# פרויקט סיום

342615648 : אילן מאיר סופיר

בן כהן: 207029786

# <u>חלק א</u>

In this first part, we were asked to build in python : a primitive instant messaging system (similar to MSN) based on network.

רשימת הודעות שנשלחות מהלקוח (לשרת):

- <connect><name> : אני רוצה להתחבר (1
- <get\_users> :אני רוצה לקבל את רשימת המחוברים
  - <disconnect> אני רוצה להתנתק: (3
- <set\_msg><Name> : לקוח: לשלוח הודעה לשלוח אני רוצה לשלוח הודעה (4
  - <set msg all> אני רוצה לשלוח הודעה לכולם: (5
- <get\_list\_file> אני רוצה לקבל את רשימת הקבצים שיש בשרת: <6</p>
- <test.txt": <download> < test.txt > <id" אני רוצה להוריד קובץ (7
  - ced> : אמשך בהעברת הקובץ

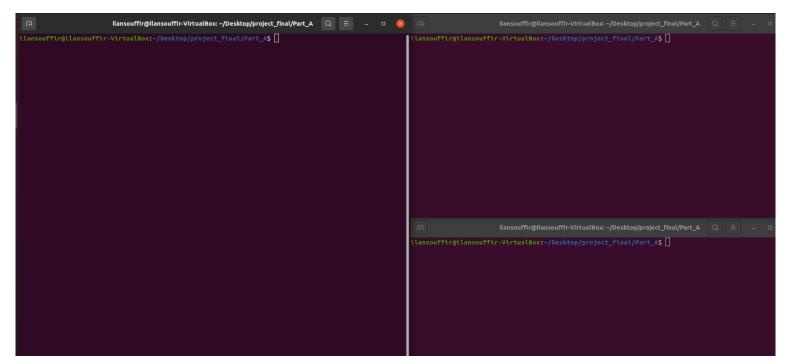
Then we created 2 classes: server1.py and client1.py to solve the requested exercise.

For our exemple we wanted there to be 2 users who communicate with each other on the same server, their names will be Ilan and Ben.

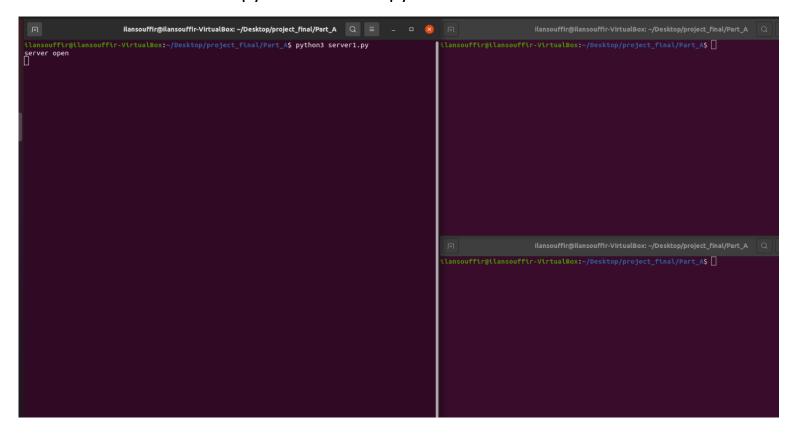
#### How to Run?

We run our messaging system on linux

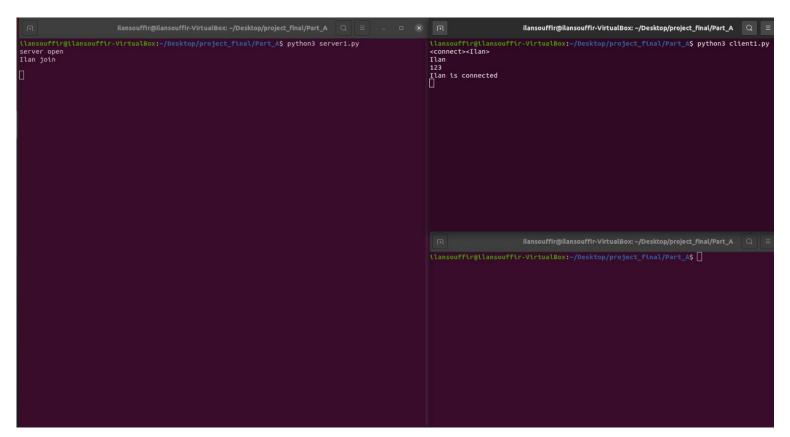
1. Open 3 terminals (one for the server and two for the users Ilan and Ben).



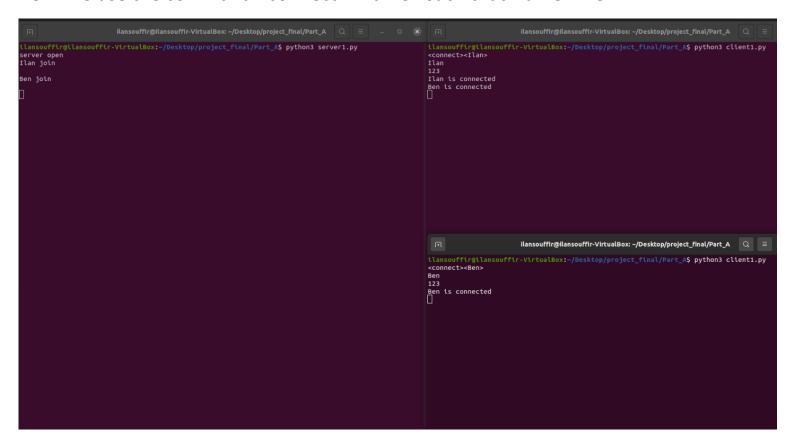
2. Write in the first one « python3 server1.py » to create the server.



3. Write in the second one « python3 client1.py » to create the first user with name llan. We use the command <connect><Name> such that Name = Ilan.

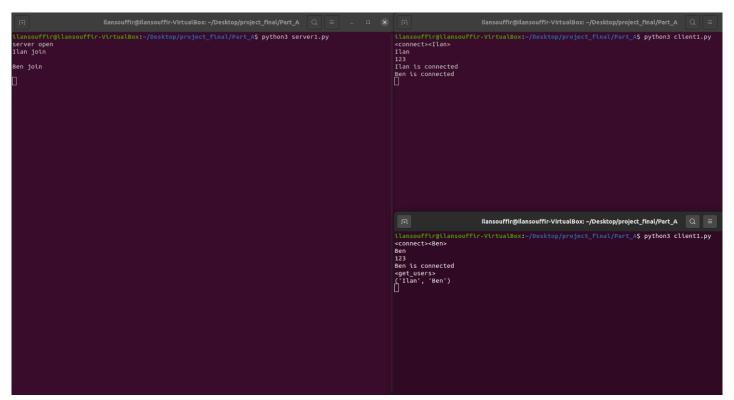


4. Write in the thirdone « python3 client1.py » to create the second user with name Ben. We use the command <connect><Name> such that Name = Ben.



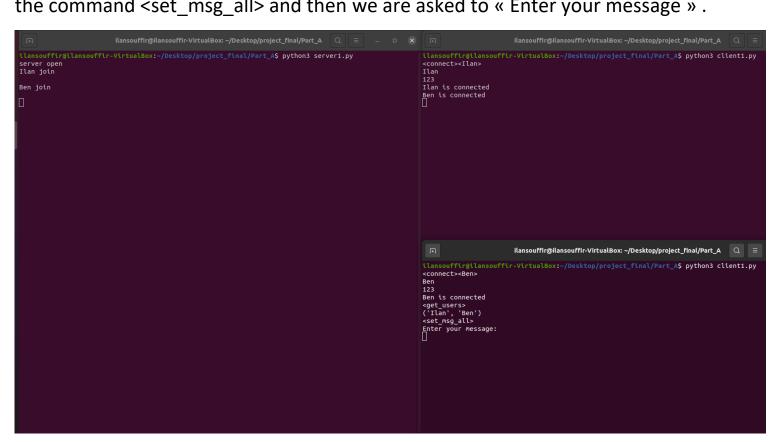
We can see on the server that Ilan can see that Ben is online now and now Ilan and Ben are connected to the same server, so they can communicate with each other.

5. Ben wants to know with whom he will be able to communicate so he wants to know the list of people connected on the server (on MSN) using the command <get\_users> .

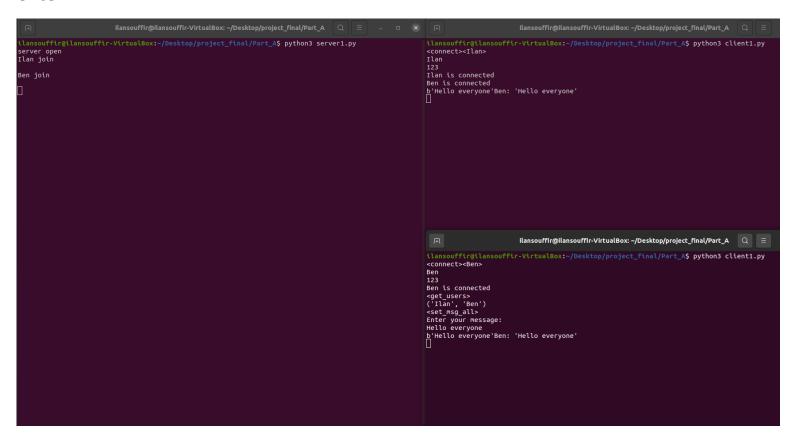


We can see that there is only two users: Ilan and Ben.

6. Ben wants now to send a message to everyone that is in the messenger, so we use the command <set\_msg\_all> and then we are asked to « Enter your message » .

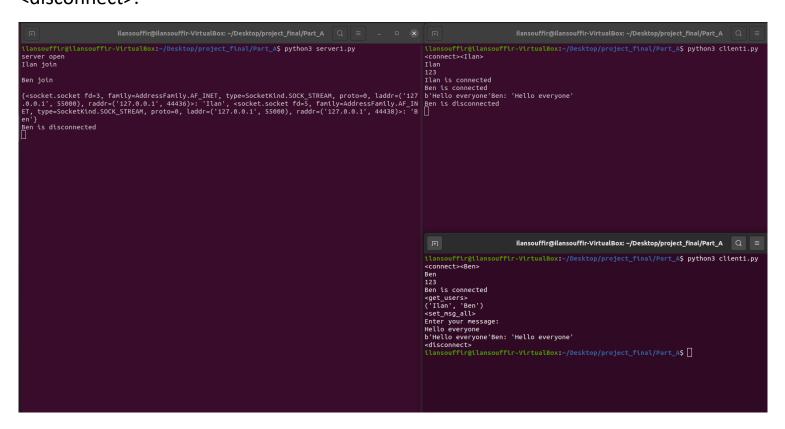


7. Ben writes the message "Hello everyone" and sends it to everyone by clicking on enter.



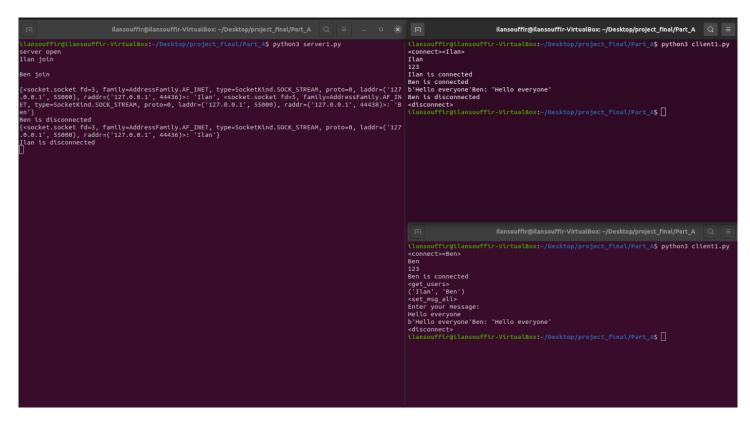
We can see that Ilan got the message from Ben.

8. Ben now wants to be offline so he disconnects from messenger using the command <disconnect>.

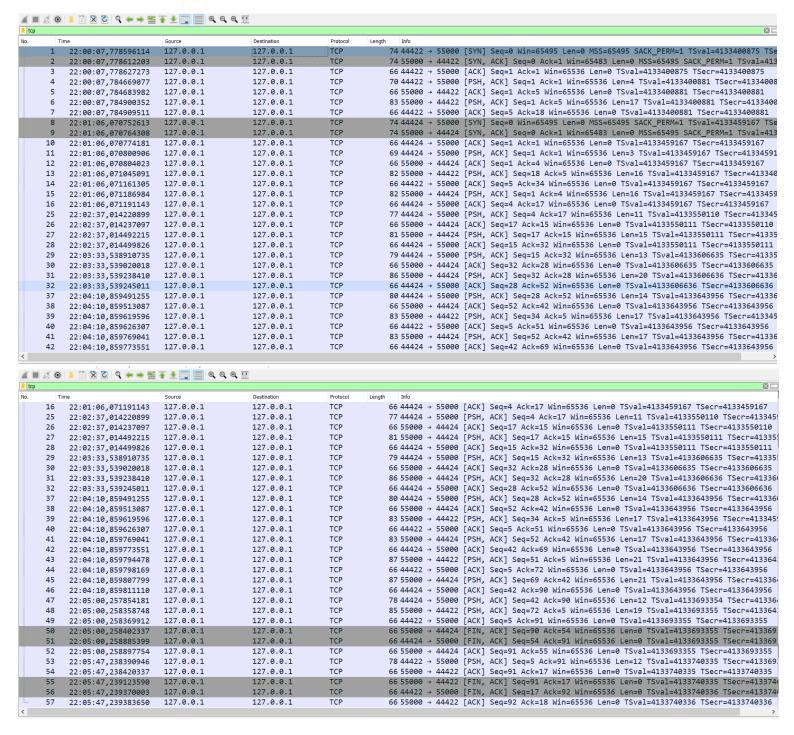


Ilan see that Ben is disconnect so he is now alone on messenger.

9. So Ilan wants to be offline too so he disconnects from messenger using the command <disconnect> too.



#### Capture on Wireshark of the example seen before (capture1.pcapng):



Lines 1-7: Ilan connects to messenger.

Lines 8-16: Ben connects to messenger.

Lines 25-28 : command <get\_users> from Ben.

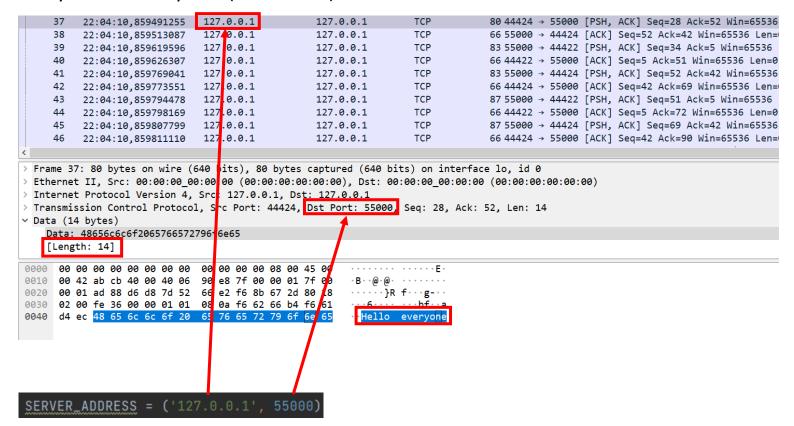
Lines 29-32 : command <set\_msg\_all> from Ben.

Lines 37-46: Ben sends « Hello everyone ».

Lines 47-52: Ben go to offline using command <disconnect>

Lines 53-57: Ilan go to offline too with the same command.

We will dissect with precision the example of when Ben sends the message "Hello everyone" to everyone. (lines 37-46).



<u>Line 37</u>: Ben write a message to all the clients saying « Hello everyone ». We can see that the length of the message is 14 (5 letters for the first word + 1 space + 8 letters for the second word).

<u>Line 38</u>: the server sent an ACK back to Ben.

<u>Line 39</u>: the server sent to Ilan the message that he sent as the broadcast message to the user Ilan that is connected to the server at the same time.

Line 40: Ilan sent an ACK back to the server.

<u>Line 41:</u> the server sent the message to Ben saying « Hello everyone ».

Line 42: Ben sent an ACK back to the server.

<u>Line 43</u>: the server sent the message to Ilan saying « Ben: Hello everyone ».

Line 44: Ilan sent an ACK back to the server.

Line 45: the server sent the message to Ben saying « Ben: Hello everyone ».

Line 46: Ben sent an ACK back to the server.

In this part, we were asked to build in python with the code from part A: to add a new layer for transferring files over UDP which is called FAST reliable UDP

And then we need to answer the questions:

- 1 How works the system overcomes packet loss?
- 2 How the system overcomes latency problems?
- 3 Diagram of situations in which the system.

We created 2 classes: server1.py and client1.py to solve the requested exercise.

#### 1:

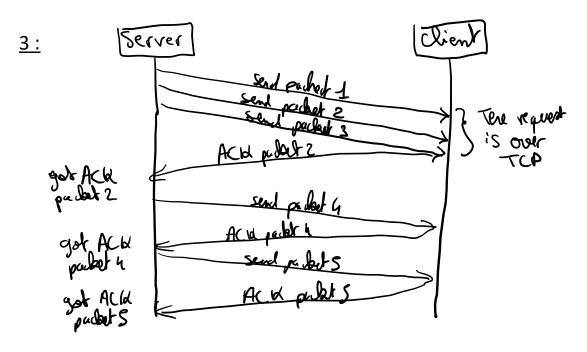
In our python code system, we are using 2 protocols: TCP and UDP.

We saw in Part A that we need TCP protocol when a user (client) wants to connect to the server messenger and then he wants to send a message to everyone or for a particular user in the messenger. In that case, if there is a packet loss, the sender resends the packet and slows down the sending of data over the network by reducing the size of the message to send. If the sender has a packet loss, he resends the packets from the first packet that didn't receive an ack. If a user wants a file download, we are using the UDP protocol. But, we implement a RDT over UDP in this case

#### <u>2:</u>

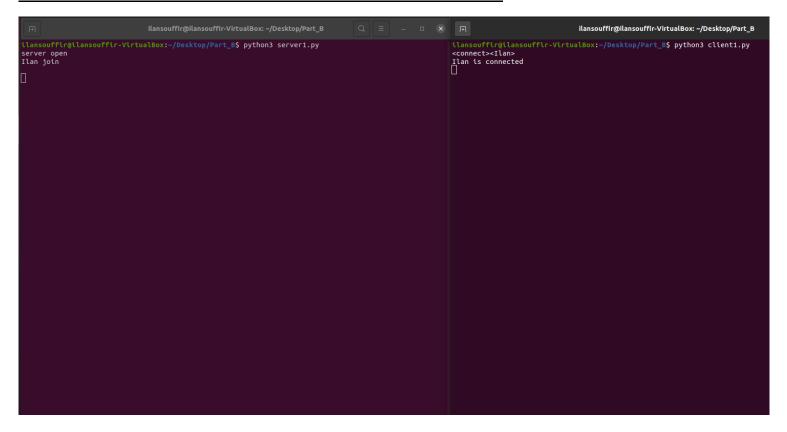
For the latency problems, we used a go-back-n RDT algorithm, we have N packets from the last packet that we received, if all the acks were received, we will send n new packets.

For the user, we send in the order we received them and the server will start sending after the first duplicate ack.

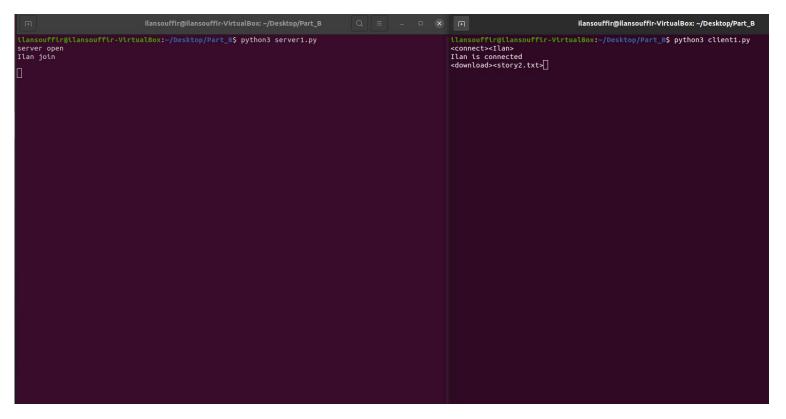


#### How to Run?

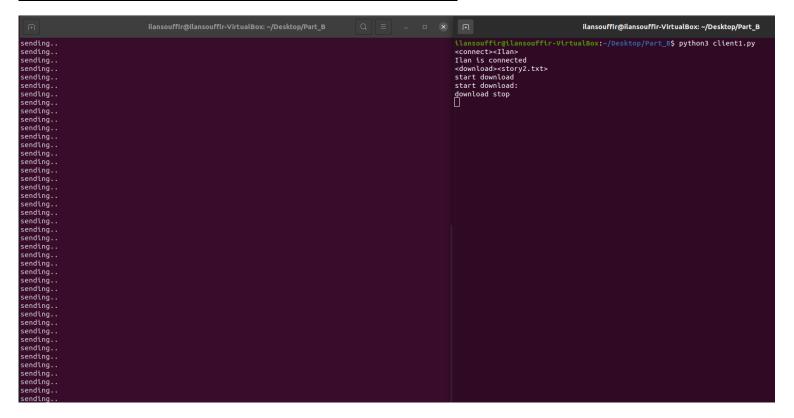
1. <connect><llan> : create a new client with name Ilan.



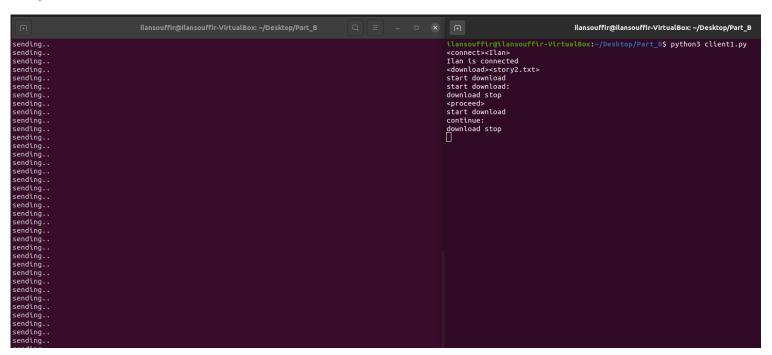
#### 2. <download><story2.txt> to download our file.txt



# 3. we can see that on the server the file is sending.



# 4.ceed>



#### <u>Capture on Wireshark of the example seen before (capture2.pcapng):</u>



We can see that we are using the protocol TCP to send messages and connect to the server.

 $\Diamond$ 

e 💼

12°C ^ @ ■ □ □) ♥ A FRA 05/03/2022

And the protocol UDP for the sends of files.

Taper ici pour rechercher

# חלק ג

: ענו על השאלות הבאות ללא קשר לחלקים הקודמים. החלק הזה עומד בפני עצמו

- בהינתן מחשב חדש המתחבר לרשת אנא תארו את כל ההודעות שעוברות החל מהחיבור switch ועד שההודעה מתקבלת בצד השני של הצאט. אנא פרטו לפי הפורמט הבא:
   a סוג הודעה, פירוט הודעה והשדות הבאים
  - .i כתובת IP מקור/יעד, כתובת פורט מקור/יעד, כתובת MAC מקור/יעד. פרוטוקול שכבת התעבורה.

#### The host joins the network:

1. Host DHCP discover message.

Src: 10.0.2.15, port: 50, MAC: 00:37:6C:E2:EB:62

Dest: 192.168.1.1, port: 55, MAC: 19:55:Y7:R5:FR:34

Protocol: UDP

2. The DHCP server responds with a DHCP message. Src: 192.168.1.1, port: 55, MAC: 19:55:Y7:R5:FR:34 Dest: 10.0.2.15, port: 50, MAC: 00:37:6C:E2:EB:62

Protocol: UDP

3. The host request IP address: DHCP request message.

Src: 10.0.2.15, port: 50, MAC: 00:37:6C:E2:EB:62

Dest: 192.168.1.1, port: 55, MAC: 19:55:Y7:R5:FR:34

Protocol: UDP

4. The DHCP sends IP address: DHCP Ack message.

Src: 192.168.1.1, port: 55, MAC: 19:55:Y7:R5:FR:34

Dest: 10.0.2.15, port: 50, MAC: 00:37:6C:E2:EB:62

Protocol: UDP

#### The new Client joins the server:

1. He sends a connect request to the server.

Src: 127.0.0.1, 55015, MAC: 00:34:8U:Y6:CF:44

Dest: 127.0.0.1, 55000, MAC: 00:34:8U:Y6:CF:44

Protocol: TCP

2. The server responds with an accept message.

Src: 127.0.0.1, 55000, MAC: 00:34:8U:Y6:CF:44

Dest: 127.0.0.1, 55015, MAC: 00:34:8U:Y6:CF:44

Protocol: TCP

3. The client sends a request 'login' to the server with a TCP Socket.

Src: 127.0.0.1, 55015, MAC: 00:34:8U:Y6:CF:44 Dest: 127.0.0.1, 55000, MAC: 00:34:8U:Y6:CF:44

Protocol: TCP

4. The server sends a response 'login' to the client.

Src: 127.0.0.1, 55000, MAC: 00:34:8U:Y6:CF:44 Dest: 127.0.0.1, 55015, MAC: 00:34:8U:Y6:CF:44

Protocol: TCP

# The server informs all the other clients in the network that there is a new client:

1. Server sends a message to all the clients on the network that there is a new client.

Src: 127.0.0.1, 55000, MAC: 00:34:8U:Y6:CF:44

Dest: to all the clients on the server.

Protocol : TCP.

2. הסבירו מה זה CRC

#### From internet:

CRC (cyclic redundancy check) is an error detecting code used to determine if a certain packet/block of data has been corrupted. Its commonly used in network protocols such as TCP/IP for data verification reasons and more. The CRC algorithm computes a checksum for each specific data set that is sent through it. This algorithm utilizes a polynomial key and the transferred data.

In summary, The Cyclic Redundancy Check is a number mathematically calculated for a packet by its source computer, and then recalculated by the destination computer. If the original and recalculated versions at the destination computer differ, the packet is corrupt and needs to be resent or ignored.

#### Exemple to calculate the CRC:

http 1.0, http 1.1, http 2.0, QUIC מה ההבדל בין

#### From internet:

#### http 1.0 :

- -Browser-friendly protocol
- -Provided header fields including rich metadata about both request and response (HTTP version number, status code, content type)
- -Response: not limited to hypertext (Content-Type header provided ability to transmit files other than plain HTML files e.g. scripts, stylesheets, media)
- -Methods supported: GET , HEAD , POST
- -Connection nature: terminated immediately after the response

Etablishing a new connection for each request – major problem :

HTTP/1.0 required to open up a new connection for each request (and close it immediately after the response was sent). Each time a new connection establishes, a TCP three-way handshake should also occur. For better performance, it was crucial to reduce these round-trips between client and server. HTTP/1.1 solved this with persistent connections.

#### http 1.1:

- -This is the HTTP version currently in common use.
- -Introduced critical performance optimizations and feature enhancements persistent and pipelined connections, chunked transfers, compression/decompression, content negotiations, virtual hosting (a server with a single IP Address hosting multiple domains), faster response and great bandwidth savings by adding cache support.
- -Methods supported: GET, HEAD, POST, PUT, DELETE, TRACE, OPTIONS
- -Connection nature: long-lived

#### Keep-Alive header:

- -The Keep-Alive header was used prior to HTTP/1.1 and was obsoleted by HTTP/1.1 making persistent connections the default behavior. Keep-Alive header can be used to define policies for long-lived communication between hosts (i.e. allows a connection to stay active until an event occurs). This laid foundation for persistence, reusable connections, pipelining, and many more enhanced capabilities in modern web communication protocols.
- -Client, server, or any intermediary can provide information for Keep-Alive header independently. Also, a host can add timeout and max parameters in order to set a timeout or limit maximum request count per connection.
- -HTTP pipelining, multiple connections, and many more improvements have been implemented, thanks to the Keep-Alive header's behavior.

#### Upgrade header:

- -With Upgrade header introduced in HTTP/1.1, it is possible to start a connection using a commonly-used protocol, such as HTTP/1.1, then request that the connection switch to an enhanced protocol type like HTTP/2.0 or WebSockets.
- -In an upgraded protocol connection, max parameter (maximum request count) is not present. The upgraded protocol can provide new policies for timeout parameter (if not specifically defined, it uses default timeout value in underlying protocol).

#### http 2.0:

More visual media was displayed and the volume and size of scripts adding interactivity also increased. Much more data was transmitted over significantly more HTTP requests and this created more complexity and overhead for HTTP/1.1 connections.

The HTTP/2 protocol differs from HTTP/1.1 in a few ways:

- -It's a binary protocol rather than a text protocol. It can't be read and created manually. Despite this hurdle, it allows for the implementation of improved optimization techniques.
- -It's a multiplexed protocol. Parallel requests can be made over the same connection, removing the constraints of the HTTP/1.x protocol.
- -It compresses headers. As these are often similar among a set of requests, this removes the duplication and overhead of data transmitted.
- -It allows a server to populate data in a client cache through a mechanism called the server push.

High-traffic websites showed the most rapid adoption in an effort to save on data transfer overhead and subsequent budgets.

This rapid adoption was likely because HTTP/2 didn't require changes to websites and applications. To use it, only an up-to-date server that communicated with a recent browser was necessary. Only a limited set of groups was needed to trigger adoption, and as legacy browser and server versions were renewed, usage was naturally increased, without significant work for web developers.

HTTP hasn't stopped evolving since the release of HTTP/2. Like with HTTP/1.x, HTTP's extensibility is still being used to add new features. Notably, we can cite new extensions of the HTTP protocol that appeared in 2016:

- -Support for <u>Alt-Svc</u> allowed the dissociation of the identification and the location of a given resource. This meant a smarter CDN caching mechanism.
- -The introduction of Client-Hints allowed the browser or client to proactively communicate information about its requirements and hardware constraints to the server.
- -The introduction of security-related prefixes in the Cookie header helped guarantee that secure cookies couldn't be altered.

#### **QUIC:**

QUIC is a new application layer protocol proposed by Google to be used for transferring web pages. QUIC is a novel approach and is starting from a fresh slate unlike HTTP.

The biggest difference between QUIC and HTTP is QUIC runs on top of UDP whereas HTTP runs on top of TCP. QUIC moves the TCP reliability, congestion control, and in order delivery to the application layer thus giving it more control and allowing for changes to be applied easier. For example, one of the features of QUIC is its congestion control algorithm is pluggable thus you can choose whichever congestion control algorithm you like without having to change the underlying network stack. QUIC offers the same multiplexing capabilities as HTTP/2 without the issue of head of line blocking. QUIC creates multiple independent streams over its single UDP connection and a loss on one of the streams doesn't cause the other streams to become blocked. QUIC also reduces the connection setup time by introducing a 1 RTT handshake unlike TCP which relies on a 3 RTT handshake.

# ?port מספרי מריך למה צריך 4.

Ports identify what process on the host the received traffic should be sent to. A computer can have multiple simultaneous connections, all receiving data for different processes (mail, web, database ...) on that computer. A port is always associated with a protocol.

Some of these port numbers are specifically defined and always associated with a specific type of service -- for example, File Transfer Protocol (FTP) is always port number 21 and Hypertext Transfer Protocol web traffic is always port 80. These are called *well-known ports* and go from 0 to 1023.

In other words, the IP address identifies the computer host, and the port number specifies what specific process is running on that host.

# ?ה מה זה subnet ולמה צריך את זה?

A subnet is a network inside a network. Subnets make networks more efficient. Through subnetting, network traffic can travel a shorter distance without passing through unnecessary routers to reach its destination.

In a network, there could be millions of connected devices, and it could take some time for the data to find the right device. This is where subnetting comes in handy. subnetting narrows down the IP address to usage within a range of devices and shorten the path for a specific packet because of it.

A media access control address (MAC address) is a unique identifier assigned to a network interface controller (NIC) for use as a network address in communications within a network segment. This use is common in most IEEE 802 networking technologies, including Ethernet, Wi-Fi, and Bluetooth.

An IP address is a unique address that identifies a device on the internet or a local network. IP stands for "Internet Protocol," which is the set of rules governing the format of data sent via the internet or local network.

In essence, IP addresses are the identifier that allows information to be sent between devices on a network: they contain location information and make devices accessible for communication. The internet needs a way to differentiate between different computers, routers, and websites. IP addresses provide a way of doing so and form an essential part of how the internet works.

Both MAC Address and IP Address are used to uniquely define a device on the internet. The main difference between MAC and IP address is that MAC Address is used to ensure the physical address of the computer. It uniquely identifies the devices on a network. While IP addresses are used to uniquely identifies the connection of the network with that device takes part in a network.

Even if your computer has an IP address, it still needs a MAC address to find other machines on the same network (especially the router/gateway to the rest of the network/internet), since every layer is using underlying layers.

#### ?Router Switch Nat מה ההבדל בין

First we will see the difference between a router and a switch :

The switch and router are the two essential components of a network. Although both are the connecting devices in a network, and sometimes people assume them as the same, both have different functionality.

A switch allows you to have all your peripherals connected to the same place and distributes the flow on the different ports. It is rather recommended for professional use.

A router offers additional functions (security, WiFi connection, etc.). It is rather recommended within a home, especially for gaming enthusiasts.

A router is a device that allows communication between your computer and the Internet, while a switch (or switch) allows you to connect several devices within the same Ethernet network.

Router store IP address in the routing table whereas Switch store MAC address in a lookup table. Routers supports NAT while Switches does not.

And now we will see the difference between NAT and both of them:

What is a NAT? Network address translation (NAT) stands for network address translation. It's a way to map multiple local private addresses to a public one before transferring the information. Organizations that want multiple devices to employ a single IP address use NAT, as do most home routers.

Also, it does the translation of port numbers in the packet that will be routed to the destination. It then makes the corresponding entries of IP address and port number in the NAT table. NAT generally operates on a router.

The global difference is that NAT is an algorithm and Routers and Switches are computers that allow communication between networks and devices.

NAT is a platform that builds on a Router since its works on the Network layer and Switches works in the Link layer.

# 8. שיטות להתגבר על המחסור בPv4 ולפרט?

There are methods to overcome the lack of IPv4 addresses such as:

#### 1. Retrieval of address blocks

use of formerly reserved address blocks (like 14.0.0.0/8). Stanford University thus returned block 36.0.0.0/8 to the IANA in 2000 and Interop did the same for block 45.0.0.0/8 in 2010.

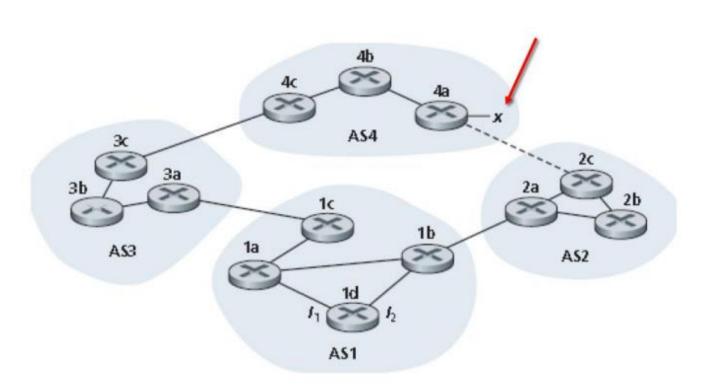
#### 2. IPv6

Accelerating the transition to IPv6: the only solution to preserve the internet. Accelerating the transition from IPv4 to IPv6 is the only sustainable solution, because only an almost total transition to IPv6 can allow content providers to do without IPv4.

#### 3. NAT

To ensure the continuity of Internet access despite the exhaustion of IPv4 addresses, operators are considering the deployment of large-scale address translators (Large Scale NAT, Carrier-Grade NAT). A public address would thus be shared by many clients simultaneously. These methodes saves a lot of IPv4 and reduce the need.

- OSPF מריצים AS2, AS3 .a
  - RIP מריצים AS1, AS4 .b
    - BGP רץ Ass. c
- AS4 ,AS2 אין חיבור פיזי בין .d
- x בעזרת איזה פרוטוקול לומד הנתב 3c על תת רשת .e
- x על תת רשת 3a בעזרת איזה פרוטוקול לומד הנתב 3. על תת רשת .f
- x על תת רשת 1c בעזרת איזה פרוטוקול לומד הנתב .g
- x על תת רשת 2c בעזרת איזה פרוטוקול לומד הנתב .h

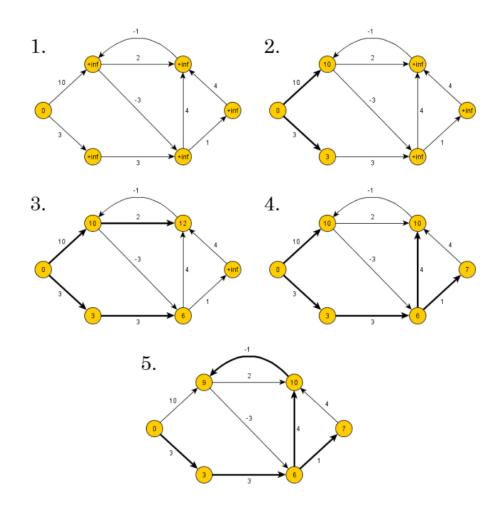


OSPF(Open Shortest Path First): While routers connect networks using the IP protocol, OSPF technology is a protocol used to determine the best path that packets can take to travel through a series of connected networks.

<u>RIP</u> (Routing Information Protocol): is a distance-vector routing protocol. Routers running the distance-vector protocol send all or a portion of their routing tables in routing-update messages to their neighbors. You can use RIP to configure the hosts as part of a RIP network.

RIP use the algorithm of Bellman-Ford. It allows each router to communicate to neighboring routers.

# Exemple of Bellman-Ford algorithm:



<u>BGP (Border Gateway Protoco)</u>: is an external route exchange protocol (an EGP), used in particular on the Internet network. Its main purpose is to exchange network routing and reachability information (called prefixes) between Autonomous Systems (AS). Since it travels over TCP, it is considered to belong to the application layer of the OSI model.

#### **Answers**:

- e. eBGP
- f. iBGP
- g. eBGP
- h. iBGP