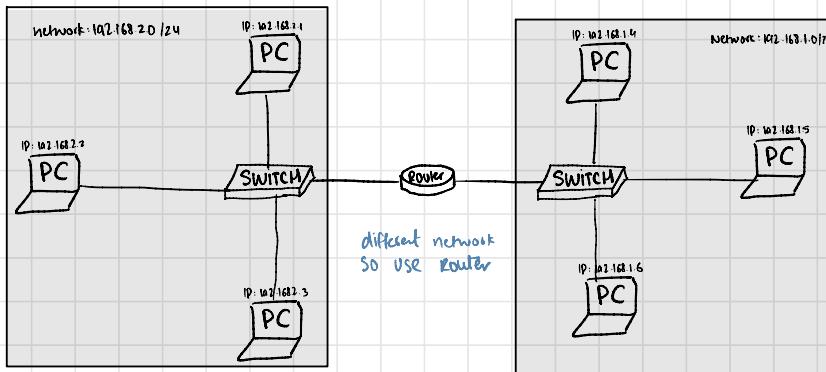
SubNet: 255.255.0.0
NP HP

Lab 2

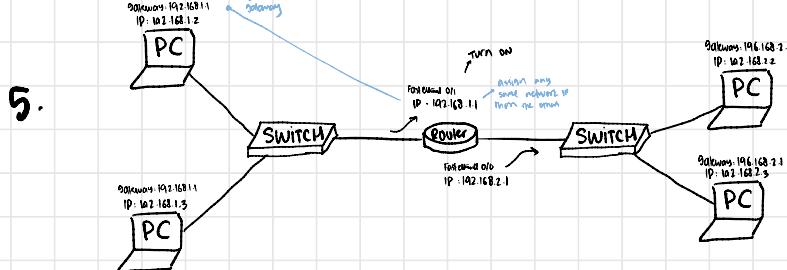
3.



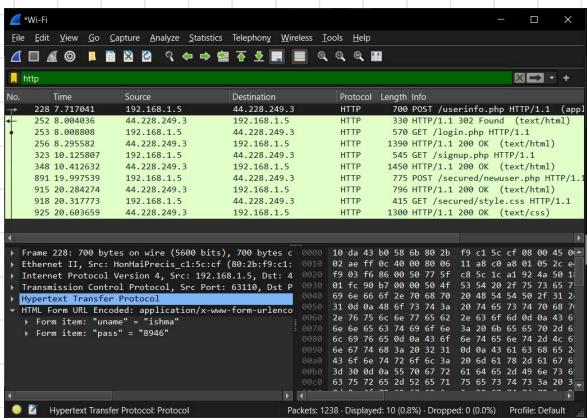
4.



Q&A 3



HTTP PACKET ON WIRE SHARK

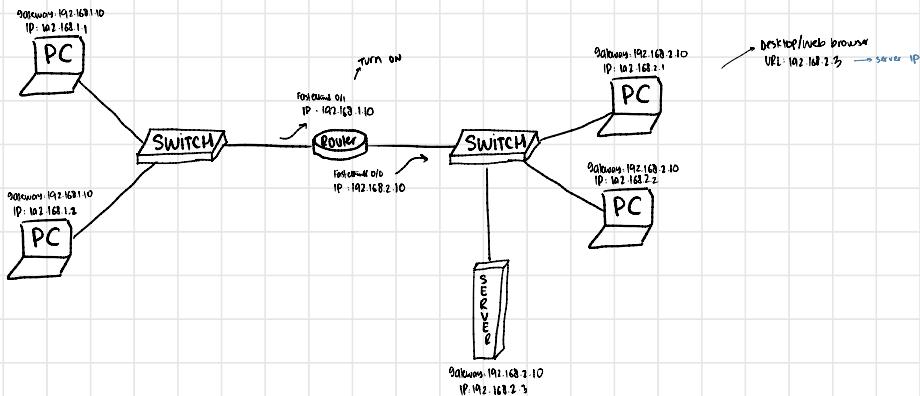


1 start

2 testphp.vulnweb.com
3 Filter http

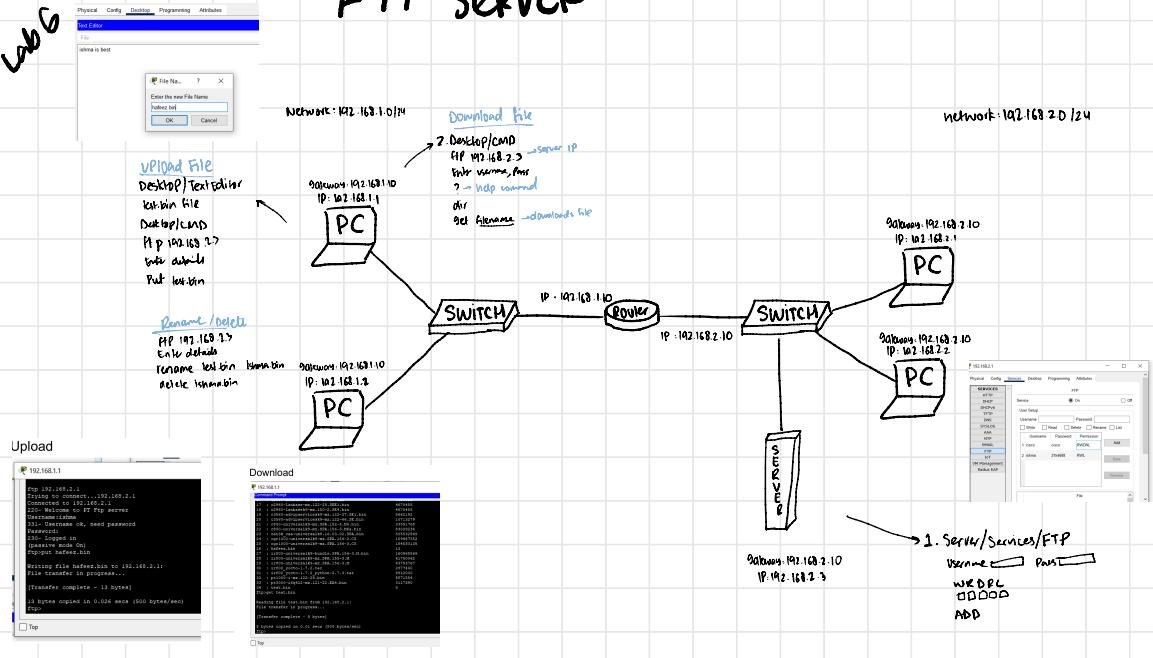
Network: 192.168.1.0/14

network: 192.168.2.0/24



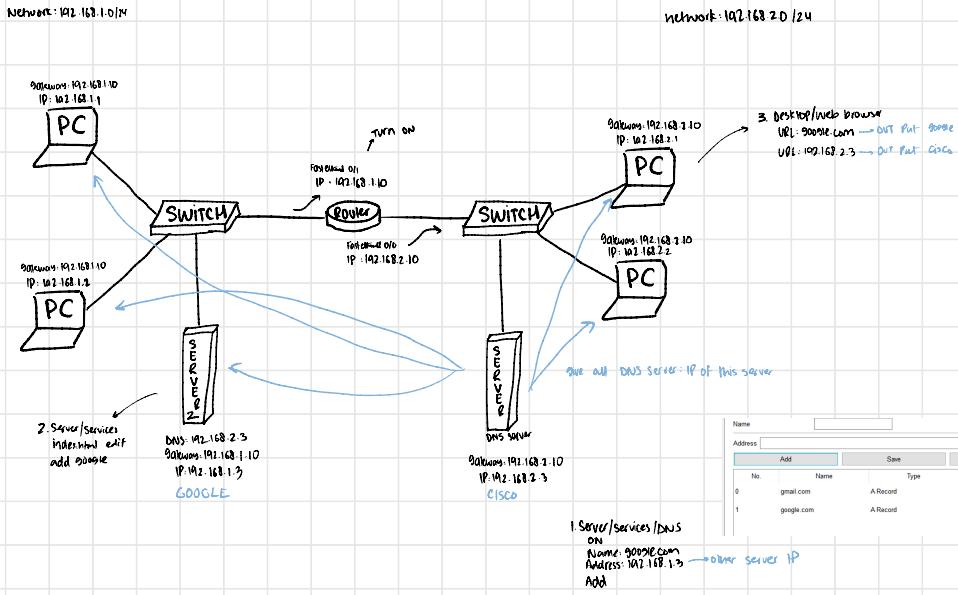
FTP SERVER

✓



DNS Server

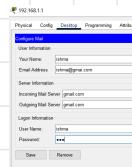
Lab 1



SMTP SERVER

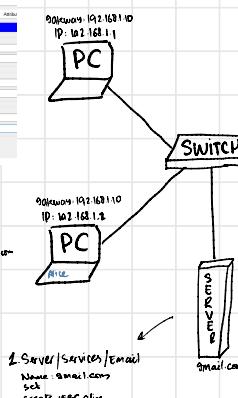
Network: 192.168.1.0/24

Network: 192.168.2.0/24

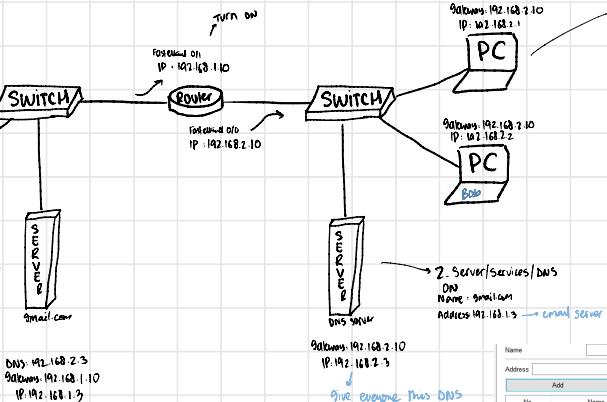
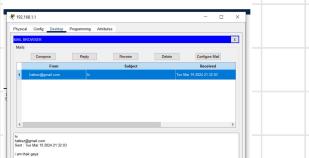


Alice
3. Desktop/Email
Name: Alice
Email: Alice@gmail.com
Username: gmail.com
Outgoing: gmail.com
User: alice
Port: 113
SSL

5. Receive



DNS: 192.168.2.3
Gateway: 192.168.1.10
IP: 192.168.1.3



Name	Type	Value
Address	Add	
0	Name	gmail.com
1	Name	google.com
	Type	A Record
	Detail	192.168.1.3
	Detail	192.168.1.3

DNS → Authoritative
name servers

its lab b

ishma hafeez
notes
represent

ACTUAL LAB 8

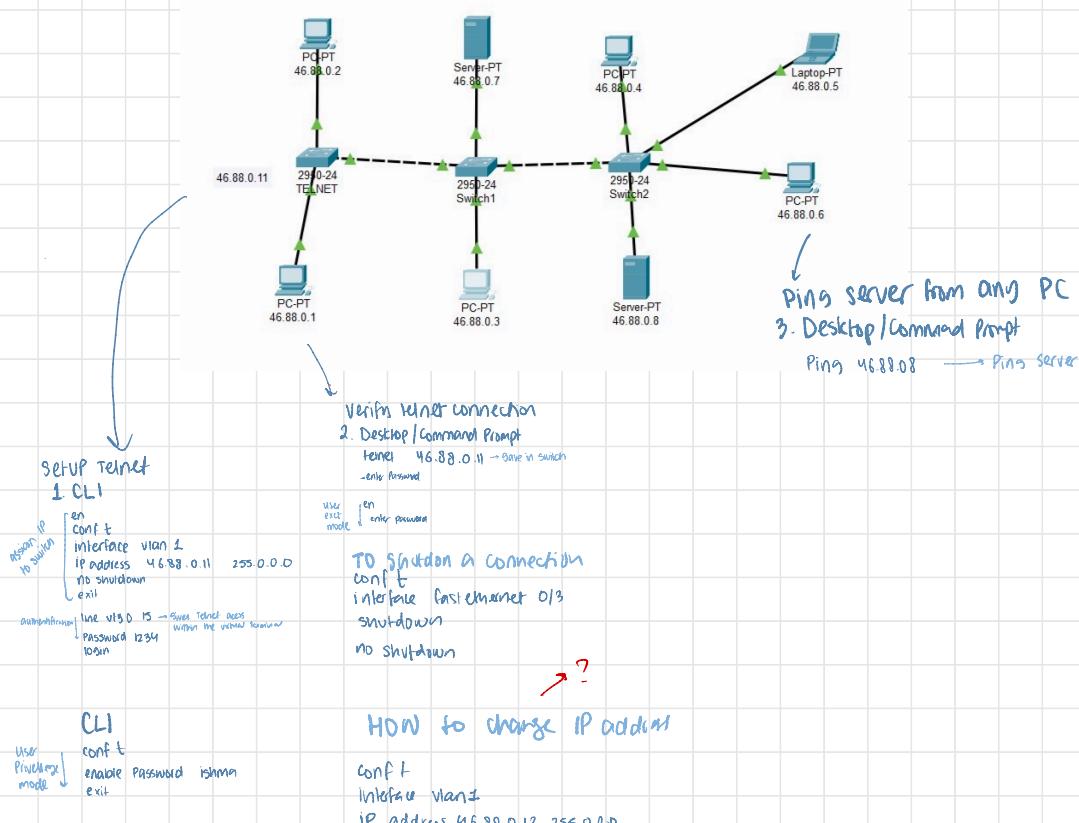
Telnet: a way to remotely control web servers
connects your PC to a server over the network

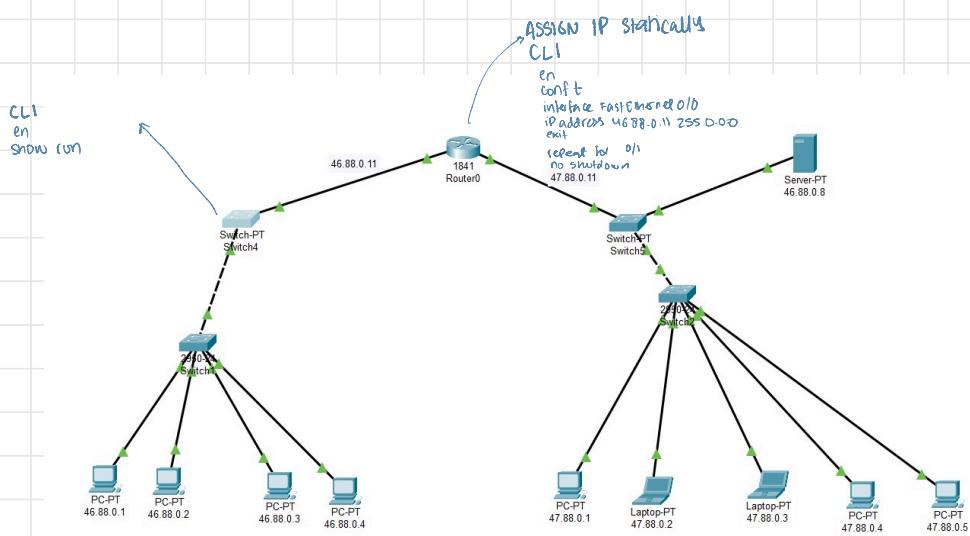
SSH (secure shell / secure socket shell)

An application layer Protocol

4th layer of OSI (open systems interconnection)

↳ supports password and key based authentication





ACTUAL LAB 9

video LAB 8

Subnet mask: 32 bits
IPV4 address = 32 bits

written as
8 bit numbers

↓
Octet - IP address 8 bits → in decimal → $2^7 \dots 2^0$
max value: 255
min value: 0

IPV4: Network Part + Host Part

192.168.1.3 /24
32: NP HP
 8 + 24

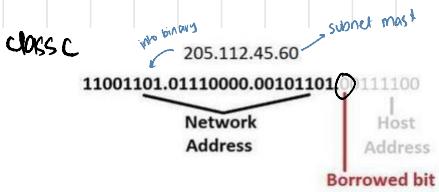
same network

192.168.1.2 /24
192.168.1.55 /24
192.168.1.99 /24
192.168.1.9 /24

Subnet: 255.255.0.0
 NP HP

Subnet

- ↳ a network inside a network
- ↳ make network more efficient
- ↳ a range of IP addresses
- ↳ all devices in same subnet can comm. directly with one another w/o going through any routers
- ↳ divides network into smaller portions called subnets

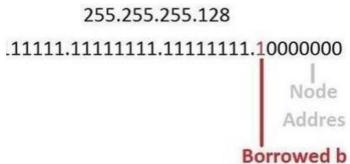


IP address components

- ↳ network part
- ↳ subnet part
- ↳ host part

create subnet by

- ↳ last bit of network : no of subnets req,
- ↳ left most bit of host: identify subnet
 - ↳ 0 → one subnet
 - ↳ 1 → second subnet



Address Class	Value in First Octet	Classful Mask (Dotted Decimal)	Classful Mask (Prefix Notation)
Class A	1–126	255.0.0.0	/8
Class B	128–191	255.255.0.0	/16
Class C	192–223	255.255.255.0	/24
Class D	224–239	—	—
Class E	240–255	—	—

Classification of IP Address

Table 3: Classification of IP address

Class	Representation based on Left Most Octet	Address Range		Support
		Address Range	Support	
A	0xxxxxx.xxxxxxxxx.xxxxxxxxx.xxxxxxxx	1.0.0.1 to 126.255.255.254	127 networks, 16777214 hosts	
B	10xxxxxxxx.xxxxxxxxx.xxxxxxxxx.xxxxxxxx	128.1.0.1 to 191.255.255.254	16384 networks, 65534 hosts	
C	110xxxxxxxx.xxxxxxxxx.xxxxxxxxx.xxxxxxxx	192.0.1.1 to 223.255.254.254	2097152 networks, 254 hosts	

2 bits is reserved for prefix identification

Networks	$2^2 = 4$
Masks	$2^{24} = 16777216 - 2$
	$2^{14} = 65536 - 2$
	$2^{11} = 2097152$
	$2^8 = 256 - 2 = 254$

network, broadcast ID

192.168.4.0/24 create 3 subnets
Sunny Subnetting Table

Original networkID:
192.168.4.0/24

Subnet	2	0	16	32	48	64	70
Host	256	128	64	32	16	8	4
Subnet掩码	/24	/25	/26	/27	/28	/29	/30

Network ID	Subnet Mask	Host ID Range	# of Usable Host	Broadcast ID
192.168.4.0	/26	192.168.4.1-192.168.4.62	62	192.168.4.63
192.168.4.64	/26	192.168.4.65-192.168.4.126	62	192.168.4.127
192.168.4.128	/26	192.168.4.129-192.168.4.190	62	192.168.4.191
192.168.4.192	/26	192.168.4.193-192.168.4.254	62	192.168.4.255

192.168.100.0 /24 → class C

4 sub networks?

↓
2² → no of bits
no borrowed

only convert host portion
to binary
↓
24
192.168.100.0000 0000 /24
 $24+2 = 26$
can change

192.168.100.0000 0000 /26
192.168.100.0100 0000 /26
192.168.100.1000 0000 /26
192.168.100.1100 0000 /26

to decimal

192.168.100.0 /26
192.168.100.64 /26
192.168.100.128 /26
192.168.100.192 /26

To find Range

1) Host = $\frac{256}{4}$ → total no of host → class C
→ no of sub networks

= 64 → add to host

0 to 64 = 63

192.168.100.0 /26 - 192.168.100.63 /26
192.168.100.64 /26 - 192.168.100.127 /26
192.168.100.128 /26 - 192.168.100.191 /26
192.168.100.192 /26 - 192.168.100.255 /26

To find Subnet mask

2) 192.168.100.0 /26
first 26 bits 1
the rest 0

Subnet mask = 1111 1111. 1111 1111. 1111. 1100 0000
255. 255. 255. 192
convert to decimal

NID	BID	Usable Range	Subnet mask
192.168.100.0	192.168.100.63	100.1 - 100.62	255.255.255.192
192.168.100.64	192.168.100.127	100.65 - 100.126	255.255.255.192
192.168.100.128	192.168.100.191	100.129 - 100.190	255.255.255.192
192.168.100.192	192.168.100.255	100.193 - 100.254	255.255.255.192

↓ subnet same for all

Sunny Subnetting Table

Subnet	1	2	4	8	16	32	64	128	256
Host	256	128	64	32	16	8	4	2	1
Subnet Mask	/24	/25	/26	/27	/28	/29	/30	/31	/32

Q) 192.168.100.0 /24 → class C
4 sub networks?

Subnet = 4

Possible hosts - 64

Usable hosts - 62

network ID

NID × 64

192.168.100.0

subnet mask

255.255.255.192

subnet same for all

??

192.168.100.64

192.168.100.128

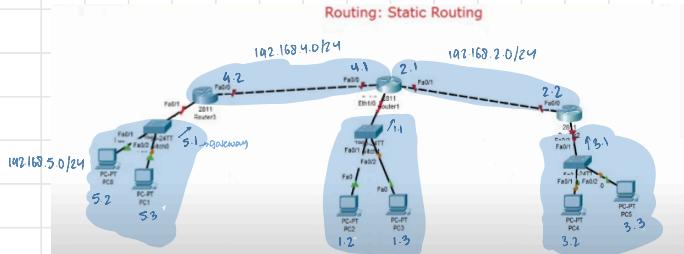
192.168.100.192

Display this

host

host</p

STATIC ROUTES



5 networks

24

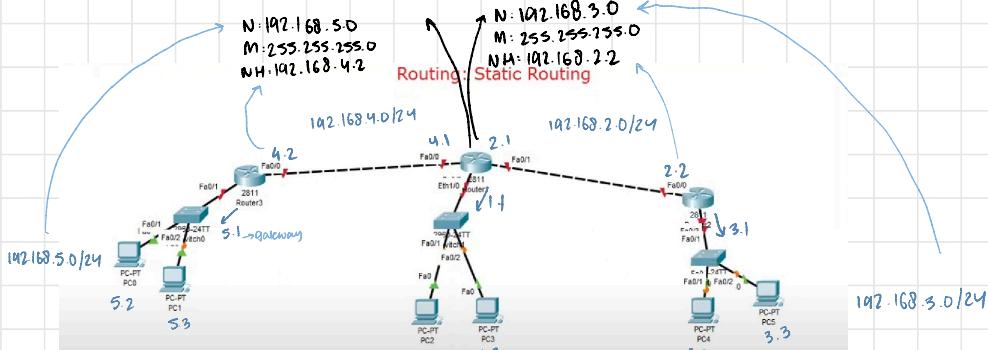
1111 1111 1111 1111.0000 0000
255.255.255.0

192.168.3.0/24

192.168.1.0/24

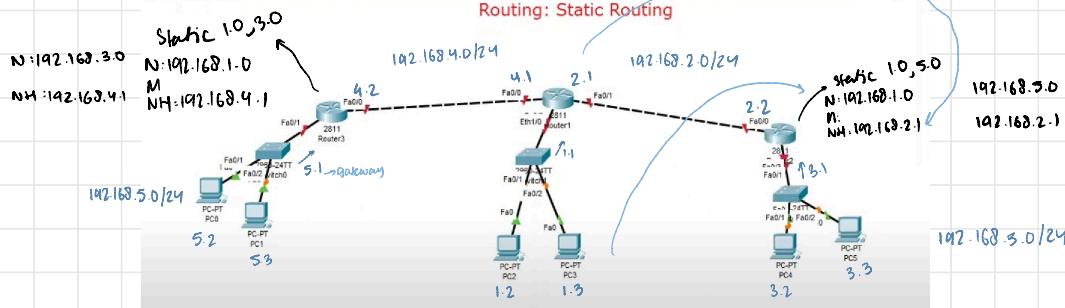
Has no subnetting
So use static routes

static so 30



192.168.1.0/24

Routing: Static Routing



192.168.1.0/24

Q) 192.168.1.0/27 OF class C

using sumy subnetting table

subnet = 8

Possible host = 32

Usable host = 32 - 2 = 30

NID

192.168.1.0
192.168.1.32
192.168.1.64
192.168.1.96

Subnet mask

255.255.255.224
11111111.11111111.11111111.11100000

HID	+1 NID	-1 BID	broadcast ID	-1 NID
1.1	— 1.30	192.168.1.31		
1.33	— 1.62	192.168.1.63		
1.65	— 1.94	192.168.1.95		
1.97	— 1.126	192.168.1.127		

6. Implementation of Subnetting in Packet Tracer:

Create an IP of Class C 192.168.1.0/27, using above IP calculate the subnets and implement the scenario in Cisco Packet Tracer.

192.168.1.0 /27
255.255.255.224
11111111.11111111.11111111.11100000
Networks
92 - 64 = 128 - 192
64 - 32 = 192 - 224
32 - 16 = 224 - 256

Fig 6. Create Subnet

Calculation:
From 192.168.1.0/24
Total no of subnets = 2^(24-27) = Possible Hosts = 12
Available Hosts in each Subnet = 2^(24-27) = 30

Note: IP address of every subnet shows network address and last address shows Broadcast address, e.g. 192.168.1.32/27 shows network address 192.168.1.32 and broadcast address 192.168.1.63
Current Subnet Mask = 255.255.255.224

New Implementing below figure (Scenario on Cisco packet Tracer).

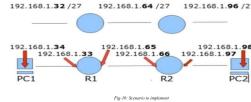
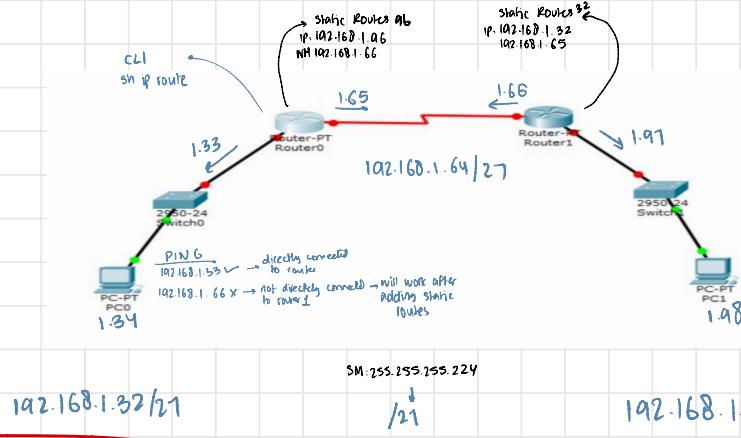
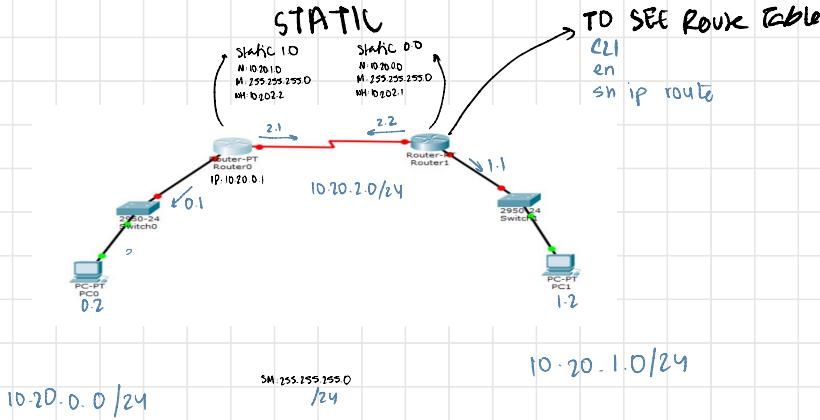


Fig 6. Scenario implemented

static route



Network Address = 10.20.0.0/24



Q) 3 Departments, perform Subnetting for the allocation of given resources

Network address: 200.16.100.0/24

↳ D → 120 PCs

↳ F → 35 PCs

↳ HR → 10 PCs

subnet the given network address 200.16.100.0/24:

1. Development: Subnet size = 128 → Subnet mask = /25 (because $24 + 7 = 31$)
Subnet range: 200.16.100.0/25

2. Finance: Subnet size = 64 → Subnet mask = /26 (because $24 + 6 = 30$)
Subnet range: 200.16.100.128/26

3. HR: Subnet size = 16 → Subnet mask = /28 (because $24 + 4 = 28$)
Subnet range: 200.16.100.192/28

The subnetting allocation for ABC Company's departments is:

- Development: 200.16.100.0/25 (Subnet mask: 255.255.255.128)

- Finance: 200.16.100.128/26 (Subnet mask: 255.255.255.192)

- HR: 200.16.100.192/28 (Subnet mask: 255.255.255.240)

Q) 192.168.1.0/24

2 D → 60 PCs, $2^6 = 64$

1 F → 28 PCs, $2^5 = 32$

1 HR → 16 PCs, $2^4 = 16$
BID NID
- 2

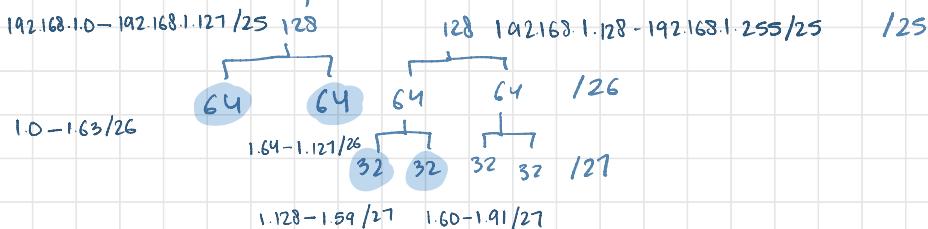
$$64 + 64 + 32 + 32 = 192$$

/24 = 256 resources

$$192 < 256$$

192.168.1.0 - 192.168.1.255 /24 /24

256



27 Subnet mask

1111.1111.1111.1111.1111.00000

.224

Given the network address 10.0.0.0/8, you need to create subnets for six different departments: Administration (100 hosts), Development (50 hosts), Testing (30 hosts), Finance (20 hosts), HR (10 hosts), and Sales (5 hosts). Calculate the subnet mask for each department.

10.0.0.0/8

11111111
255.0.0.0

DA	100	$2^6 : 128$
D	50	$2^5 : 64$
T	30	$2^5 : 64$
F	20	$2^4 : 32$
HR	10	$2^4 : 16$
S	5	$2^3 : 8$

Q) 3 Departments, Network : 195.168.10.0 /24

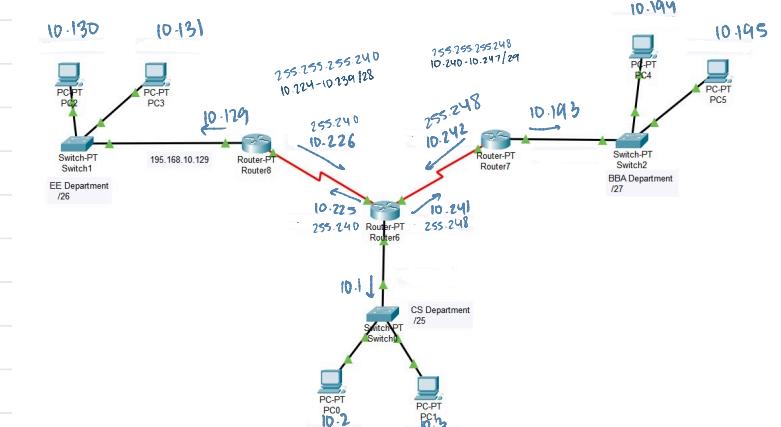
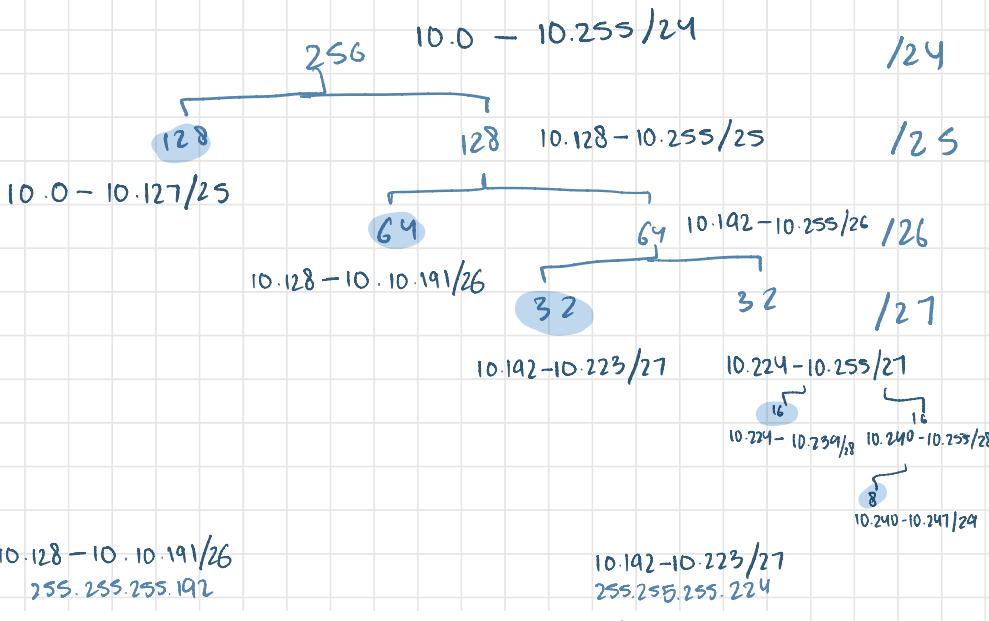
	2^7	2^6	2^5	2^4	2^3	2^2	2^1	2^0
CS → 90 PCs								
EE → 50 PCs								
BBA → 20 PCs								
$128 + 64 + 32 = 224$	128	64	32	$10.0 - 10.127/25$	$10.128 - 10.191/26$	$10.192 - 10.223/27$	$10.224 - 10.255/28$	$10.240 - 10.247/29$
				$10.0 - 10.127/25$	$10.128 - 10.191/26$	$10.192 - 10.223/27$	$10.224 - 10.255/28$	$10.240 - 10.247/29$

24
NETMASK SUBNET MEDIUM

255.255.255.128

255.255.255.192

255.255.255.224

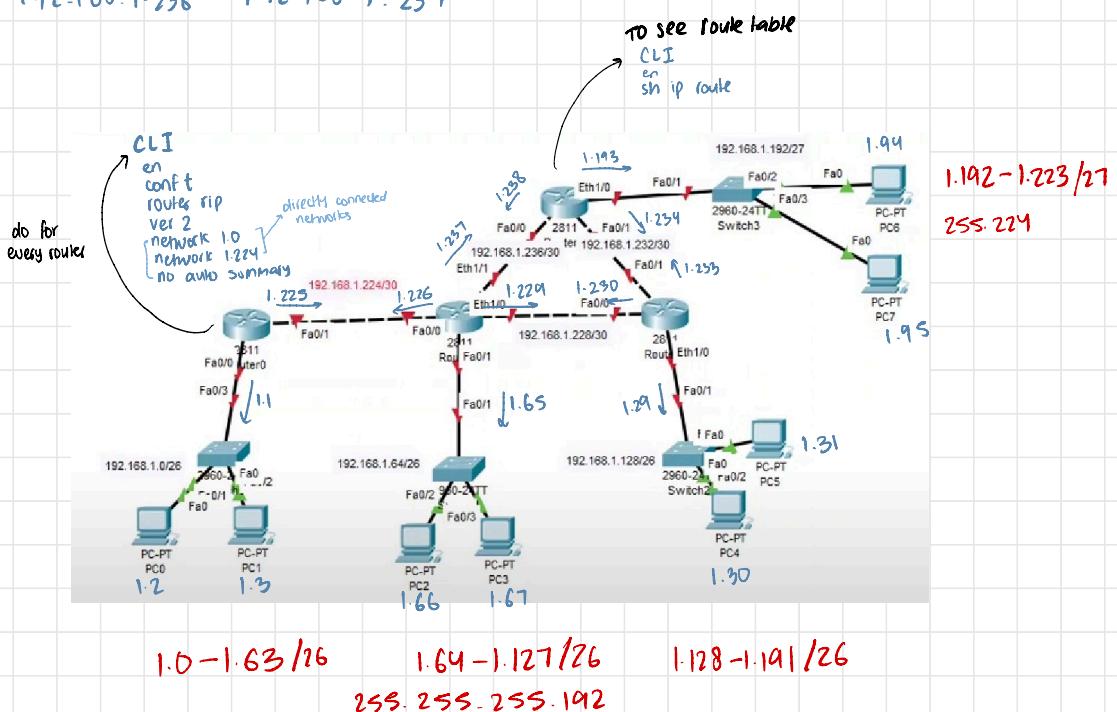
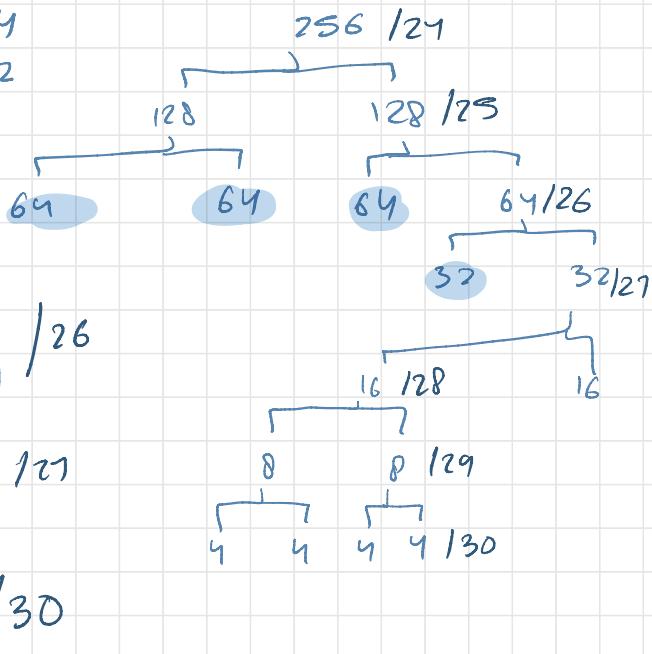


255.255.255.128

10.0 - 10.127/25

DYNAMIC ROUTING

Q) 3 Networks 42 PCs $2^6 = 64$
 1 Network 22 PCs $2^5 = 32$
 $64 + 64 + 64 + 32 = 194$



Advantages	Disadvantages
Easy to implement in a small network.	Suitable for simple topologies or for special purposes such as a default static route.
Very secure. No advertisements are sent, unlike with dynamic routing protocols.	Configuration complexity increases dramatically as the network grows. Managing the static configurations in large networks can become time consuming.
It is very predictable, as the route to the destination is always the same.	If a link fails, a static route cannot reroute traffic. Therefore, manual intervention is required to re-route traffic.
No routing algorithm or update mechanisms are required. Therefore, extra resources (CPU and memory) are not required.	

Table-1: Advantages & Disadvantages of Static Routing

Route Source	Default AD
Connected interface	0
Static route	1
EIGRP	90
IGRP DV	100
OSPF LS	110
RIP 1 DV	120
External EIGRP	170

Table-2: Administrative Distances

The smaller the AD is, the more preferable to route is.

Distance Vector (DV)

↳ routers with least hops

each time a packet goes through a router

HIBRID

↳ DV + LS

Link State (LS)

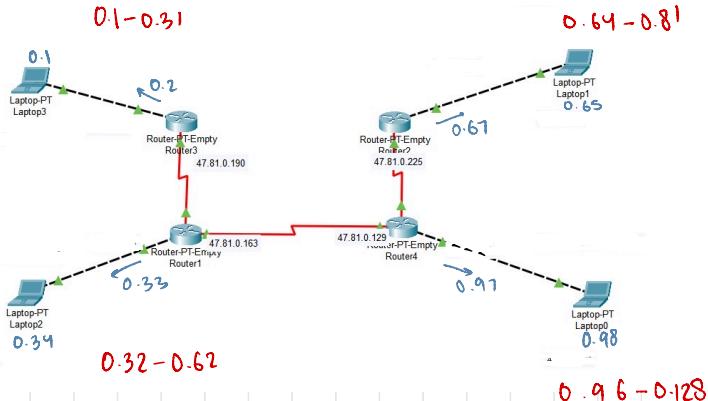
↳

0) 47.81.0-0/24

0.0
0.32
0.64
0.128

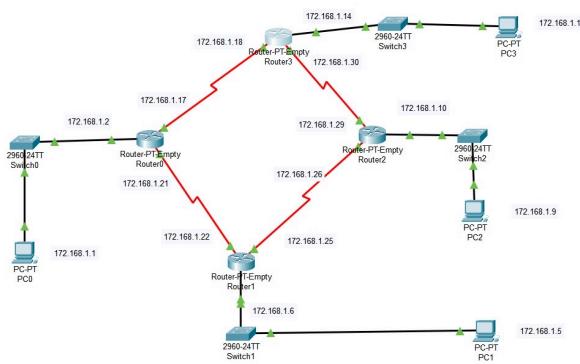
Q1: Implement Subnetting with IP address of XX.XX.0.0/24. where xx is your roll no same are midterm. Then assign ip address in such a way that very less ip address should waste. Last run RIP routing protocol in such a way that all devices can communicate easily. What will be the administrative distance of the routing? Use figure 8 for your reference.

Subnet = 255.255.255.224



Q2: Implement Subnetting with IP address of 172.168.1.0/24. All the assignment of IP should be done dynamically in such a way that there should be less waste of IPs. Run the dynamic routing protocol with less administrative distance. What will be the administrative distance of the routing? Use Figure 9 as reference.

Subnet: 255.255.255.252



Q3: Implement the subnetting on the given scenario of figure 10. You have to implement the static routing on the same. What will be the administrative distance of the routing?
Use Network Address as follows: 192.168.4.0/24.

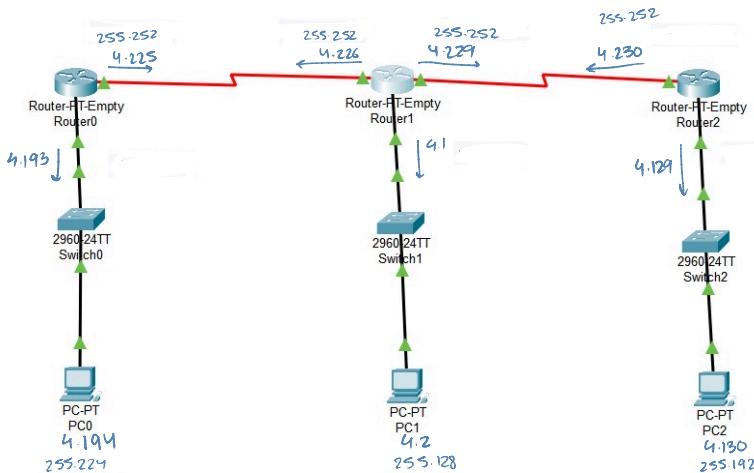


Fig-10: Network Topology for Q3

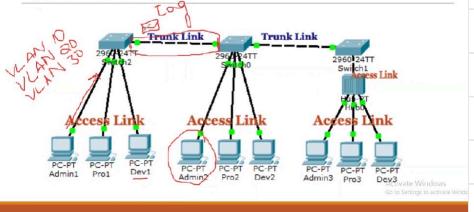
Q4: In what case we use static routing or dynamic routing given a topological reason for this question.

TASK 4:

Static Routing: Static routing suits smaller, less complex networks, offering simple implementation and enhanced security with predictable routes. However, it becomes cumbersome in larger setups, requiring manual configuration and lacking automatic rerouting capabilities.

Dynamic Routing: Dynamic routing is ideal for larger networks with changing topologies, ensuring scalability and resilience. It adapts to network changes automatically but demands more resources and introduces potential security risks from routing advertisements. Configuration complexity may also arise in diverse network environments.

Types Of Links



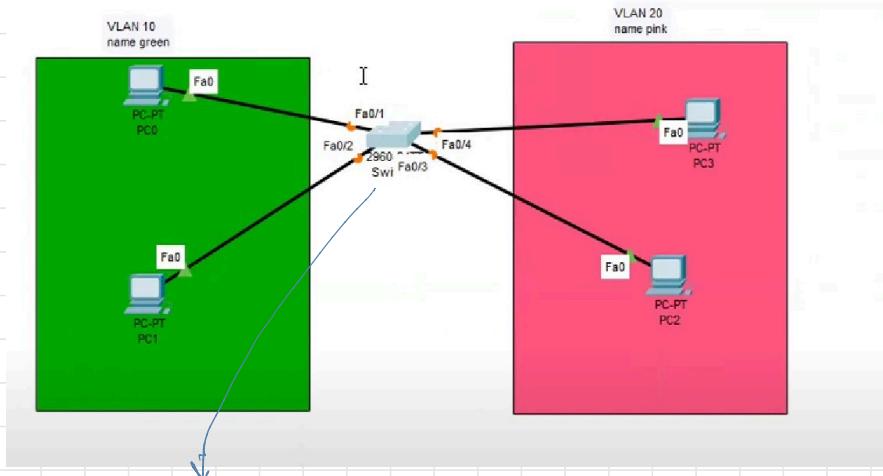
Difference between LAN and VLAN

LAN

All configuration is done in PC	Configuration is done in both PC and the Switch
In LAN we cannot configure ports available in a switch	VLAN allows you to configure each and every port available in a switch
The router is required to communicate between two different networks.	Even without Router, two or more virtual networks can communicate. All the configuration is done in the switch itself.
Hence all the data travels in the same broadcast domain, the network traffic is more	Each VLAN has their own broadcast domain. Hence, the network traffic is less
Network cost is high.	Network cost is low.
Network construction is simple.	Network construction is complex.

VLAN

Activate Windows
Go to Settings to activate Windows



CLI

en conf t

Vlan 10
name green

ex

int fa0/1
Switchport mode access
Switchport access vlan 10
ex

int fa0/2
Switchport mode access
Switchport access vlan 10
ex

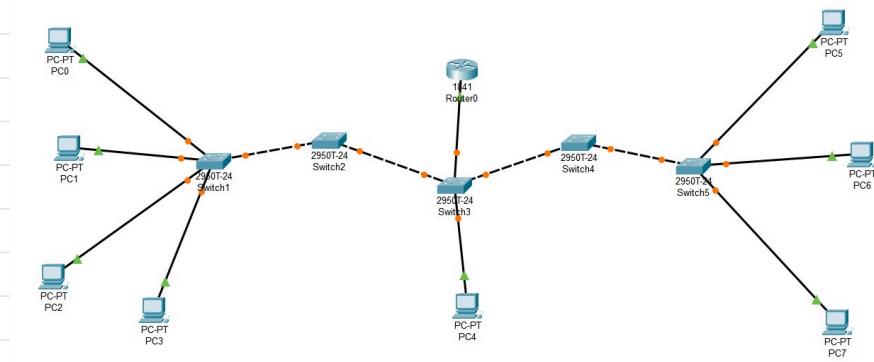
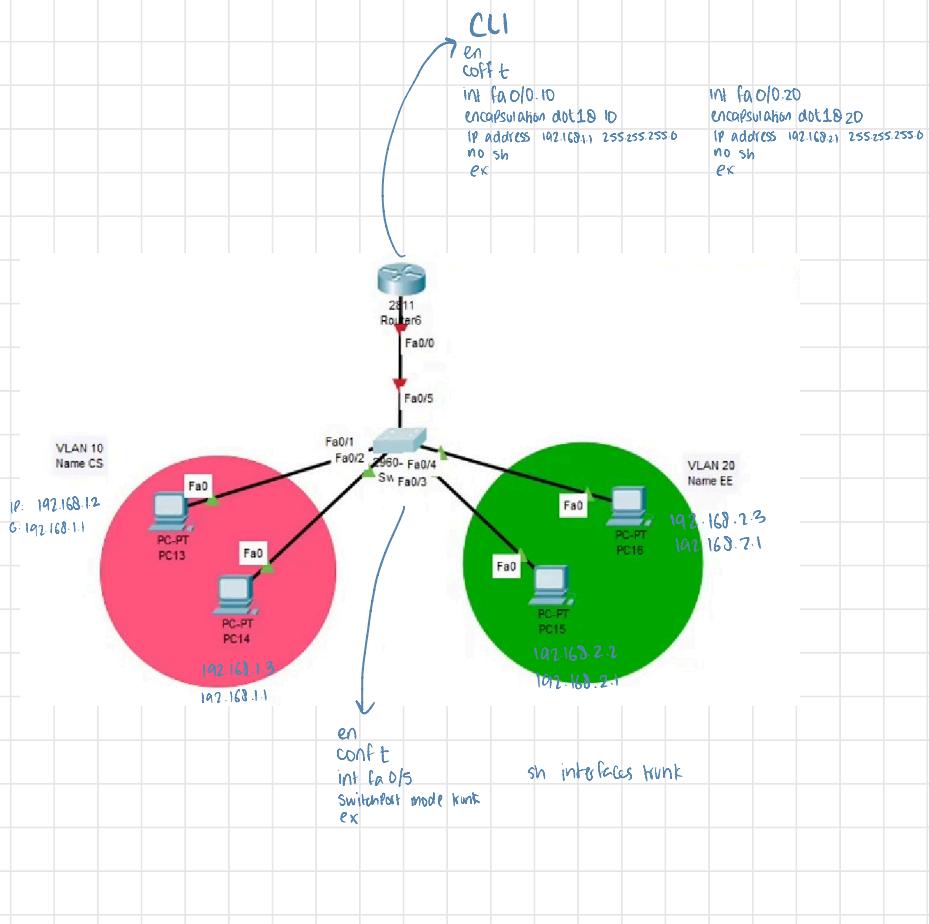
Vlan 20

name pink

ex

int fa0/1
Switchport mode access
Switchport access vlan 20
ex

int fa0/2
Switchport mode access
Switchport access vlan
ex



Gufran Ahmed

Overview/roadmap:

- What *is* the Internet? What *is* a protocol?
- Network **edge**: hosts, access network, physical media
- Network **core**: packet/circuit switching, internet structure
- Performance: loss, delay, throughput
- Protocol layers, service models
- Security
- History

INTERNET: NUTS AND BOLTS VIEW

Packet switches

- ↳ chunks of data
- e.g. routers, switches



Communication links

- e.g. fiber, radio, satellite



Networks

- collection of devices, routers, links
- managed by an organization

Internet

- ↳ network of networks
- Interconnected ISPs

Protocols

- ↳ control sending, receiving of messages
- e.g. HTTP, skype, streaming video, WiFi, 4G, Ethernet

Internet standards

- ↳ ICF (Request for comments)
- IETF (Internet Engineering Task Force)

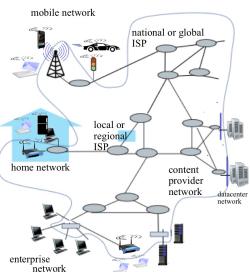
INTERNET: SERVICES VIEW

Infrastructure

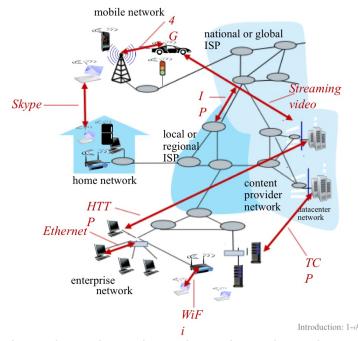
- ↳ provides services to applications
- e.g. streaming video, social media, games
- e-commerce

Programming Interface

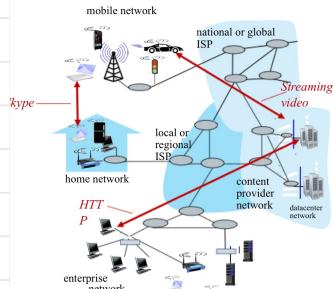
- ↳ provided to distributed applications
- hooks: allows apps to connect to Internet transport service
- ↳ provides service options
- ↳ analogous to postal service



NUTS AND BOLTS VIEW



SERVICE VIEW



PROTOCOLS

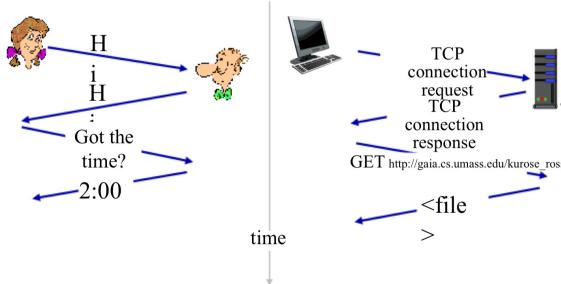
↳ defines format, order of messages sent and received, receipt, actions taken on msg transmission

HUMAN PROTOCOL

- “what’s the time?”
- “I have a question”
- introductions

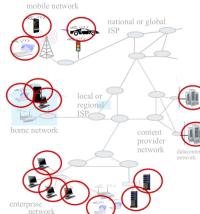
NETWORK PROTOCOL

- ↳ b/w computers
- ↳ all comm activity on the Internet is governed by protocols



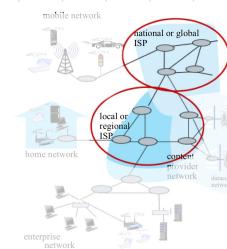
NETWORK EDGE

- ↳ hosts: clients and servers
- ↳ often in data centers



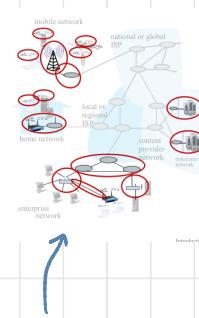
NETWORK CORE

- ↳ inter connected routers
- ↳ network of networks



ACCESS NETWORKS, PHYSICAL MEDIA

- ↳ wired comm. link
- ↳ wireless comm. link



Q: How to connect end systems to edge router?

- residential access nets
- institutional access networks (school, company)
- mobile access networks (WiFi, 4G/5G)

Wireless Access Networks

↳ connects end system to router

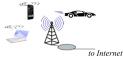
Wireless local area networks (WLANs)

- typically within or around building (~100 m)
- 802.11b/g/n (WiFi): 11, 54, 450 Mbps transmission rate



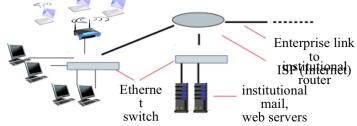
Wide-area cellular access networks

- provided by mobile cellular network operator (10's km)
- 10's Mbps
- 4G cellular networks (5G coming)



Enterprise Networks

↳ mix of wired + wireless technologies
↳ connects mix of switches and routers
e.g. universities, companies



Access Networks

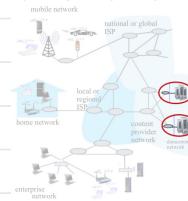
Data centre Networks

↳ high bandwidth links

↳ connects thousands of servers together and to the internet



Courtesy: Massachusetts Green High Performance Computing Center (mghpcc.org)



Ethernet: wired access at 100Mbps, 1 Gbps, 10 Gbps

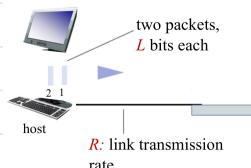
WiFi: wireless access at 11, 54, 450 Mbps

Host sending function

↳ sends packets of data

Steps

1. take application message
2. breaks into packets → smaller chunks → of length 'L' bits
3. transmits packet into access network → at transmission rate 'R'



$$\text{Packet transmission delay} = \frac{L}{R}$$

↳ time to transmit L bits into link at R bps

One-hop numerical example:

- $L = 10$ Kbits
- $R = 100$ Mbps
- one-hop transmission delay = 0.1 msec

Links: Physical Media

- bit: propagates between transmitter/receiver pairs
- physical link:** what lies between transmitter & receiver
- guided media:**
 - signals propagate in solid media: copper, fiber, coax
- unguided media:**
 - signals propagate freely, e.g., radio

Twisted pair (TP)

- two insulated copper wires
- Category 5: 100 Mbps, 1 Gbps Ethernet
- Category 6: 10Gbps Ethernet



Coaxial cable:

- two concentric copper conductors
- bidirectional
- broadband:
- multiple frequency channels on cable
- 100's Mbps per channel



Fiber optic cable:

- glass fiber carrying light pulses, each pulse a bit
- high-speed operation:
- high-speed point-to-point transmission (10's-100's Gbps)
- low error rate:
- repeaters spaced far apart
- immune to electromagnetic noise



Wireless radio

- signal carried in various "bands" in electromagnetic spectrum
- no physical "wire"
- broadcast, "half-duplex" (sender to receiver)
- propagation environment effects:
 - reflection
 - obstruction by objects
 - Interference/noise

Radio link types:

- Wireless LAN (WiFi)
 - 10-100's Mbps, 10's of meters
- wide-area (e.g., 4G cellular)
 - 10's Mbps over ~10 Km
- Bluetooth: cable replacement
 - short distances, limited rates
- terrestrial microwave
 - point-to-point; 45 Mbps channels
- satellite
 - up to 45 Mbps per channel
 - 270 msec end-end delay

Network Core

↳ mesh of interconnected routers



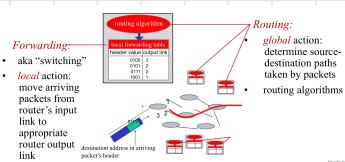
1. Routing ↳ Global action

↳ determines source destination Paths taken by Packets

↳ routing algos

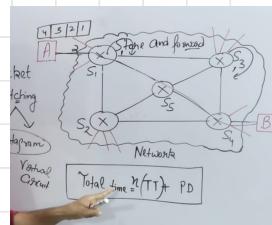
2. Forwarding ↳ Local action

↳ moves arriving packets from router's input link to appropriate router output link



Packet switching

- ↳ host breaks application layer into packets
- ↳ network forwards packets router to router across links from source to destination



WHAT?

- 1) Data Link and Network layer
- 2) Store and forward
- 3) Pipelining used
- 4) Efficiency ↑
- 5) Delay ↑

Store and forward

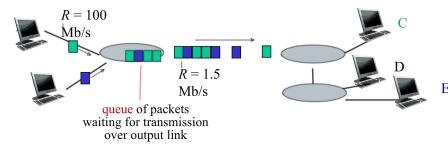
- ↳ entire packet must arrive at router before it can be transmitted on next line



Queuing

- ↳ when work arrives faster than it can be served
- ↳ if arrival rate exceeds transmission rate

1. Packets will queue Waiting to be transmitted on output link
2. Packets can be lost if buffer in Router fills up

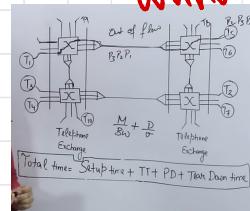
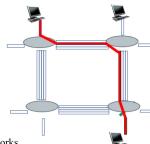


WHICH?

Circuit switching

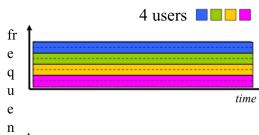
end-end resources allocated to, reserved for "call" between source and destination

- in diagram, each link has four circuits.
- call gets 2nd circuit in top link and 1st circuit in right link.
- dedicated resources: no sharing
- circuit-like (guaranteed) performance
- circuit segment idle if not used by call (no sharing)
- commonly used in traditional telephone networks



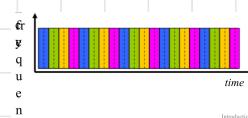
- Physical Layer
- Contiguous Flow
- No Reorder
- Efficiency ↓
- Delays ↓

Frequency Division Multiplexing (FDM)



- ↳ Optical, electromagnetic frequencies divided into frequency bands
- ↳ each call allocated its own band
- ↳ can transmit max rate of frequency band

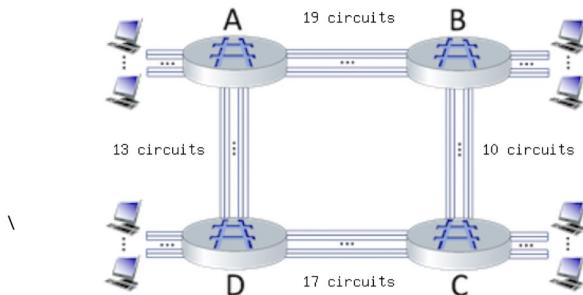
Time Division Multiplexing (TDM)



- ↳ time divided into slots
- ↳ each call allocated periodic slots
- ↳ can transmit max rate of frequency band only during its time slot

Question 1: Circuit Switching**(10) Points**

Explain Circuit Switching. Consider the circuit-switched network shown in the figure below, with circuit switches A, B, C, and D. Suppose there are 19 circuits between A and B, 10 circuits between B and C, 17 circuits between C and D, and 13 circuits between D and A. What is the maximum number of connections that can be ongoing in the network at any one time?



There are no bottlenecks in the given network so all you have to do is sum the individual circuit counts.

$$\text{Total Circuits} = 13 + 19 + 17 + 10 = 59$$

PACKET SWITCHING VS CIRCUIT SWITCHING

PROS

↳ good for bursty data

sometimes has data to send
sometimes not

↳ resource sharing

↳ simple *no call setup*

CONS

↳ packet delay

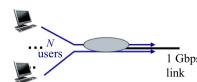
not good for excessive congestion

↳ packet loss

due to buffer overflow

example:

- 1 Gb/s link
- each user:
- 100 Mb/s when "active"
- active 10% of time



Q: how many users can use this network under circuit-switching and packet switching?

- **circuit-switching:** 10 users

- **packet switching:** with 35 users, probability > 10 active at same time is less than .0004 *

- **Q:** how did we get value 0.0004?
A: HW problem (for those with course in probability only)

* Check out the online interactive exercises for more examples: http://gaias.cs.umass.edu/kurose_ross/interactive

Internet Structure: a network of networks

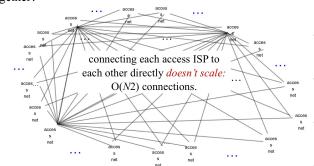
↳ host connects to Internet via access Internet Service Providers (ISPs)

↳ ISPs must be interconnected so that

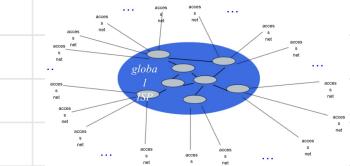
any two hosts can send packets to each other

↳ results in network of networks

networks: given millions of access ISPs, how to connect them together?

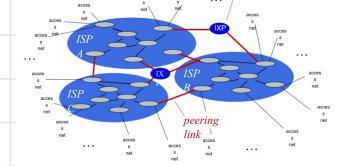


WORKS: connect each access ISP to one global transit ISP? Customer and provider ISP's have economic agreement.



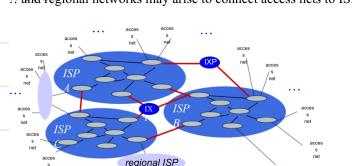
WORKS: global ISP is viable business, there will be competitors who will want to be connected

Internet exchange point



works?

and regional networks may arise to connect access nets to ISPs



networks: provider networks (e.g., Google, Microsoft, Akamai) may run their own network, to bring services, content close to end users



NETWORKS

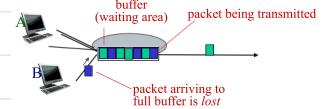


At "center": small # of well-connected large networks

- "tier-1" commercial ISPs (e.g., Level 3, Sprint, AT&T, NTT), national & international coverage
- content provider networks (e.g., Google, Facebook): private network that connects its data centers to Internet, often bypassing tier-1, regional ISPs

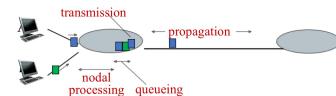
HOW PACKET DELAY AND LOSS OCCURS

- ↳ Packets queue in routers buffers, waiting for their turn for transmission
- ↳ When a packet arrives and buffer is full the packet is lost
- ↳ Lost packet may be retransmitted
 - ↳ by previous node
 - ↳ by source end system
- ↳ or not at all



Packet delay: 4 resources

1. **Transmission Delay:** Packet being transmitted → d_{trans}
2. **Queueing Delay:** Packets in buffers → d_{queue}
3. **Nodal Processing Delay:** → d_{proc}
4. **Propagation Delay:** → $d_{prop} = \frac{d}{S}$



$$dnodal = d_{proc} + d_{queue} + d_{trans} + d_{prop}$$

- d_{proc} : nodal processing
 - check bit errors
 - determine output link
 - typically < microseconds

- d_{queue} : queueing delay
 - time waiting at output link for transmission
 - depends on congestion level of router

Total Delay: Transmission + Propagation

$$dnodal = d_{proc} + d_{queue} + d_{trans} + d_{prop}$$

↓
Total delay

$$d_{trans} = \frac{L_{packet \ length \ bits}}{R \ link \ transmission \ rate \ bps}$$

$$d_{prop} = \frac{d}{S}$$

length of physical link
propagation speed
2 ns/m

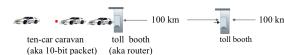
traffic intensity: $\frac{La}{R}$

= 0: avg queuing delay small
→ 1 avg queuing delay large
≥ 1 avg queuing delay infinite

Caravan analogy



- car ~ bit: car service ~ link
- toll booth takes 1 sec to service car (bit transmission time)
- "propagate" at 100 km/hr
- Q: How long until caravan is lined up before 2nd toll booth?
- time to "push" entire caravan through toll booth onto highway = $12 * 10 = 120$ sec
- time for last car to propagate from 1st to 2nd toll booth: $100km/(100km/hr) = 1$ hr
- A: 62 minutes



- suppose cars now "propagate" at 1000 km/hr
- and suppose toll booth now takes one min to service a car
- Q: Will cars arrive to 2nd booth before all cars serviced at first booth?
A: Yes! after 7 min, first car arrives at second booth; three cars still at first booth



Introduction

$$Pd = 30 \text{ ms}$$

Question #5: Consider the scenario shown in figure 1, with four different servers connected to four different clients over four three-hop paths. The four pairs share a common middle hop with a transmission capacity of $R = 100$ Mbps. The four links from the servers to the shared link have a transmission capacity of $RS = 20$ Mbps. Each of the four links from the shared middle link to a client has a transmission capacity of $RC = 90$ Mbps per second. You might want to review Figure 1 in the text before answering the following questions:

- a) What is the maximum achievable end-end throughput (in Mbps) for each of four client-to-server pairs, assuming that the middle link is fair-shared (i.e., divides its transmission rate equally among the four pairs)?

Answer:

The maximum achievable end-end throughput is 20 Mbps.

- b) Which link is the bottleneck link for each session?

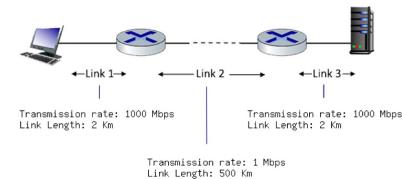
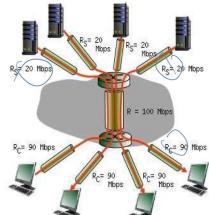
Answer:

This is the transmission capacity of the first hop, which is the bottleneck link, since the first-hop transmission capacity of 20 Mbps is less than one quarter of the shared-link transmission capacity ($100/4 = 25$ Mbps) and less than the third-hop transmission capacity of 90 Mbps.

- c) Assuming that the senders are sending at the maximum rate possible, what are the link utilizations for the sender links (RS), client links (RC), and the middle link (R)?

Answer:

The utilization of sender links is 100%. The utilization of receiver links is 22.22%. The utilization of the middle link is 80%.



Assume the length of a packet is 16000 bits. The speed of light propagation delay on each link is $3 \times 10^8 \text{ m/sec}$.
 $L = 16000 \text{ bits} \times 3 \times 10^8 \text{ m/sec}$

Answer the following questions:

- 1) What is the transmission delay of link 1?

$$R = 1000 \text{ Mbps} = 1000 \times 10^6 \text{ bps}$$

$$\text{Transmission Delay} = L/R = 16000/1000 \times 10^{-9} = 1.6 \times 10^{-5} \text{ seconds}$$

- 2) What is the propagation delay of link 1?

$$d = 2 \text{ Km} = 2000 \text{ m}$$

$$\text{Propagation delay} = d/s = 2000/3 \times 10^8 = 6.66 \times 10^{-6} \text{ seconds}$$

- 3) What is the total delay of link 1?

$$\text{Total Delay} = \text{Transmission} + \text{Propagation} = 1.6 \times 10^{-5} + 6.66 \times 10^{-6} = 2.267 \times 10^{-5} \text{ sec}$$

- 4) What is the transmission delay of link 2?

$$R = 1 \text{ Mbps} = 1 \times 10^6 \text{ bps}$$

$$\text{Transmission Delay} = L/R = 16000/1 \times 10^6 = 0.016 \text{ seconds}$$

- 5) What is the propagation delay of link 2?

$$d = 500 \text{ Km} = 500 \times 10^3 \text{ m}$$

$$\text{Propagation delay} = d/s = 500 \times 10^3 / 3 \times 10^8 = 1.667 \times 10^{-5} \text{ seconds}$$

- 6) What is the total delay of link 2?

$$\text{Total Delay} = \text{Transmission} + \text{Propagation} = 0.016 + 1.667 \times 10^{-5} = 0.017 \text{ seconds}$$

- 7) What is the transmission delay of link 3?

$$R = 1000 \text{ Mbps} = 1000 \times 10^6 \text{ bps}$$

$$\text{Transmission Delay} = L/R = 16000/1000 \times 10^{-9} = 1.6 \times 10^{-5} \text{ seconds}$$

- 8) What is the propagation delay of link 3?

$$d = 2000 \text{ m}$$

$$\text{Propagation delay} = d/s = 2000/3 \times 10^8 = 6.66 \times 10^{-6} \text{ seconds}$$

- 9) What is the total delay of link 3?

$$\text{Total Delay} = \text{Transmission} + \text{Propagation} = 1.6 \times 10^{-5} + 6.66 \times 10^{-6} = 2.267 \times 10^{-5} \text{ sec}$$

- 10) What is the total delay?

$$\text{Total Delay} = \text{Sum of all individual (total delays)} = 0.017 \text{ seconds}$$

Question #3: Suppose Host A wants to send a large file to Host B. The path from Host A to Host B has three links, of rates $R_1 = 1800$ kbps, $R_2 = 4.6$ Mbps, and $R_3 = 1.8$ Mbps.

- a) Assuming no other traffic in the network, what is the throughput for the file transfer?

Answer:

$$\text{Throughput} = \min(1800 \text{ kbps}, 4600 \text{ kbps}, 1800 \text{ kbps}) = 1800 \text{ kbps}$$

b) Suppose the file is 8 MB. Dividing the file size by the throughput, roughly how long will it take to transfer the file to Host B?

Answer:

$$[(8 \times 10^6 \text{ bytes}) / (1800 \times 10^3 \text{ kbps})] = 35.55 \text{ seconds}$$

Question #4: How long does it take a packet of length 50Kbytes to propagate over a link of distance 2,500 km, propagation speed 1.8×10^8 m/s, and transmission rate 200.5 Mbps? Recalculate for distance = 3000m.

Answer:

$$\text{Propagation delay} = d/s = (2500 \times 10^3 / 1.8 \times 10^8) = 13.8 \text{ msec}$$

$$\text{Propagation delay} = d/s = (3000 / 1.8 \times 10^8) = 16.6 \text{ usec}$$

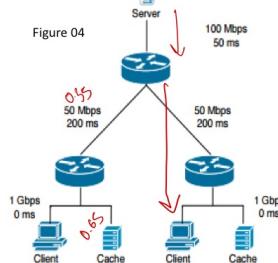
Part b) Consider the scenario as shown in Figure 04 in which a server is connected to a router by a 100Mbps link with a 50ms propagation delay. Initially this router is also connected to two routers, each over a 50Mbps link with a 200ms propagation delay. A 1 Gbps link connects a host and a cache (if present) to each of these routers and we assume that this link has 0 propagation delay. All packets in the network are 20,000 bits long.

i) What is the end-to-end delay when a packet is transmitted by the server to when it is received by the client? In this case, we assume there are no caches, there's no queuing delay at the routers, and the packet processing delays at routers and nodes are all 0?

ii) Here we assume that client hosts send requests for files directly to the server (caches are not used or off in this case). What is the maximum rate at which the server can deliver data to a single client if we assume no other clients are making requests?

iii) Again we assume only one active client but in this case the caches are on and behave like HTTP caches. A client's HTTP GET is always first directed to its local cache. 65% of the requests can be satisfied by the local cache. What is the average rate at which the client can receive data in this case?

Figure 04



Part c) Explain how a higher education institution, such as FAST-NU Karachi campus, can use private IP addresses for its 3000 devices as PCs and mobile devices and still manage internet connectivity using NAT and one public IP address. First, give a label diagram explaining how your setup NAT within a campus network of three labs, and office spaces for faculty, admin and accounts. Later explain, how an IP datagram from a private IP address are sent and received from the campus network to access bcc.co.uk website.

$$i) \frac{20000}{100 \times 10^6} + 50 \times 10^6 + \frac{20000}{50 \times 10^6} + 200 \times 10^6 + \frac{20000}{1 \times 10^9}$$

$$ii) 50$$

$$iii) 50 \times 10^6 \times 0.25 + 1 \times 10^9 \times 0.65 \\ = 667.5 \text{ Mbps}$$

Traceroute Program

↳ provides delay measures from source to router along end-end Internet Path towards destination

For all i

↳ sends 3 Packets to router i

↳ router i will return packets to sender

↳ sender measures time interval b/w transmission and delay



Real Internet delays and routes

traceroute: gaia.cs.umass.edu to www.eurecom.fr

```

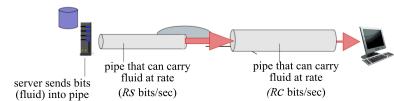
3 delay measurements from
gaia.cs.umass.edu to cs-gw.cs.umass.edu
1 cs-gw (128.119.240.254) 1 ms 1 ms 2 ms
2 border1-rt-fa5-1-0.gw.umass.edu (128.119.3.149) 1 ms 1 ms 2 ms
3 cht-vr1.gw.umass.edu (128.119.3.130) 6 ms 5 ms 5 ms
4 jn1-rt-fa5-1-0.wae.vbnns.net (204.147.136.136) 21 ms 18 ms 18 ms
5 jn1-so7-0-0-wae.vbnns.net (204.147.136.136) 21 ms 18 ms 18 ms
6 abilene-vbrn.abilene.ucsb.edu (198.32.11.91) 22 ms 18 ms 22 ms
7 abilene-rt-fa5-1-0.vbnns.net (204.147.136.136) 22 ms 22 ms 22 ms
8 62.40.103.253 (62.40.103.253) 104 ms 109 ms 106 ms
9 dg2-1.de.debian.net (62.40.103.129) 102 ms 104 ms
10 dg2-1.de.debian.net (62.40.103.129) 102 ms 104 ms
11 renater-gw1.fr.renater.net (62.40.103.54) 112 ms 114 ms 112 ms
12 195.220.98.102 (195.220.98.102) 123 ms 125 ms 124 ms
13 nice.csci.renater.fr (195.220.98.102) 123 ms 125 ms 124 ms
14 i32c-nice.csci.renater.fr (195.220.98.110) 126 ms 126 ms 124 ms
15 fantasie.eurecom.fr (193.55.113.142) 132 ms 128 ms 133 ms
16 194.214.211.25 (194.214.211.25) 126 ms 128 ms 126 ms
17 * * *
18 * * * means no response (probe lost, router not replying)
19 fantasie.eurecom.fr (193.55.113.142) 132 ms 128 ms 136 ms
  
```

looks like delays decrease! Why?

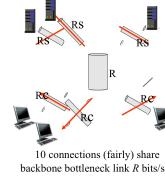
Throughput

↳ rate at which bits are being sent from sender to receiver

$$\min\left(\frac{R_c}{\text{no. of connections}}, R_s, R\right)$$



Throughput: network scenario



* Check out the online interactive exercises for more examples: [http://gaia.cs.umass.edu/traceroute](#)

instantaneous: rate at any given point in time

average: rate over long period of time

bottleneck link: link on end Path

that constrains end end throughput

Network Security

1. How bad guys attack computer networks
2. How to defend networks against attacks
3. How to design architech that are immune to attacks

BAD GUYS

Packet interception

- ↳ Packet sniffing
 - ↳ broadcast media e.g. wireless
 - ↳ shared ethernet
 - ↳ Promiscuous network interface
 - ↳ reads all packets e.g. Passwords

Fake Identity

- ↳ IP spoofing
 - ↳ injection of packet with false source address

Denial of Service (DoS)

- ↳ attackers make resources unavailable to legitimate traffic by overwhelming resource with bogus traffic



1. select target
2. break into hosts around the network (see botnet)
3. send packets to target from compromised hosts



Lines of defence

- **authentication:** proving you are who you say you are
 - cellular networks provides hardware identity via SIM card; no such hardware assist in traditional Internet
- **confidentiality:** via encryption
- **integrity checks:** digital signatures prevent/detect tampering
- **access restrictions:** password-protected VPNs
- **firewalls:** specialized “middleboxes” in access and core networks:
 - off-by-default: filter incoming packets to restrict senders, receivers, applications
- ... lots~~detecting/reacting to~~ DDoS attack Chapter 8

PROTOCOL LAYERS

Protocol “layers” and reference models

Networks are complex, with many “pieces”:

- hosts
- routers
- links of various media
- applications
- protocols
- hardware, software

Question: is there any hope of organizing structure of network? and/or our discussion of networks?

Why layering?

- dealing with complex systems:
- explicit structure allows identification, *identify, hierarchical relationship of complex system's pieces*
- layered reference model for discussion
- modularization eases maintenance, updating of system
 - change of implementation of layer's service transparent to rest of system
 - e.g., change in gate procedure doesn't affect rest of system
- layering considered harmful?

1. APPLICATION LAYER [M]

- ↳ supporting network applications
- e.g. HTTP, IMAP, SMTP, DNS

↳ exchanges msgs to implement some application service using services of transport layer

2. TRANSPORT LAYER [HT|M]

- ↳ process process data transfer
- e.g. TCP, UDP

Process to process delivery

↳ transfers msgs from one process to another using services of network layer

↳ encapsulates application layer msg **M** with transport layer header **HT** to create transport layer segment **[HT|M]**

3. NETWORK LAYER [Hn|HT|M]

- ↳ routing of datagrams host to host from source to destination delivery
- e.g. IP, routing protocols

↳ transfers Transport layer segment **[HT|M]** from one host to another, using layer links service

↳ encapsulates transport layer segment **[HT|M]** with network layer header **Hn** used by network layer to create network layer datagram **[Hn|HT|M]**

4. LINK LAYER [Hn|Hl|HT|M]

- ↳ data transfers b/w neighbouring network elements
- e.g. Ethernet, WiFi, PPP

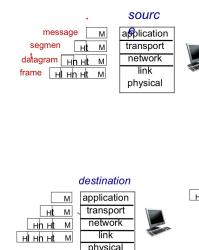
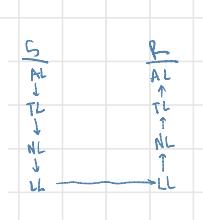
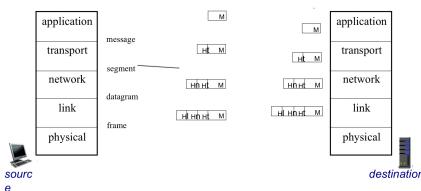
↳ transfers datagram **[Hn|HT|M]** from host to neighbouring hosts, using network layer services

↳ encapsulates network layer datagram **[Hn|HT|M]** with link layer header **Hl** to create link layer frame **[Hl|Hn|HT|M]**

5. PHYSICAL LAYER

- ↳ bits on the wire

Services, Layering and Encapsulation

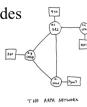


Encapsulation: an end-end view

INTERNET HISTORY

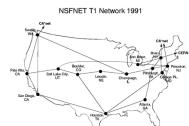
1961-1972: Early packet-switching principles

- 1961: Kleinrock - queueing theory shows effectiveness of packet-switching
- 1964: Baran - packet-switching in military nets
- 1967: ARPAnet conceived by Advanced Research Projects Agency
- 1969: first ARPAnet node operational
- 1972:
 - ARPAnet public demo
 - NCP (Network Control Protocol) first host-host protocol
 - first e-mail program
 - ARPAnet has 15 nodes



1980-1990: new protocols, a proliferation of networks

- 1983: deployment of TCP/IP
- 1982: smtp e-mail protocol defined
- 1983: DNS defined for name-to-ip-address translation
- 1985: ftp protocol defined
- 1988: TCP congestion control
- new national networks: CSnet, BITnet, NSFnet, Minitel
- 100,000 hosts connected to confederation of networks



2005-present: scale, SDN, mobility,

- cloud
- explosive deployment of broadband home access (10-100's Mbps)
- 2008: software-defined networking (SDN)
- increasing ubiquity of high-speed wireless access: 4G/5G, WiFi
- service providers (Google, FB, Microsoft) create their own networks
- bypass commercial Internet to connect "close" to end user, providing "instantaneous" access to social media, search, video content, ...
- enterprises run their services in "cloud" (e.g., Amazon Web Services, Microsoft Azure)
- rise of smartphones: more mobile than fixed devices on Internet (2017)
- ~18B devices attached to Internet (2017)

Introduction:

1972-1980: Internetworking, new and proprietary networks

- 1970: ALOHAnet satellite network in Hawaii
- 1974: Cerf and Kahn - architecture for interconnecting networks
- 1976: Ethernet at Xerox PARC
- late 70's: proprietary architectures: DECnet, SNA, XNA
- 1979: ARPAnet has 200 nodes

Cerf and Kahn's internetworking principles:
minimalism, autonomy - no internal changes required to interconnect networks
best-effort service model
stateless routing
decentralized control
define today's Internet architecture

Introduction: 1-1

1990, 2000s: commercialization, the Web, new applications

- early 1990s: ARPAnet decommissioned
- 1991: NSF lifts restrictions on commercial use of NSFnet (decommissioned, 1995)
- early 1990s: Web
 - hypertext [Bush 1945, Nelson 1960's]
 - HTML, HTTP: Berners-Lee
 - 1994: Mosaic, later Netscape
- late 1990s: commercialization of the Web
- late 1990s - 2000s:
 - more killer apps: instant messaging, P2P file sharing
 - network security to forefront
 - est. 50 million host, 100 million+ users
 - backbone links running at Gbps

Introduction:

Chapter 1: summary

We've covered a "ton" of material!

- Internet overview
- what's a protocol?
- network edge, access network, core
- packet-switching versus circuit-switching
- Internet structure
- performance: loss, delay, throughput
- layering, service models
- security
- history

You now have:

- context, overview, vocabulary, "feel" of networking
- more depth, detail, and fun to follow!

Introduction: 1-1

NETWORK APPS

- social networking
- Web
- text messaging
- e-mail
- multi-user network games
- streaming stored video (YouTube, Hulu, Netflix)
- P2P file sharing
- voice over IP (e.g., Skype)
- real-time video conferencing (e.g., Zoom)
- Internet search
- remote login
- ...

Q: your favorites?

CREATING NETWORK APPS

↳ write programs that

↳ run on diff end systems

↳ comm. over network

e.g. web server software comm. with browser software

↳ don't write softwares for network core devices

↳ as they do not run user application

↳ apps on end system allows for rapid app development, propagation



Client-Server Paradigm

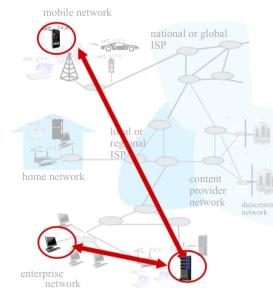
server

- ↳ always on host
- ↳ permanent IP address
- e.g. data center for scaling

Clients

- ↳ communicates with server
- ↳ may be intermittently connected
- ↳ may have dynamic IP address
- ↳ doesn't comm. directly with each other

e.g. HTTP, IMAP, FTP

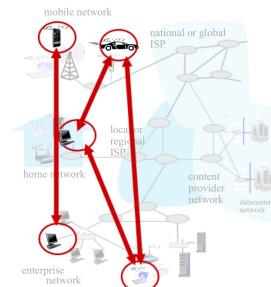


Peer Peer architecture

- ↳ no always on server
- ↳ arbitrary end systems directly comm.
- ↳ are intermittently connected
- ↳ can change IP address → complex management
- ↳ peers request service from other peers, provide service in return to other peers → self scalability

e.g. P2P file sharing

self scalability – new peers bring new service capacity, as well as new service demands



PROCESS COMMUNICATING

PROCESS

↳ a program running within a host

CLIENT PROCESS

↳ initiates communication

SERVER PROCESS

↳ waits to be contacted

INTER PROCESS COMMUNICATION

↳ used by two processes to comm. within the same host

↳ process in diff host communicate by exchanging messages

* applications with P2P architecture have client and server processes

SOCKETS

↳ process sends/receives messes to/from its sockets

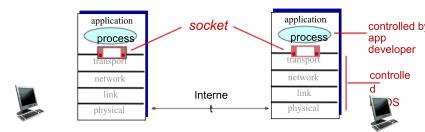
↳ sending process

↳ shoves messg out door

↳ it relies on transport infrastructure

↳ to deliver messg to socket at receiving process

two sockets involved : one on each side



ADDRESSING PROCESSES

identifier

↳ IP address + Port numbers

↳ every process must have an identifier *→ to receive msg*

- example port numbers:
 - HTTP server: 80
 - mail server: 25
- to send HTTP message to `gaia.cs.umass.edu` web server:
 - IP address: 128.119.245.12
 - port number: 80
- more shortly...

- *Q:* does IP address of host on which process runs suffice for identifying the process?
- *A:* no, *many* processes can be running on same host

Application Layer Protocol defines

1. Types of msg exchange

REQUEST
RESPONSE

2. MSG syntax

what fields in msgs
how fields are delineated

3. MSG Semantics

meaning of info in fields

4. Rules

when → Process sends and
now → Responds to msgs

5. Open Protocols

→ allows for interoperability e.g. HTTP, SMTP

6. Proprietary protocols

e.g. SKYPE ZOOM

What transport service does an app need?

data integrity

- some apps (e.g., file transfer, web transactions) require 100% reliable data transfer
- other apps (e.g., audio) can tolerate some loss

timing

- some apps (e.g., Internet telephony, interactive games) require low delay to be "effective"

throughput

- some apps (e.g., multimedia) require minimum amount of throughput to be "effective"
- other apps ("elastic apps") make use of whatever throughput they get

security

- encryption, data integrity, ...

Application

Transport service requirements: common apps

application	data loss	throughput	time sensitive?
file transfer/download	no loss	elastic	no
e-mail	no loss	elastic	no
Web documents	no loss	elastic	no
real-time audio/video	loss-tolerant	audio: 5Kbps-1Mbps video: 10Kbps-5Mbps	yes, 10's msec
streaming audio/video	loss-tolerant	same as above	yes, few secs
interactive games	loss-tolerant	Kbps+	yes, 10's msec
text messaging	no loss	elastic	yes and no

Application Layer

Internet Transport Protocols Services

TCP Service

Transmission Control Protocol

↳ reliable transport ↳ b/w Sending and receiving process

↳ flow control ↳ sender closes off overenough receiver

↳ congestion control ↳ remote sender when network overloaded

↳ connection oriented ↳ setup req b/w client and server processes

↳ does not provide

↳ timing

↳ min throughput guarantee

↳ security

UDP Service

User Datagram Protocol

↳ unreliable data transfer ↳ b/w sending/receiving process

↳ does not provide

↳ reliable transport

↳ flow control

↳ congestion control

↳ connection setup

↳ timing

↳ throughput guarantee

↳ security

why is there a UDP?

- no connection establishment (which can add delay)
- simple: no connection state at sender, receiver
- small header size
- no congestion control: UDP can blast away as fast as desired

25

why then?

↳ fast data transfer

↳ for video streaming, real time applications

Securing TCP

Vanilla TCP & UDP sockets:

- no encryption
- cleartext passwords sent into socket traverse Internet in cleartext (!)

Transport Layer Security (TLS)

- provides encrypted TCP connections
- data integrity
- end-point authentication

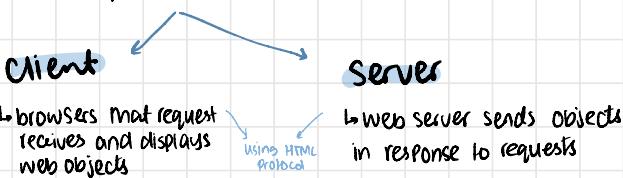
TSL implemented in application layer

- apps use TSL libraries, that use TCP in turn
- cleartext sent into "socket" traverse Internet *encrypted*
- more: Chapter 8

HYPER TEXT Transfer PROTOCOL (HTTP)

↳ web's application layer protocol

↳ client/server model



HTTP USES TCP

↳ Client initiates TCP connection to server ^{Creates socket} _{Port 80}

↳ Server accepts TCP connection from client

↳ Application layer protocol message

↳ HTTP msgs exchanged b/w browser and Web server

↳ TCP connection closed

WEBPAGE

↳ consists of

↳ Objects

↳ HTML file

↳ Images

↳ Audio file

↳ base HTML file

www.someschool.edu/someDept/pic.gif

HTTP IS STATELESS

↳ Server maintains no info about

Past clients requests

HTTP Connections

1. NON Persistent HTTP

- ↳ TCP connection opened
- ↳ 1 object sent to TCP connection
- ↳ TCP connection closed

CON

- ↳ downloading multiple objects require multiple connections
- ↳ requires 2 RTTs per object
- ↳ OS overhead for each TCP connection

2. Persistent HTTP

- ↳ TCP connection opened to server
 - ↳ multiple objects can be sent to TCP connection
 - ↳ TCP connection closed
- ↑ close over single TCP connection

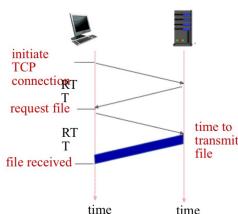
Persistent HTTP (HTTP1.1):

- server leaves connection open after sending response
- subsequent HTTP messages between same client/server sent over open connection
- client sends requests as soon as it encounters a referenced object
- as little as one RTT for all the referenced objects (cutting response time in half)

RESPONSE TIME

↳ 2RTT + file transmission time

- one RTT to initiate TCP connection
- one RTT for HTTP request and first few bytes of HTTP response to return
- object/file transmission time



RTT

- ↳ time for packet to travel from client to server and back

Non Persistent

$$= 2\text{RTT} \text{ (connection time)} + 2\text{RTT} \text{ (no of images)}$$

$$= 2(23.2) \quad 2(23.2)(1036) = 48116.8$$

Non-persistent HTTP: example

User enters URL: www.someSchool.edu/someDepartment/home/index (containing text, references to 10 jpeg images)

- 1a. HTTP client initiates TCP connection to HTTP server (process) at www.someSchool.edu on port 80
- 1b. HTTP server at host www.someSchool.edu waits for TCP connection at port 80 "accepts" connection, notifying client
2. HTTP client sends HTTP request message (containing URL) into TCP connection socket. Message indicates that client wants object `someDepartment/home/index`
3. HTTP server receives request message, forms response message containing requested object, and sends message into its socket
4. HTTP client receives response message containing html file, displays html. Parsing html file, finds 10 referenced jpeg objects
5. HTTP client receives response message containing html file, displays html. Parsing html file, finds 10 referenced jpeg objects
6. Steps 1-5 repeated for each of 10 jpeg objects

Persistent

$$2\text{RTT} \text{ (connection time)} + \text{RTT} \text{ (no of images)}$$



Question 3: Persistent and Non-Persistent HTTP

(10+10) Points

Suppose 1036 images need to be downloaded from the HTTP server. Assume $\text{RTT} = 23.2$ seconds, the total time taken to request and download 1036 images in a non-persistent and persistent connection is:

Always remember that in any case 2RTTs are a must-need to establish a connection.
For Non-Persistent Connections,

$$\begin{aligned}\text{Non-Persistent Time} &= 2\text{RTT}(\text{connection time}) + 2 * \text{RTT} * \# \text{ of images} \\ &= 2(23.2) + 2(1036) \\ &= 2(23.2) + 2(1036) \\ &= 2074.4 \text{ seconds}\end{aligned}$$

For Persistent Connections,

$$\begin{aligned}\text{We only need to establish a connection at the beginning and that's it.} \\ \text{Persistent Time} &= 2\text{RTT}(\text{connection time}) + \text{RTT} * \# \text{ of images} \\ \text{Persistent Time} &= 2\text{RTT} + 1036 \text{ RTT} = 1038 \text{ RTT} \\ \text{Persistent Time} &= 1038(23.2) = 24081.6 \text{ seconds}\end{aligned}$$

Non Persistent

$$\begin{aligned}&= 2\text{RTT}(\text{connection time}) + 2\text{RTT}(\# \text{ of images}) \\ &= 2(23.2) + 2(1036) = 48116.8\end{aligned}$$

Persistent

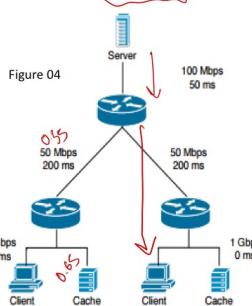
$$2\text{RTT}(\text{connection time}) + \text{RTT}(\# \text{ of images})$$

Part b) Consider the scenario as shown in Figure 04 in which a server is connected to a router by a 100Mbps link with 50ms propagation delay. Initially this router is also connected to two routers, each over a 50Mbps link with a 200ms propagation delay. A 1 Gbps link connects a host and a cache (if present) to each of these routers and we assume that this link has 0 propagation delay. All packets in the network are 20,000 bits long.

i) What is the end-to-end delay from when a packet is transmitted by the server to when it is received by the client? In this case, we assume there are no caches, there's no queuing delay at the routers, and the packet processing delays at routers and nodes are all 0?

ii) Here we assume that client hosts send requests for files directly to the server (caches are not used or off in this case). What is the maximum rate at which the server can deliver data to a single client if we assume no other clients are making requests?

iii) Again we assume only one active client but in this case the caches are on and behave like HTTP caches. A client's HTTP GET is always first directed to its local cache. 65% of the requests can be satisfied by the local cache. What is the average rate at which the client can receive data in this case?



Part c) Explain how a higher education institution, such as FAST-NU Karachi campus, can use private IP addresses for its 3000 devices as PCs and mobile devices and still manage internet connectivity using NAT and one public IP address. First, give a label diagram explaining how your setup NAT within a campus network of three labs, and office spaces for faculty, admin and accounts. Later explain, how an IP datagram from a private IP address are sent and received from the campus network to access bbc.co.uk website.

$$\text{i) } \frac{20000}{100 \times 10^6} + 50 \times 10^6 + \frac{20000}{50 \times 10^6} + 200 \times 10^6 + \frac{20000}{1 \times 10^6}$$

$$\text{ii) } 50$$

$$\text{iii) } 50 \times 10^6 \times 0.35 + 1 \times 10^9 \times 0.65$$

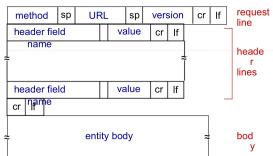
$$= 667.5 \text{ Mbps}$$

HTTP Messages TYPES

1. REQUEST MESSAGE

↳ ASCII → human readable format

HTTP request message: general format



2. RESPONSE MESSAGE

HTTP response message

status line (protocol → HTTP/1.1 200 OK
status code status phrase)

Other HTTP request messages

- POST method:
 - web page often includes form input
 - user input sent from client to server in entity body of HTTP POST request message

- HEAD method:
 - requests headers (only) that would be returned if specified URL were requested with an HTTP GET method.

PUT method:

- uploads new file (object) to server
- completely replaces file that exists at specified URL with content in entity body of POST HTTP request message

GET method (for sending data to server):

- include user data in URL field of HTTP GET request message (following a "?")
www.somesite.com/animalsearch?monkeys&banana

HTTP response status codes

- status code appears in 1st line in server-to-client response message.
- some sample codes:
 - 200 OK
 - request succeeded, requested object later in this message
 - 301 Moved Permanently
 - requested object moved, new location specified later in this message (in Location: field)
 - 400 Bad Request
 - request msg not understood by server
 - 404 Not Found
 - requested document not found on this server
 - 505 HTTP Version Not Supported

Trying out HTTP (client side) for yourself

1. netcat to your favorite Web server:

% nc -c -v gaia.cs.umass.edu 80

- opens TCP connection to port 80 (default HTTP server port) at gaia.cs.umass.edu.
- anything typed in will be sent to port 80 at gaia.cs.umass.edu

2. type in a GET HTTP request:

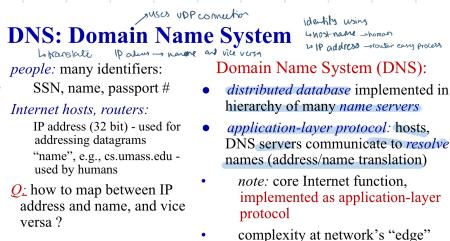
GET /kurose_ross/interactive/index.php HTTP/1.1
Host: gaia.cs.umass.edu

- by typing this in (hit carriage return twice), you send this minimal (but complete) GET request to HTTP server

3. look at response message sent by HTTP server!

(or use Wireshark to look at captured HTTP request/response)

↳ Web cache
↳ cookies
↳ missing



DNS: services, structure

DNS services:

- hostname-to-IP-address translation
- host aliasing
- canonical, alias names
- mail server aliasing
- load distribution
- replicated Web servers: many IP addresses correspond to one name

Q: Why not centralize DNS?

↳ single point of failure
 ↳ traffic volume
 ↳ distant centralized database
 ↳ maintenance

*↳ is at 2 particular locations
 ↳ high cost
 ↳ 1F for location*

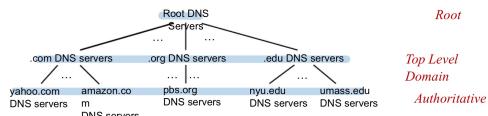
A: doesn't scale!

- Comcast DNS servers alone: 600B DNS queries/day
- Akamai DNS servers alone: 2.2T DNS queries/day



Application Layer: 2-59

DNS: a distributed, hierarchical database



Client wants IP address for www.amazon.com; 1st approximation:

- client queries root server to find .com DNS server
- client queries .com DNS server to get amazon.com DNS server
- client queries amazon.com DNS server to get IP address for www.amazon.com

Application Layer: 2-61

Local DNS name servers

- A:** when host makes DNS query, it is sent to its *local DNS server*
- B:** Local DNS server returns reply, answering:
- from its local cache of recent name-to-address translation pairs (possibly out of date!)
 - forwarding request into DNS hierarchy for resolution
 - each ISP has local DNS name server; to find yours:
 - MacOS: % scutil --dns
 - Windows: >ipconfig /all
- local DNS server doesn't strictly belong to hierarchy

(L DNS) → Root DNS

WWW FORWARD **TCP connection reuse**

B

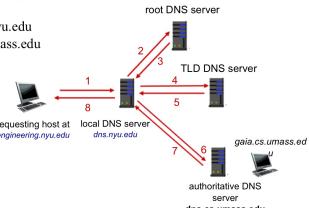
Application Layer: 2-65

DNS name resolution: iterated query

Example: host at engineering.nyu.edu wants IP address for gaia.cs.umass.edu

Iterated query:

- contacted server replies with name of server to contact
- "I don't know this name, but ask this server"

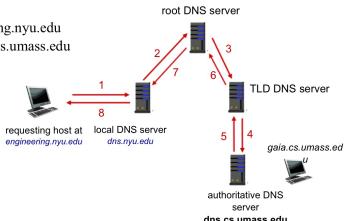


Application Layer: 2-61

Example: host at engineering.nyu.edu wants IP address for gaia.cs.umass.edu

Recursive query:

- puts burden of name resolution on contacted name server
- heavy load at upper levels of hierarchy?



Application Layer: 2-67

DNS security

DDoS attacks

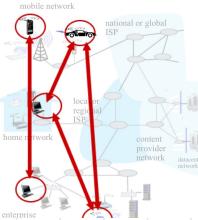
- bombard root servers with traffic
- not successful to date
- traffic filtering
- local DNS servers cache IPs of TLD servers, allowing root server bypass
- bombard TLD servers
- potentially more dangerous

Spoofing attacks

- intercept DNS queries, returning bogus replies
- DNS cache poisoning
- RFC 4033; DNSSEC authentication services

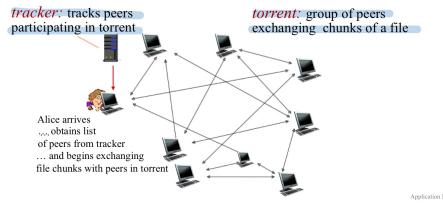
Peer-to-peer (P2P) architecture

- no always-on server
- arbitrary end systems directly communicate
- peers request service from other peers, provide service in return to other peers
- self scalability – new peers bring new service capacity, and new service demands
- peers are intermittently connected and change IP addresses
 - complex management
- examples: P2P file sharing (BitTorrent).



P2P file distribution: BitTorrent

- file divided into 256KB chunks
- peers in torrent send/receive file chunks



P2P file distribution: BitTorrent

- peer joining torrent:
 - has no chunks, but will accumulate them over time from other peers
 - registers with tracker to get list of peers, connects to subset of peers ("neighbors")
- while downloading, peer uploads chunks to other peers
- peer may change peers with whom it exchanges chunks
- **churn:** peers may come and go
- once peer has entire file, it may (selfishly) leave or (altruistically) remain in torrent



BitTorrent: requesting, sending file chunks

Requesting chunks:

- at any given time, different peers have different subsets of file chunks
- periodically, Alice asks each peer for list of chunks that they have
- Alice requests missing chunks from peers, rarest first

so if many leave from network
chunk still awl

Sending chunks: tit-for-tat

- Alice sends chunks to those four peers currently sending her chunks at highest rate
 - other peers are choked by Alice (do not receive chunks from her)
 - re-evaluate top 4 every 10 secs
- every 30 secs: randomly select another peer, starts sending chunks
 - "optimistically unchoke" this peer
 - newly chosen peer may join top 4

$$\text{Transmission time: } L \geq \frac{NF}{U_s}$$

↑
no of copies

$$\text{min download time} = \frac{F}{d_{\min}}$$

$$\text{Distribution time} = \max \left[\frac{NF}{U_s}, \frac{F}{d_{\min}} \right]$$

$$\text{Avg Req/sec} = 17$$

$$\text{Avg size} = 3.600 \text{ Mbits}$$

$$\text{Cache hit ratio} = 0.37$$

$$\text{Total Load} = \text{Avg Req/sec} \times \text{Avg size}$$

$$= 17 \times 3.6 \text{ Mbits}$$

$$= 61.2 \text{ Mbits}$$

Question 1: Please calculate the LAN and Access link utilization in following scenario.

- a. Average object size is 450 Kilo Bytes
- b. Average request rate from the browsers to origin server is 17 requests/seconds
- c. Cache Hit ratio is 0.37

$$\text{Average Requests/Sec} = 17$$

$$\text{Avg Size} = 450 \text{ KB} \times 8 = 3.600 \text{ Mbits}$$

$$\text{Total load} = (3.6 \times 17) = 61.2 \text{ Mbits/sec}$$

$$\text{Lan Utilization} = 61.2\%$$

$$\text{Cache hit ratio} = 0.37 \text{ which would put load of } 22.65 \text{ Mbits/sec}$$

to access link

So it will be 100% utilized and packets will drop due to queuing.

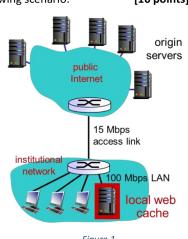


Figure 1

$$\text{LAN Utilization} = \text{Total load} \times \text{cache hit ratio}$$

$$= 61.2 \times .37$$

$$= 22.64 \text{ Mbits/sec}$$

ishma halfeez
notes

represent