



ملخص لمادة شبكات الحاسوب

للطالبة المبدعة
رهف ملحم

إرادة - ثقة - تغيير

ملاحظات :

1- هذا التلخيص يشمل

ch8 + ch9 + ch10 + ch13 + ch17 +

2- ch18 تم تلخيصه الـ Concept وتم ارفاقه بعض

الاصلة التي اخذت أثناء المحاضرة ولكن هذا الـ Chapter
يفضل دراسته من الكتاب والفيديوهات

+ ch19 + ch

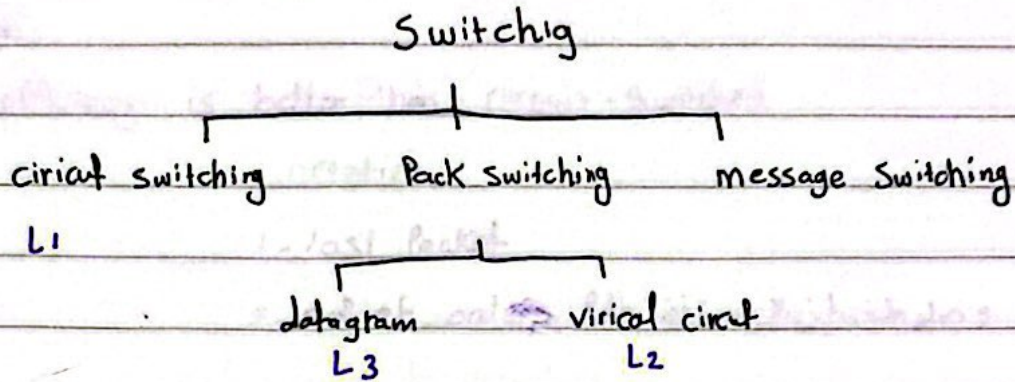
(تلخيص شامل لمادة شبكات حاسوب)

دكتور أحمد قحوم

لا تنسونا من صالح رعايتكم

ch 8: switching

Switched network: communicating devices with switch and we have multiple links



1- circuit switching

* by physical link L1

→ in the circuit, the resource need to be reserved by during Setup phase

Positive:

negative:

- | | |
|--|---------------------------|
| 1- fast transmission | 1- low efficiency |
| 2- in order delivery | 2- Lack of bandwidth (BW) |
| 3- minimal delay, Performance, Backlog | |

Ex: maybe we have a big sized in circuit switch in connection with computer

three setup:

- 1- setup → connect
- 2- data-transfer
- 3- Tear down → cancel or disconnected

* 2 type of delay here.

- 1- Transmission delay T_0
- 2- Propagation delay P_0

→ continuous because it's reserved all the time.

Packet switching, there is no resource reservation resource are allocated on demand

2- Data gram Switching:

* Packet ينقلها بطريقة أخرى عن الأخرى

→ in different path

effecincy: the efficiency is better than circuit-switched

Positive

negative

1- no wasted

1- lost Packet

2- out of order ← وهو ما يترتب عن كونها في مسارات مختلفة

Phase:

no setup and no teardown

one single Phase: data transfer

a switch in datagram use routing table that based on destination address

→ data transfer Packet transfer indival Pack no setup (تأخير) Delay:

- 1- Transmation delay
- 2- Processing delay
- 3- waiting
- 4- queueing delay.



destination	Port
1234	1
4150	2

we have 2 option:

- 1- reserved
- 2- on demand

datagram: indivsual Packet order

* in the pic in P 221:

on demand beacuse transmation line discrete

* Routing table > VCI address more complex

3- Vertkual-circuit network

mix between circuit and data-gram

circuit: setup + data transfer + tear down

* in the same path (تتبع المسار نفسه)

→

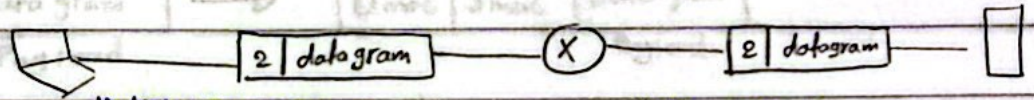
Ch 9: Data-link-layer.

nodes: end devices + connected devices (Sender + receiver + Router) → 6 devices

Links: network between connected devices (Point-to-Point) → 5 links



Server: data link layer is node-to-node



link: type one

link: type 2

link: \downarrow Protocol is 2 type يعني يكون 2

* IP end to end from source to destination

* mac Point-to-Point from client to router then from router to server

services:

1- Framing: A Packet at the data-link layer is normally called Framing

2- Flow control: sender + receiver في bit يعني تا يوصل في bit

* we cannot have an unlimited buffer-size at the receiving size we have 2 choice:

1 - drop the frame if it's buffer full

2 - the receiving data-link a feedback to stop or slow down.

3- Error control: Source address & destination address (detect + correct)

4- Congestion control: relate in wire lack of flow control (losses + drop)

data link layer → node-to-node
network layer → end-to-end

Two categories of links:

- 1- Point to Point (DLC)
- 2- broad cast (DLC+MAC)

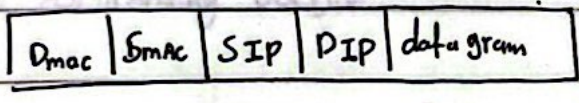
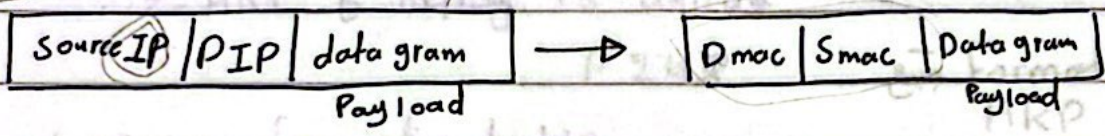
Two sublayers:

- 1- data link control (DLC)
- 2- media access control (MAC)

Link layer addressing

datagram

Frame



* in Page 243 in the book N: IP address L: MAC address

* IP → Logical address + Layer 3 network Layer

* MAC → Physical address + Layer 2 data link layer

خروجی سنج الیوم 9.5

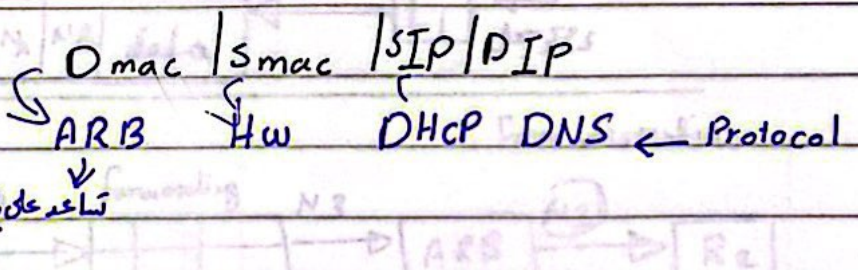
1- Application →

2- transport → Port number

3- network → Source + IP

4- data link → MAC

5- physical →



* Sender + receiver address router address

Three type of address:

1- unicast

2- multicast

3- broad cast

digit 2
even 0 → unicast
odd 1 → multicast
all ones → Broadcast.

Address Resolution Protocol (ARP) : network Layer

mac address \rightarrow I address \leftarrow

* ICMP: Internet control message Protocol.

* IGMP: Internet grouping management Protocol.

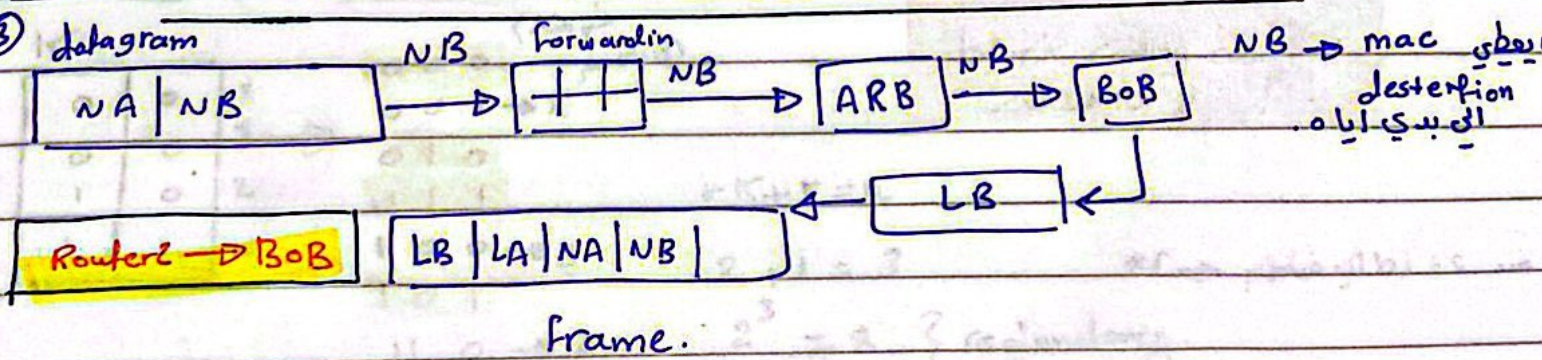
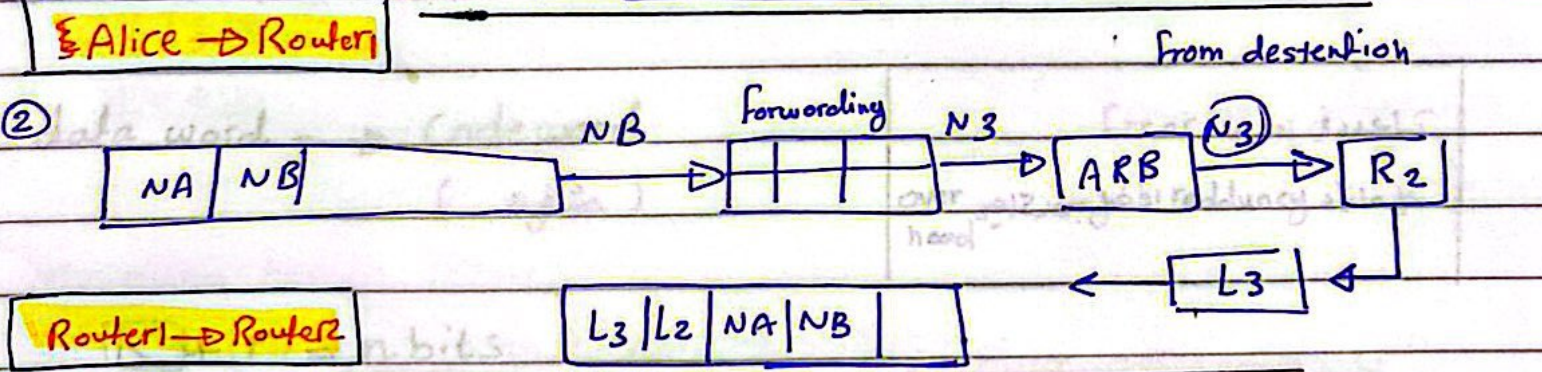
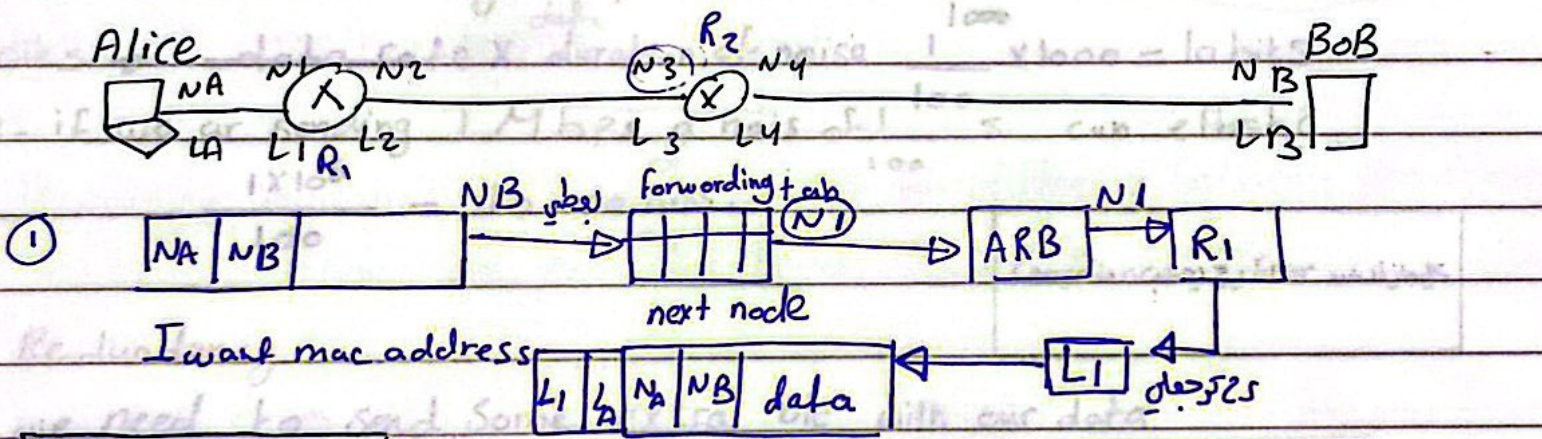
1- ARP \rightarrow Request is broadcast

2- ARP \rightarrow Reply is unicast

P 248

Format \rightarrow \times ARP

* we use forwarding table



Ch 10: Error detection and correction.

Type of error

Single-bit error

means that only 1 bit error

burst error.

means that 2 or more bit error

الاشياء التي يتغير اذالك Single or Burst.

*the number of bits effected depends on: ^① data rate and ^② duration of noise

1- ex: if we are sending 1 Kbps a noise of $\frac{1}{1000}$ secon (can effect?)

bits \rightarrow $\text{data rate} \times \text{duration of noise} = \frac{1}{1000} \times 1000 = 1 \text{ bit}$

2- if we ar sending 1 Mbps a nois of $\frac{1}{100}$ s can effect?

$$\frac{1 \times 10^6}{100} = 10,000 \text{ bits.}$$

Correction Error راجع

Redundancy:


we need to send some extra bit with our data

data word \rightarrow Codeword
(مشفرة)

Error راجع في
over head redundancy
المزيد من redundancy

$$|K| + |r| = n \text{ bits}$$

K			
0	0	x	000
0	0	y	001 \rightarrow x
1	0	z	010
1	1	w	011
			100 \rightarrow y
			101
			110 \rightarrow z
			111 \rightarrow w

block coding called
codeword # 

$$*K+r=n$$

$$2+1=3$$

$$2^3 = 8 \text{ } \left. \begin{array}{l} \\ \end{array} \right\} \text{ redundancy.}$$

$$2^2 \leq 2^3$$

*r \rightarrow عدد البتات الإضافية

*الشفرة 4 وبت 4

→ How to find the extra bits? data word (k) = 8bit, extr(r) = 2
 $1 - 8 + 2 = 10 \rightarrow 2^8 = 256 \quad 2^{10} = 1024$ then

linear Block coding:
 XORing any codeword
 with any codeword is a valid
 Codeword #.

$\begin{array}{r} 1024 \\ - 256 \\ \hline 768 \end{array}$	← illegal extra bit
--	---------------------

P260 لا اطلاع على مثال الكتاب 10.1
 ← ملائمة صيغة في الرفع الثالث

* an error-detecting code can detect only the types of error for which it is designed; other types of error may remain undetected.

لا يصحني Error في قيمة Bit لا يصح
 عليا اي احد Error

Hamming distance: it is the number of difference between the corresponding bits: → can be easily be found if apply XORing

ex:

$$\begin{array}{r} 0110 \\ \oplus \\ 1110 \\ \hline 1001 \end{array}$$

①00① hamming distance = 2

$$\begin{array}{r} 10101 \\ \oplus \\ 11110 \\ \hline 01011 \end{array}$$

01011 hamming distance = 3

minimum hamming distance:

is the smallest hamming between all possible pair of code word.

$d_{min} = S + 1$

S → garntess
 max # of error (إذا اوس مطر 4 valid code)

ex: 011 101 → $d_{min} = 2$ بطالع XORing الهم كلام

$2 = S + 1 \quad \boxed{S = 3}$ وبا حد أقل و 10 #

hamming distance = 0 إذا كان *
 Error بين 8 يوجد
 fall in line
 with
 hamming distance > 0 إذا كان *
 Error بين على وجود

Ch 13: Ethernet.

Servies: (Format + frame)

- 1 - min length
- 2 - feilds
- 3 - Size

IEEE 802: Instute of Electric and Elctronic Engeenir

1 - the relationship of the 802. Standrd TCP/IP and subdiviole

in 2 subplaires: Logical link control and meadia access control

2 - SFD: Start Frame (LLC) (MAC)

data link Layer →

Logical MAC و جود

Logical } flow error
consegition

LLC (framing)				Data link physical
Ethernet	Token Ring	Bus	Frang	
Ethernet	Token Ring	Bus		

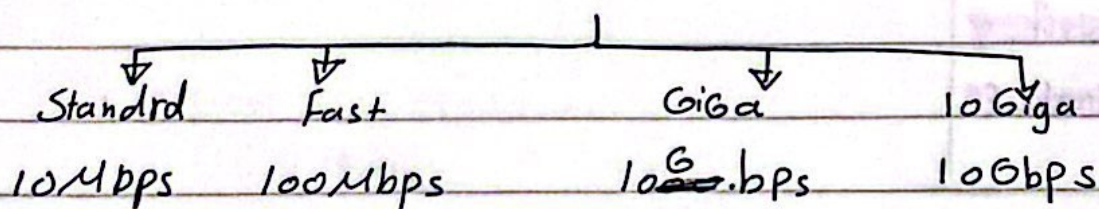
* با قلاق technology مكن يتقال

Length.

(media Access control) (MAC)

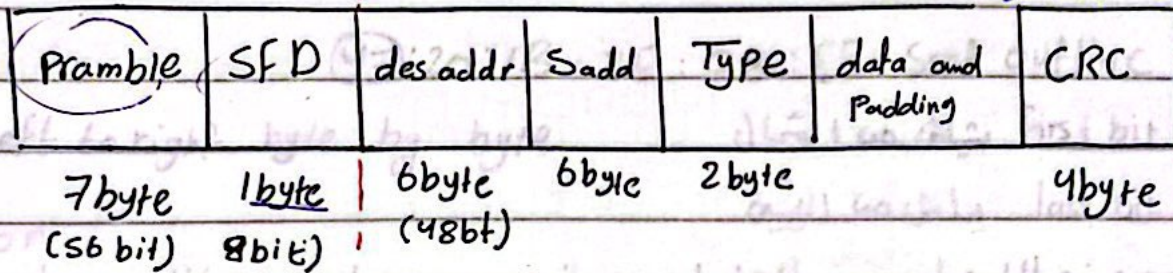
define CSMA/CD → Carrier sense multiple Access/collison detedion

Ethernet evolution



Ethernet frame

min frame length: 512 bit 64 byte $\rightarrow 14 + 46 + 4 = 64$ byte
 max frame length: 1518 byte $\rightarrow 14 + 1500 + 4 = 1518$ byte



1- Preamble: 56 bit alternating 0 and 1 **Header**
 (Physical layer) Frame **منها جزئ**

2- SFD: Start frame delimiter (10101011) Frame **بها اشارة الوارد مع رقم من بيان**

3- Type: ARB or IP Type of Protocol upper layer:

4- data and padding (Pay load) **لن يكون في Range**
 min pay load length: 46 byte **سكان يكون حقا حرة مضافة + sense**
 max pay load length: 1500 byte **تبعه انوني Level of signal energy**
 \rightarrow Packet **بها** by CSMA/CD

5- CRC: the last field contains error detection information.
 \rightarrow (4 byte)

Format:
1- total length
2- fields
3- length each field

Addressing: The Ethernet is 6 byte (48 bit)

ex: 47:20:1B:2E:08:EE sent outline

left to right byte by byte

إذا بدأنا من اليمين first bit

إذا بدأنا من اليمين last bit

mirror

0100 0111 : 0010 0000 : 0001 1011 : 0010 1101 : 0000 1000
1110 0010 : 0000 0000 : 1101 1000 : 0111 0100 : 0001 0000

1) Standard Ethernet: (bus + hub) topology

* we need to use an access method to control sharing medium

CSMA/CD

implementation: UTP or fiber optic

* 10 Base 5: we use Trasiver (kind of cost) (thick coaxial)

Signal attenuation من اتجيب (repeater) في كل 500 م

* 10 Base 2: Tconnector (flexible + short distance + cost ↓)

long distance more repeater - then I will use 10 Base 2

* 10 Base T: (star topology)

Hub → collision

CSMA/CD

من اتجيب

Star → No collision

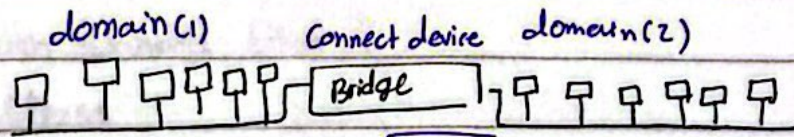
* 10 Base F: 10Base collision تجاوزنا

full duplex

في

Changes in Standard

① Bridge Ethernet



Logical devices = 6 + 7 = 13 \times $\frac{10}{7}$ Mbps (العنقزة من الـ 10) \times
 1 - increase Bw 2 - Separation domain for collision

② Switched network (Ethernet) half-duplex

Switch = N-Port Bridge

of -the node in network

* (more complex increase Logical Bandwidth increase the collision domain)

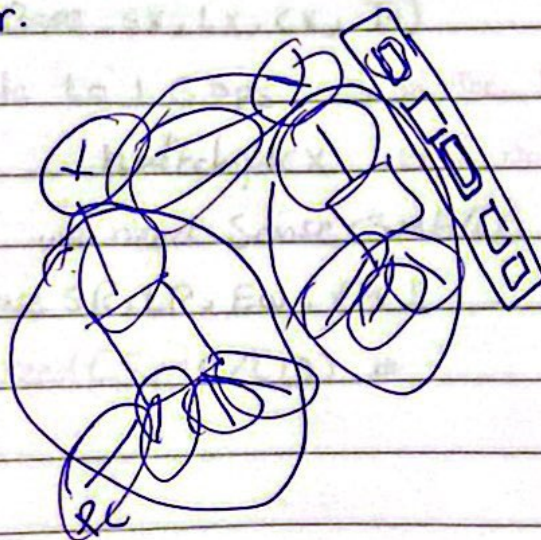
③ Full-duplex Switched network.

no need CSMA/CD we have transmit + receive

* MAC Sublayer:

designed as a connectionless. There is no explicit flow control or error control to inform the sender that the frame has not been sent to destination without error.

coll



[2] Fast Ethernet

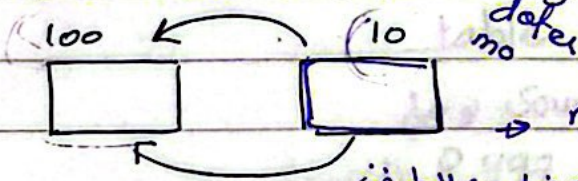
- 1- upgrade the data rate 100 Mbps
- 2- make it compatible with standard Ethernet
- 3- Keep same 48 bit address
- 4- Keep the same frame format.

* CSMA/CD depends on the transmission rate, the minimum size of the frame and maximum network length.

* A new feature added to Fast Ethernet is called **Autonegotiation**. it allows a station or a pair range of capabilities

half-duplex *
CSMA/CD

it allow two device to negotiate the mode or data rate of operation.



وهو ما يسمى 10 ميجابتا
بينما يتم تفاوض ال Rate مع كل من الجانبين
في حال بقا 100 Mbps وكان Receiver - فاله 10M

* Point-to-Point → Star topology. (hub or switch) #

[3] Giga Ethernet (1000Base-SX, LX, CX, T4)

- | | |
|-------------------------------------|---|
| [1] upgrade the data rate to 1 Gbps | in the full duplex mode of * |
| Full duplex | half duplex |
| no need Sens | Giga no collision, the maximum |
| | we need Sense CSMA/CD length of the cable deter |

[4] Giga Ethernet (10GBase-SR, LR, ER, X4) by the signal attenuation.

Just full duplex no need CSMA/CD) #

Ch17: connecting devices and virtual LANS. the same tool to all

hubs (Layer 1 Physical Layer)

- 1 - Connection.
- 2 - Amplify this signal.
- 3 - Look if it has attenuation.

Switch (L1 + L2 data link layer)

- 1 - has table used in filtering (Learning table)
- 2 - we can put the information in this table (manual or Automatic)
- 3 - does not change the link layer (mac) address in a frame

- * الجواز لي بيت مابكون على table
* الجدول بيتي من كل Source mac بوجه
* Learning Switch P 497 Fig (17.4)

address	Port
71:28:	1

Example في اجمع عن Loop problem in 2 switch بسبب وجود a Learning Switch

حل المشكلة هاي عن طريق حذف امد ال Switch
* P 498 Fig 17.5.

System of connected LANS and it's graph representation

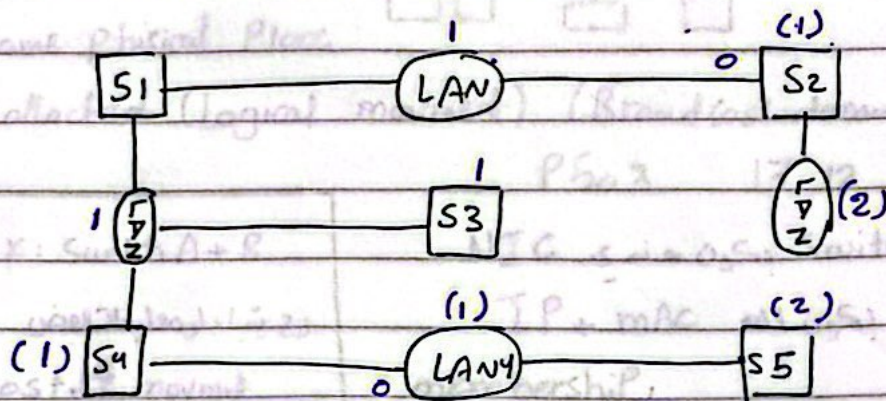
in Fig 17.6 P 499 we have 5 Switches and we have redundant path

- 1 - Spanning tree one single path: حل المشكلة عن طريق
- 2 - collection of Path.

Spanning tree: only one single path from the same root to all possible destination

* **Least Cost tree:** Path بحدرافض
 # of node + destination + traffic

- * Switch → LAN → (1)
- * LAN → Switch → (0)



3 - Router (L1, L2, L3 network layer)

1 - Amp + mac + IP address

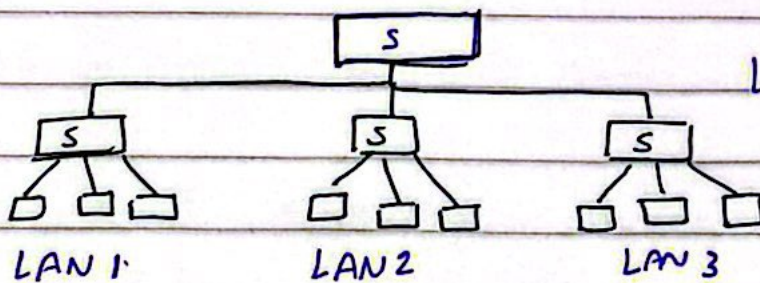
we used routing table

router change the link layer address

connected over the internet

a switch connecting three LANs

* أنا ما بين ال Path
 Blocking وال Path ال بي
 Forwarding اياه بيلو



مثلا اذا احتجنا جهاز مثلا على LAN2

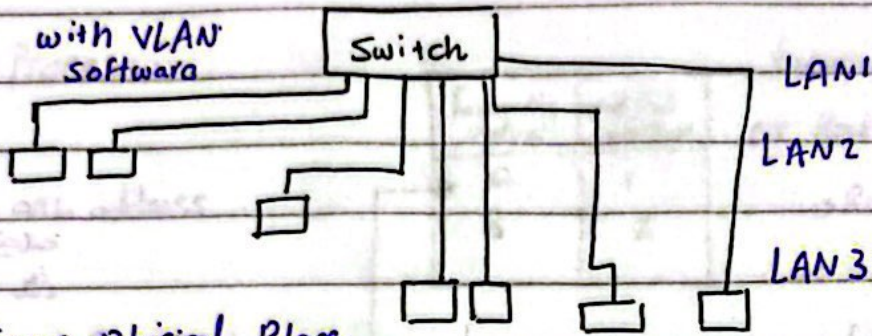
محتاج يصير عندي Physical member

وهذا ده شي مش عملي ممكن

يصير عندي Firewall

← في عندي حل

Switch using VLAN : virtual Local Area network



not in the same physical place

I can do reallocated (logical movement) (Broadcast domain)

17.12 * في قباله

<p>Tax: Switch A+B (2) يبا دلوه معلومات بعض</p>
<p>Cost ↓ ↑ movment</p>
<p>time ↓</p>
<p>more Privacy</p>

Switch يكون عندي NIC ←

IP + MAC يكون ال ←

membership,

interface, Port number, MAC, IP, IPmulticast

*configuration:

1- manual: -the administrator uses VLAN

2- Automatic: Connected or disconnect

VLAN using criteria defined by administral

3- semiautomatic: intilze manual.

migrations automatically.

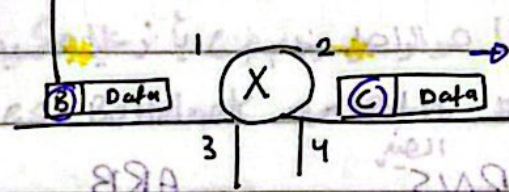
Ch 18: Network Layer

Forwarding Process:

Forwarding value	output interface
A	1
B	2

forwarding table
or Routing table

→ Routing Protocol
we can build table.



→ forwarding

الترتيب المطلوب

* Routing address صلاحية
(VLS) غير

* VCI اذا كان اي متخدم

B and C ممكن يكون

* IP اذا كان متخدم

Just-B لازم يكون

Other Services:

* Lack of flow control → Error control + congestion control.

1- Flow Control: regulates the amount of data a source can send without overwhelming (ينظم كمية التدفق دون إرباك الـ destination)

2- Congestion: too many datagram are present in an area of the internet - maybe send duplicates of the lost Packet

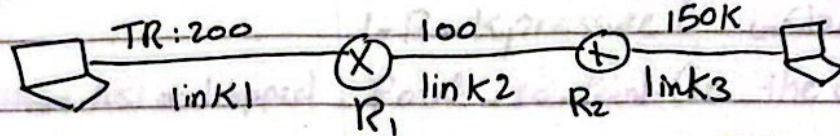
3- Quality of Service: real time communication

4- Security: To Provide Security for connection less we need to have another virtual level to protect connection oriented this virtual layer call IPsec

network layer Performance

1 - Throughput: the number of bits passing through the point in a second

2 - delay



عندي هون ال link بتوعب 200 بوال link بتوعب 100 فالتالي 150
 برعبانو link هو الواصل بينهم فرح يا ترعيا بحسبة ال Bit ال بوجرة

→ simulation using pipes: Bottleneck (negative) ال فير



- Packet loss

- more retransmission

2 - delay

* Congestion control: improve performance

- Transmission delay: $T_r = \frac{\text{message size or Packet length (Sender} \rightarrow \text{router)}}{\text{transmission rate}}$

- Propagation delay: $P_r = \frac{\text{distance (on Link)}}{\text{Prop Speed}}$ Prop Link on

- Processing delay = P_r = Time required to process Packet in Route or destination h or desion

- Queueing delay = The time a packet waits in input and output queue in Router

$$\text{Total delay} = (n+1)(T_r + Prop + Proc) + (n)(Queue)$$

(n+1)

S.

Congestion Control (throupt)

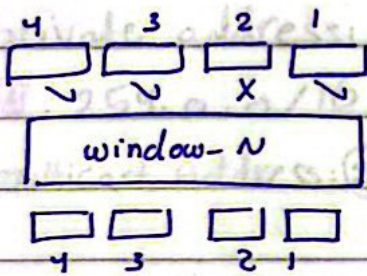
Open loop (Prevent) ← تجنب

1- retransmission:

if the packet feel there lost or droped

2- window Policy:

* go - Back - N



* يعنيهم كلهم مع الـ 2 في مشكلة
بها في الحالة تتفتم
المشكلة

* Selective - retransmission: أخرى انتقائية يكون محدود مكان فيه

3- Acknowledgment: إذا ما وصل (Pas Ack) بسبب

او صارت في Timeout (تو) Sender & receiver

اذا وصلتك فلاك (SS) لازم ارجع بالوقت المتبقي ACK

Packet field

4- discarding: ~~server~~ different level of server

تبادل packet الى الواجهة قبل وجعل

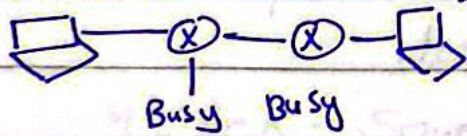
discarding

5- admission:

* V.C.S ← من الواجهة ما عنى قدرة

- set up

deny because



I don't have any space.

Close loop (to remove)

1- Backpressure (توصيل في الاتجاه العكس)

Point-to-Point (in the opposite direction)

لعمل Block على Path معين

بس لما يكون Path معروف (virtual-circuit)

2- Chock - Packet:

Send by node to source directly
ex (ICMP)

3- Implicit Signaling:

The delay in receiving an Acknowledgment

the source to slow down

4- Explicit Signal:

the Packet that carry signal

Signaling can occur in either

forward or Backward direction

ex: ATM.

Private:

- 10.0.0.0 / 8
- 172.16.0.0 / 12
- 192.168.0.0 / 16
- 169.254.0.0 / 16

Special address:

- 1- This host address: 0.0.0.0 / 32 * 10.0.0.0 / 8
* 172.16.0.0 / 12
- 2- limited broadcast: 255.255.255.255 / 32 * 192.168.0.0 / 16
* 169.254.0.0 / 16
- 3- loopback Address: 127.0.0.0 (8)
- 4- private address: 10.0.0.0 / 8, 172.16.0.0 / 12, 192.168.0.0 / 16, 169.254.0.0 / 16
- 5- multicast Address: 224.0.0.0 (4)

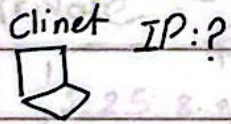
DHCP: Dynamic Host Configuration Protocol (الخدمة التي توفر عناوين الـ IP)

* Application layer Solve the problem IP from static to dynamic

→ broadcast limit connection less

~~10.0.0.0 / 32~~

1- subnetmask, IP, gateway, DNS,



→ Server IP: 181.14.16.170

this host

1- DHCP discover → Source address ~~to broadcast~~ limit
 destination address → Broadcast (0.0.0.0 / 32)
 (255.255.255.255 / 32)

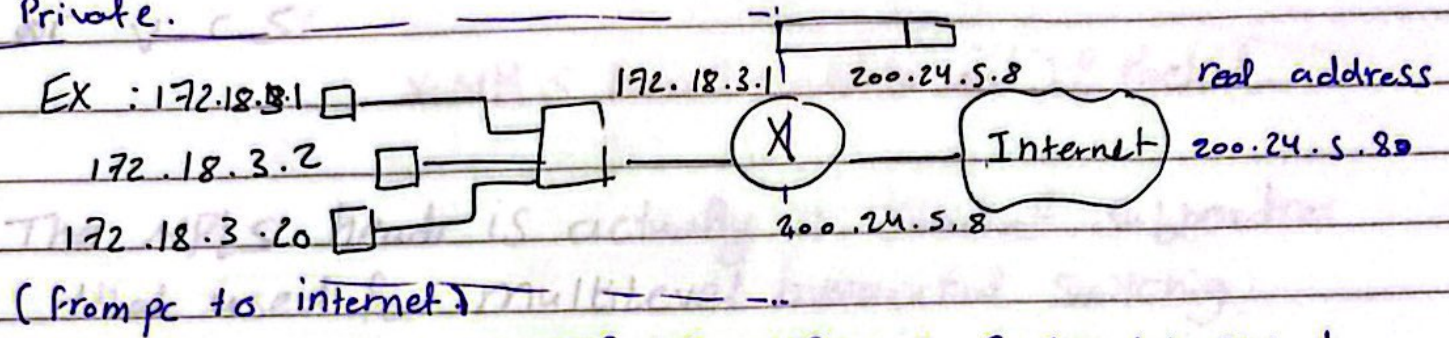
2- DHCP offer: source address (server) → 181.14.16.170 / 32
 destination address → 255.255.255.255 / 32

3- Request: Source address → 181.14.16.182 (Client) (تتالي)
 destination → 255.255.255.255

4- ACK: Source address → server. 181.14.16.170
 destination → 255.255.255.255

NAT: Private address (التقليد في الذاكرة) IP depletion
 كل الجامعة الواجبة يكون لها 2 IP (real)

Private.

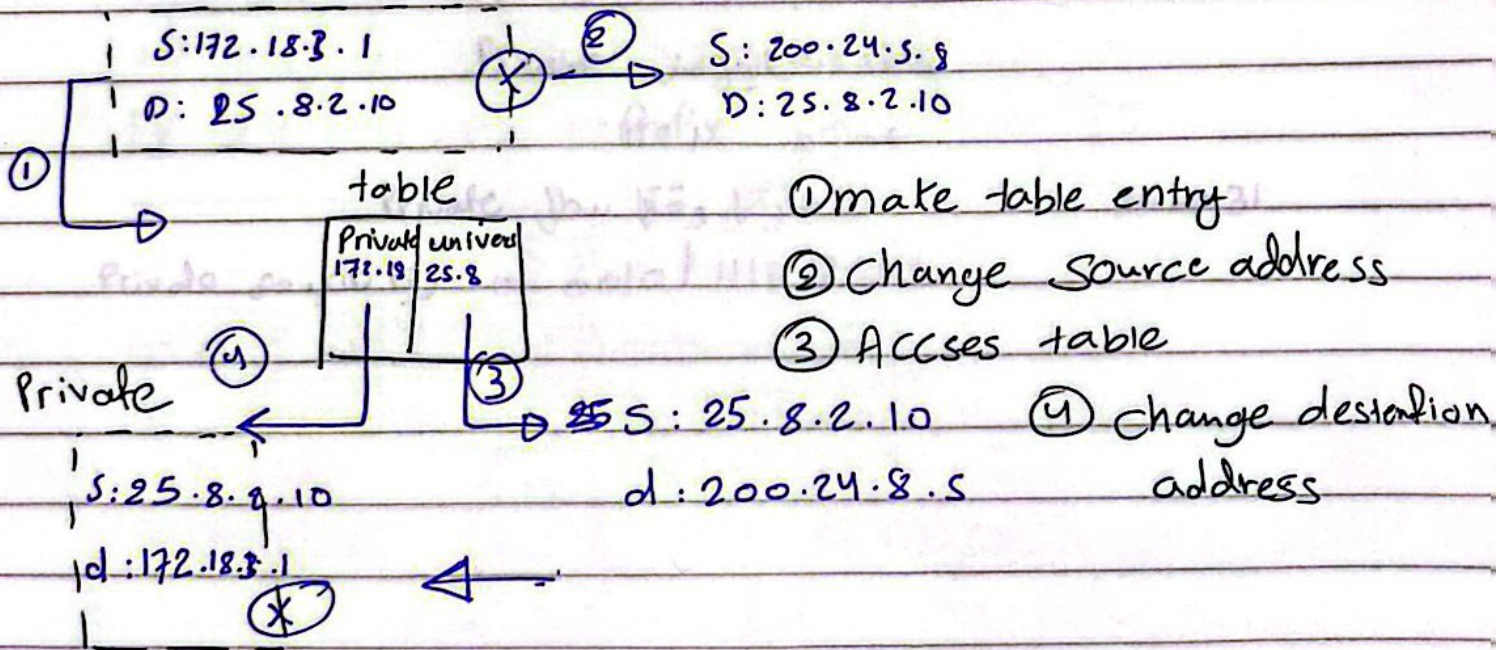


1- كل اتصال Router (2) يتبدل Private IP -> real IP

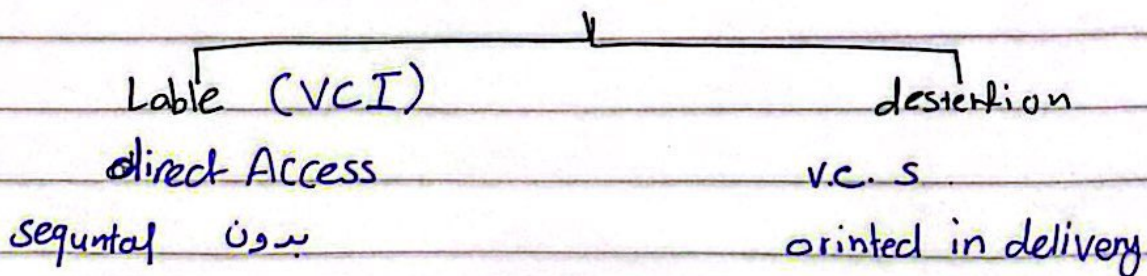
3- لا يصير Replay على real IP على Router Private IP

4- كل ما يكون عندي NAT table

Private



forwarding Based



Router as Packet Switches

* Router can be configured to act as either data gram or v.c.s

* MPLS header add to an IP Packet.

The MPLS header is actually a stack of subheaders that used for multilevel hierarichal switchig.

14.24.74 Private / 24

172.17 172.0001 | 0000.0.0

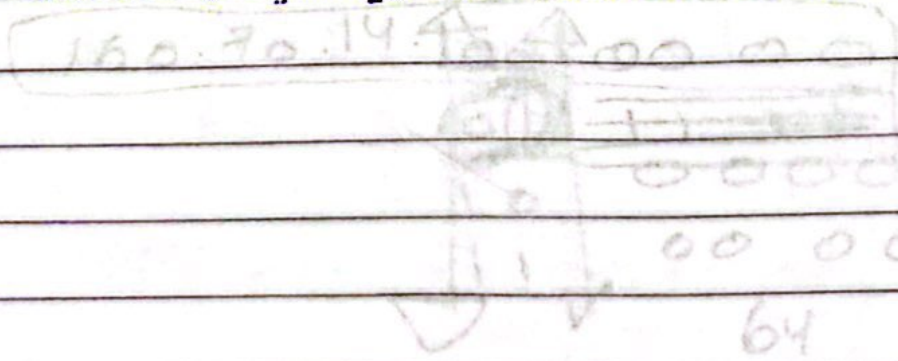
14.24.74 172.0001 | 1111.255

14.24.74 Private / 25

14.24.74.00 12 Prefix

Private ليا هارجل وقا بطل 16-31

Private زي هاي صر 0010 | 1111.2555



Class less

Class A 0 [000 0000 - n=8bits (0-127)] Class D 11110
 [111 1111] Class E 11111

Class B 10 [00 0000 n=16 (128-191)]
 [11 1111]

Class C 110 [0 0000 n=24 (192-223)]
 [1 1111]

→ Subnetting (Prefix)
 suberneting (Suffix)

* we will learn:

- ① Total number of address in the network [network space]
- ② The first address in the network (network Id)
- ③ Last address

230.8.16.132 / 16 ← Prefix Length
 # of network

① network space → $2^{32-16} = 2^{16}$ address
 # of address

② First address → 230.8.0.0 / 16

③ Last address → 230.8.255.255 / 16

230.8.16.132 / 26

① network space = $2^{32-26} = 64$ address
 # of address

② First address = 230.8.16.10000100 (network Id)

230.8.16.128 / 26

③ Last address = 230.8.16.191 / 26

23.30.67.92 / 12 → 23. [000] 1110.

① network space: $2^{32-12} = 2^{20}$

② First address: 23.0001 0000.0.0 / 12 → 23.16.0.0 / 12

Last address: 23.0001 1111.255.255 / 12 → 23.31.255.255 / 12

Ex 18.1: 167.199.170.82 / 27 $82 = 01010010$

of address $\rightarrow 2^{32-27} = 32$ address

First address = 167.199.170.01000000 \rightarrow 167.199.170.64 / 27
network Id

Last address = 167.199.170.01011111 \rightarrow 167.199.170.95 / 27

mask address

Prefix bits
all ones
net

Suffix
all zero
Host

167.199.170.82 / 27

255.255.255.11110000 mask

255.255.255.224 (not mask)

mask
(not mask)

\rightarrow 255.255.255.00011111
01010010
01011111

\rightarrow First address [network Id] Anding \rightarrow 167.199.170.01000000
167.199.170.64

\rightarrow Last address [host] \rightarrow 167.199.170.01011111
167.199.170.95

Address space \rightarrow NOT(mask) + 1 = 32 address

Example: 230.8.24.56 / 16

- ⑩ 230.8.0.0 \rightarrow 230.8.25.255 host ايس في Suffix *
Last address
- ⑫ 230.8.16.0 \rightarrow 230.8.31.255 ايس في Zero *
network Id.
- ⑭ 230.8.24.0 \rightarrow 230.8.31.255

1000 address $\rightarrow \log_2 1000 \approx 10$ 1024 address

Prefix length $\rightarrow 32 - 10 = 22$

18.14.12.0 / 22 \leftarrow big #

Ex: 18.5

14.24.74.0/24

Space address = $2^{32-24} = 2^8 = 256$

Block $\left\{ \begin{array}{l} 10 \rightarrow 16 \\ 60 \rightarrow 64 \\ 120 \rightarrow 128 \end{array} \right.$ * we want start with biggest on

① Find the new prefix length (Prefix the sub-block)

1 - Subnet 120-128

$32 - \log_2 128 = 25$

First: 14.24.74.0/25

Last: 14.24.74.127/25

2 - Subnet 60-64

new Prefix = $32 - \log_2 64 = 26$

14.24.74.128/26 ←

First: 14.24.74.128/26 (لاست على)

Last: 14.24.74.191/26

3 - Subnet 10-16

new prefix: $32 - \log_2 16 = 28$

First 14.24.74.192/28

Last 14.24.74.207/28

Example of address aggregation

Block 1

160.70.14.255/24

unequal subBlock

$\log_2 4 = 2$

Prefix

الرقم

$24 + 2 = 26$ ← new prefix

0 160.70.14.0/26

First: 160.70.14.0/26

Last: 160.70.14.63/26 ←

1 160.70.14.64/26

160.70.14.64/26 (first)

Block 4

3 160.70.14.192/26 (first)

160.70.14.127/26 (last)

160.70.14.255/26 (last)

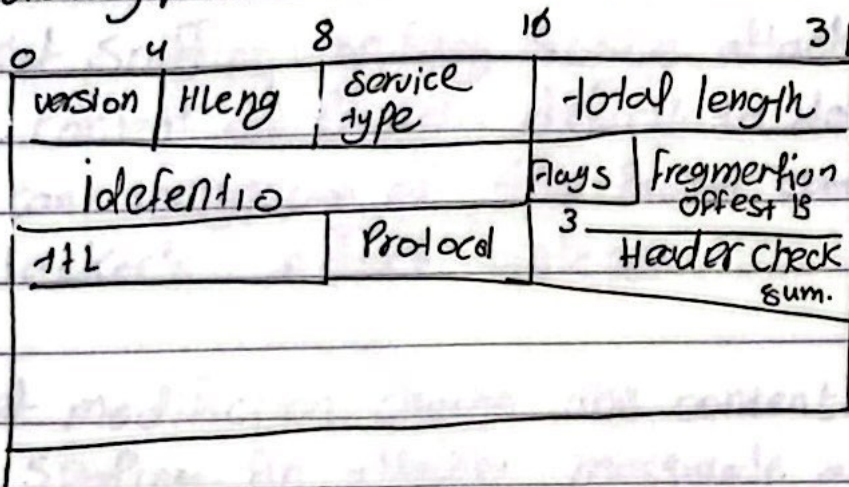
2 160.70.14.128/26

160.70.14.128/26 (first)

160.70.14.191/26 (last)

Ch 19: network Layer Protocol.

IP datagram



→ ICMP 01 Ex:

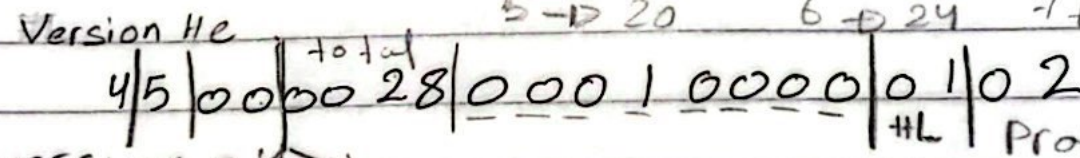
IGmp 02

TCP 06

UDP 17

OSPF 89

- 1 → 4 3 → 12
- 2 → 8 4 → 16
- 5 → 20 6 → 24 7 → 28 8 → 32



version = 4

Header length = $5 \times 4 = 20$

How many hops can
can this packet travel before
* → being dropped?

- D → 1 not allowed 0 → Fragment allowed
- M → 1 NOT Last Packet 0 → last Packet

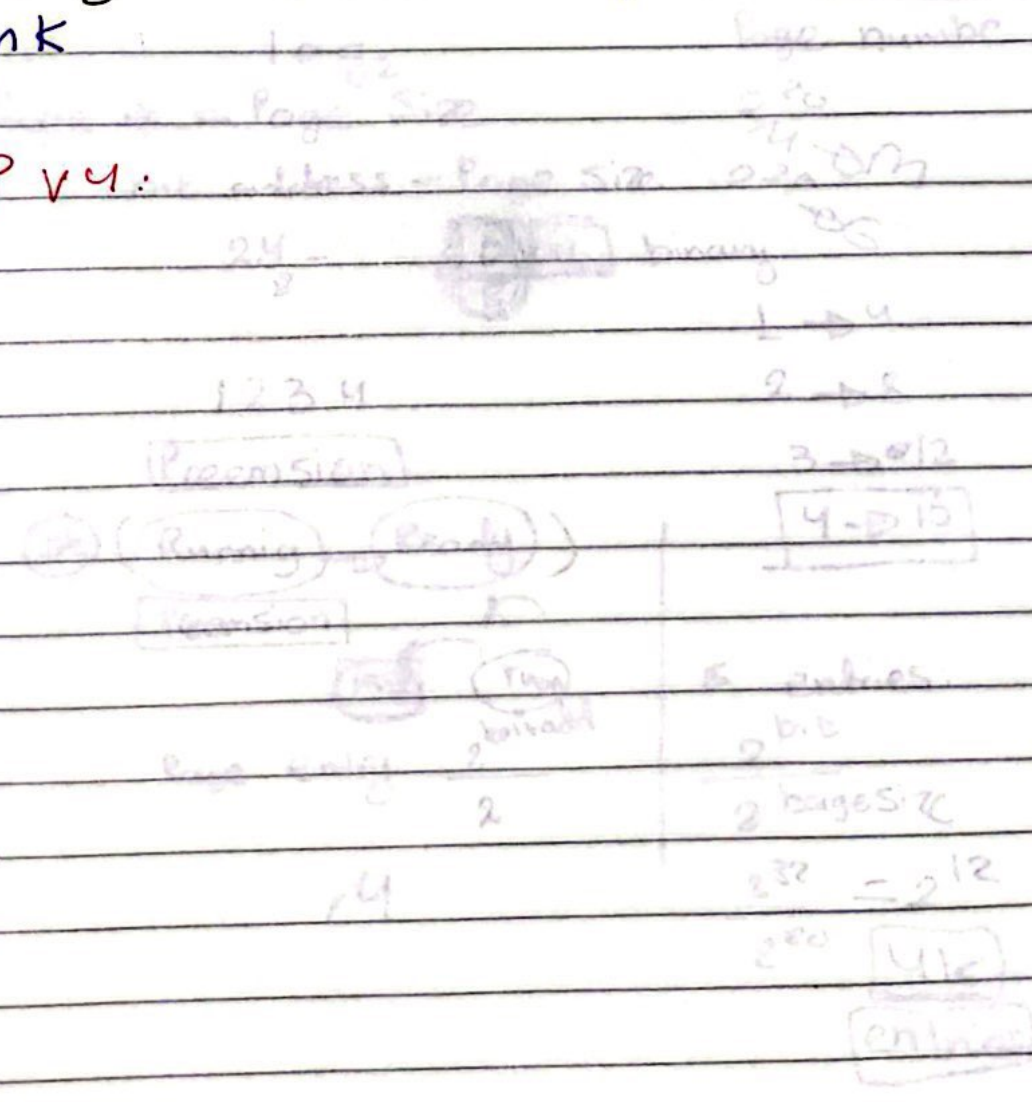
Security Sniffing

① Packet Sniffing: sniffing passive attack doesn't change the content of packet. default to detect we can encryption of the packet can make the attacker's effort unless

② Packet modification: change the contents. attac can detected.

③ IP Spoofing: An attacker masquerade as somebody bank

ICMP v4:



Ch 20

Unicast Routing

Routing Algorithm

1 - distance vector: is the rationale for the name distance vector routing. A least cost tree is a combination of least-cost path. each node send what it know about the whole internet to it's neighbour.

we use Bellman-Ford equation: to find shortest distance between source

$$D_{xy} = \min \{ d_{xy}, C_{xz} + D_{zy} \}$$

But we have a problem: Count to infinity \rightarrow Tow node instability

* we have two solution:

① Split Horizon: Entry فاضية في المرسلة اذا تجاوزنا timer \rightarrow unreachable

② Poison Reverse: Entry لا بعث vector A \rightarrow طر فاضية ∞

2 - link state Routing: it is send as A information to all whole internet (flooding) \rightarrow Link State Data Base (LSDB)

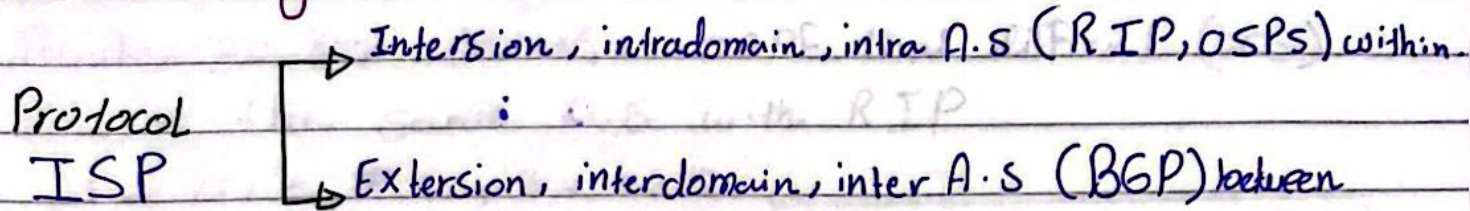
we use Dijkstra Algorithm \rightarrow Shortest Path

3 - Path Vector Routing: Best Path with Policy.

Each node send what it know about the whole internet

to it's neighbour. $x [] = \text{Best} [x (), x + y []]$

unicast Routing Protocols:



Autonomous System: ISP \rightarrow A.S \rightarrow self management

we cannot use single Protocol: administrative, Scalability.

- 1- Stub A.S: has only one connection to another AS (Customer network)
- 2- Multihomed A.S: can have more than one connection to other AS. but doesn't allow data to traffic to through it (Provider networks)
- 3- Transient AS: connected to more than one other AS also allows the traffic to pass through it (Backbone).

□ RIP Protocol: Routing Information Protocol

hop count = # of Router from Source Router to host destination. depend forwarding table.

implemented: use UDP L4 demon Process (Running in background). (abbreviation for route demon and pronounced Route dee)

Timer in RIP:

- 1- Periodic timer: 25 \rightarrow 35s
- 2- expiration time: 180s
- 3- garbage collision timer: 120s

Performance:

update message: Local Simple

Convergence: RIP used distance vector (16 considered infinity)

Robustness: if there is a failure or corruption in one router then it will be affected for all router.

(Dijkstra Algo).

② metric in OSPF: layer 3 we calculate from source Router to host destination OSPF that different (TOS) * the same A.S with RIP

1 - $N_1 = 5 + 3 + 4 = 12$

2 - $N_2 = 3 + 4 = 7$

3 - $N_3 = 4$ cost

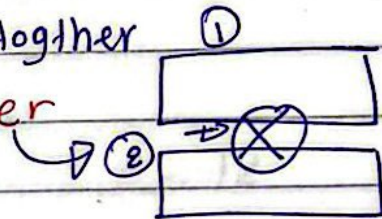
Area 5: we need to divided into small Area to prevent the volume of traffin in a ~~large~~ Large A.S. and we need

RIP and OSPF - the reason for this consistency is that both protocol use shortest path -ness to define best Route from source to destion

Area Backbone for gluing area together

and we have Area Border Router

Performance:



① Update: complex format

② Convergence of forwarding table: quickly

③ Robustness: better than RIP.

③ BGP: Border Gateway Protocol only interconnection using in internet today

