



## Lab 06

**Course:** Networks System Design

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**Instructor:** Mr. Kuy Movsun

**Due Date:** Tuesday, 2 December 2025, 12:00 AM

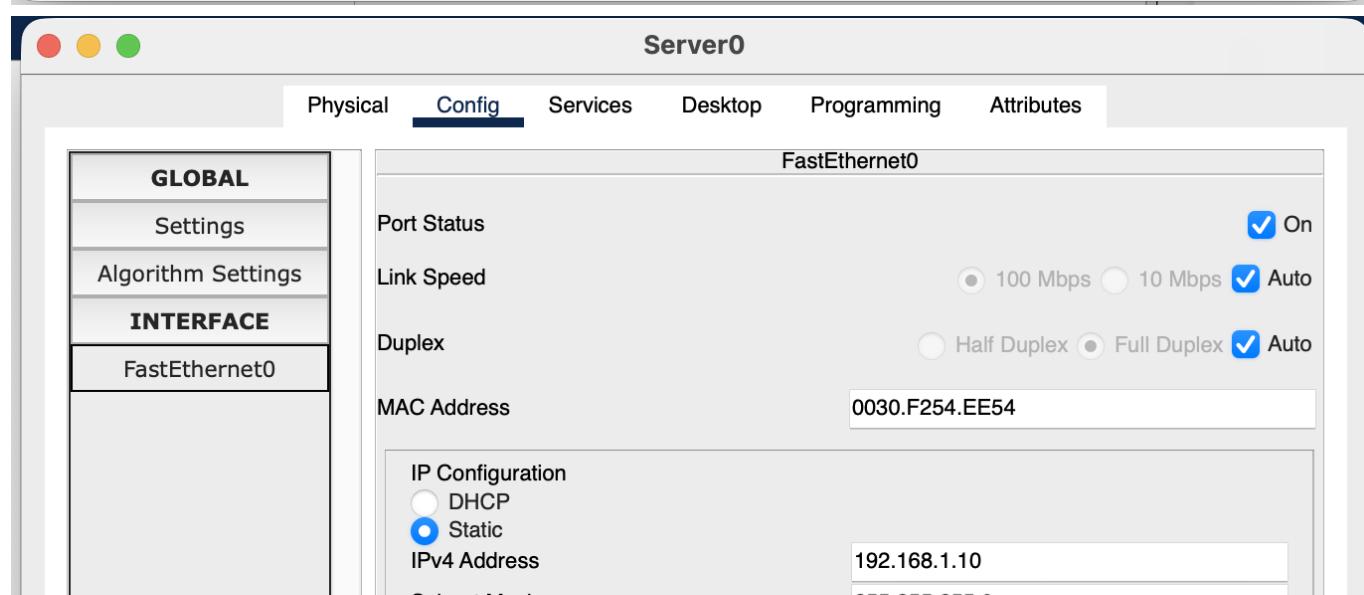
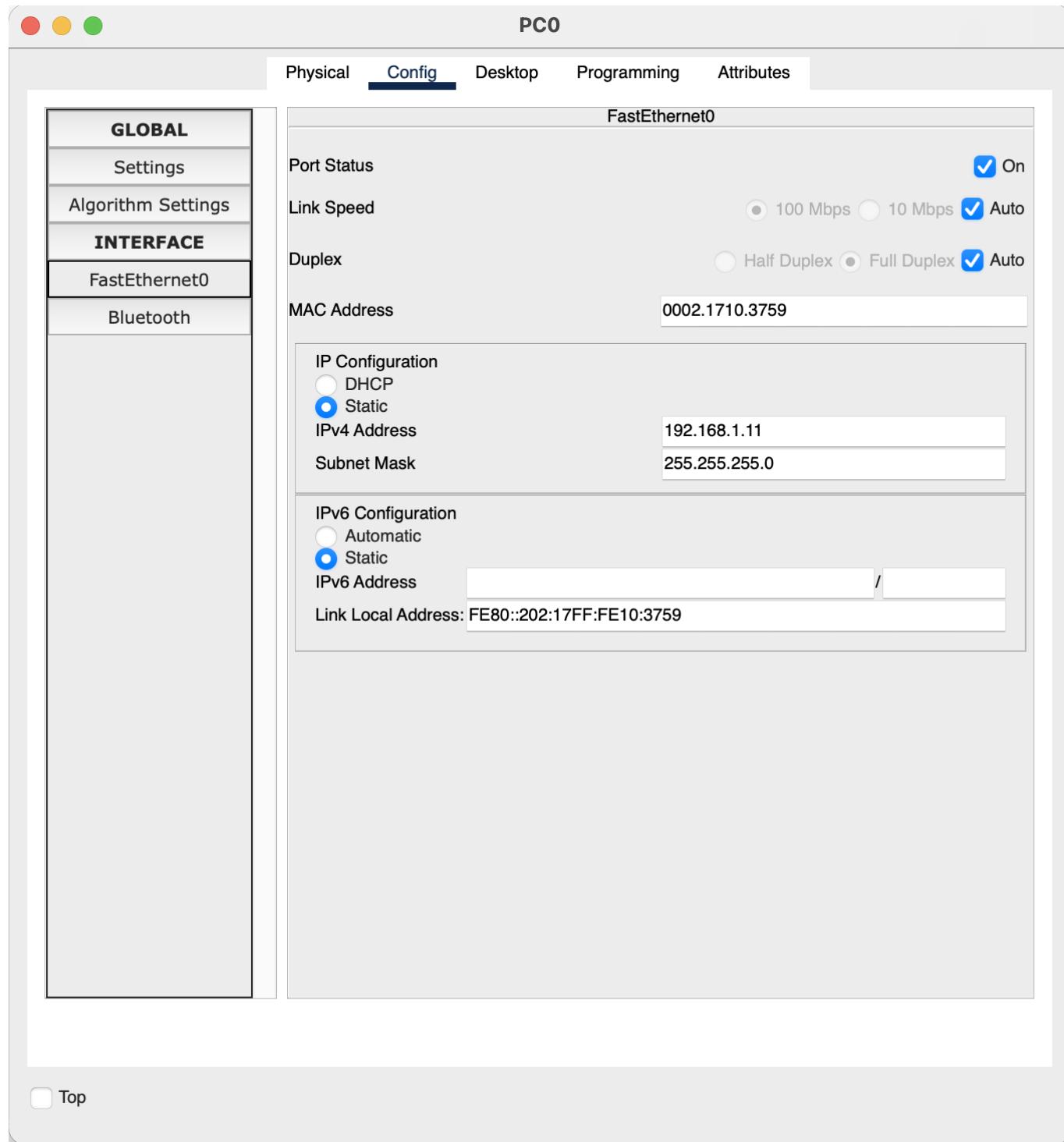
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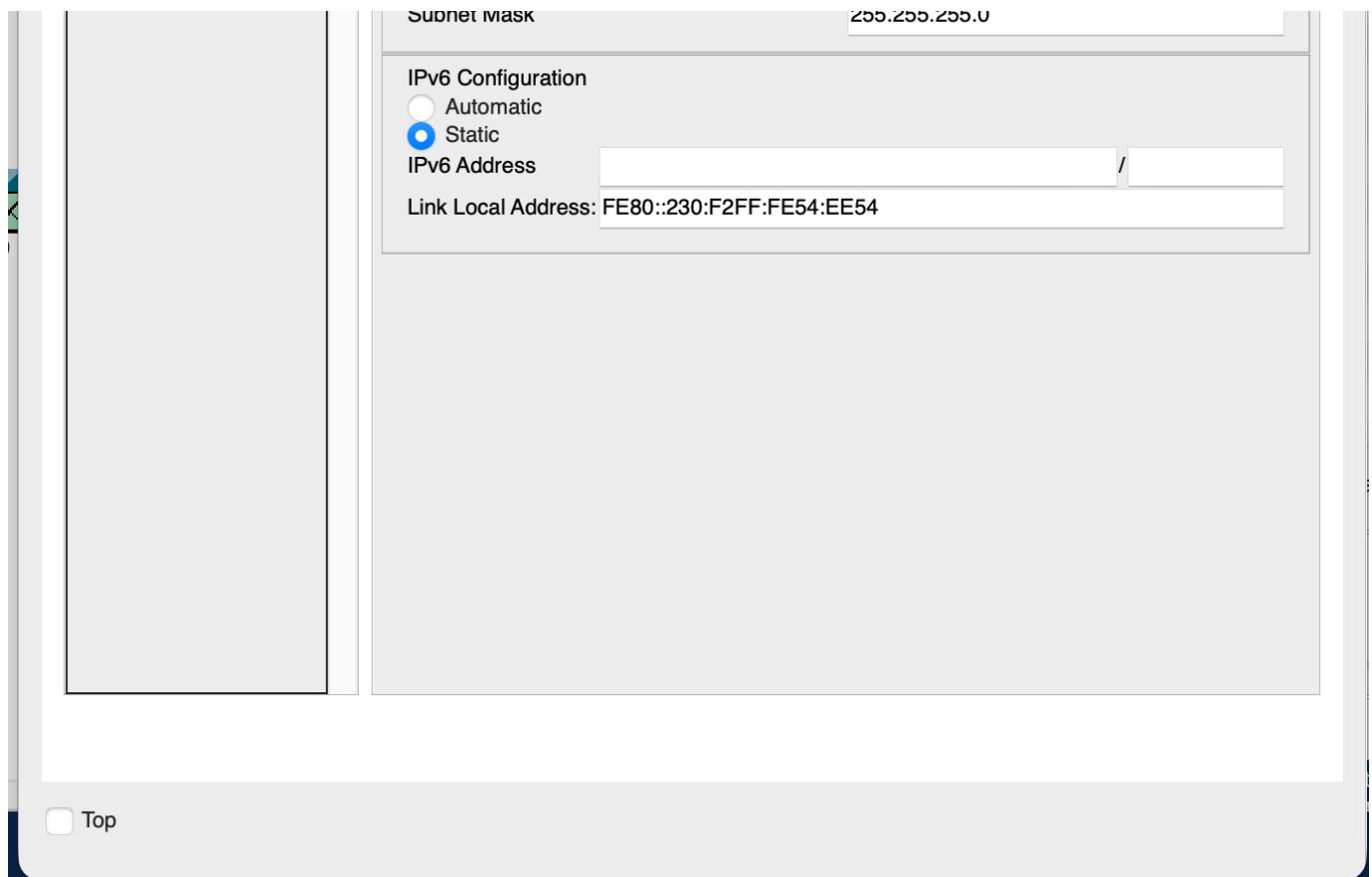
Link to my GitHub: <https://github.com/Do-Davin/Network-Lab.git>

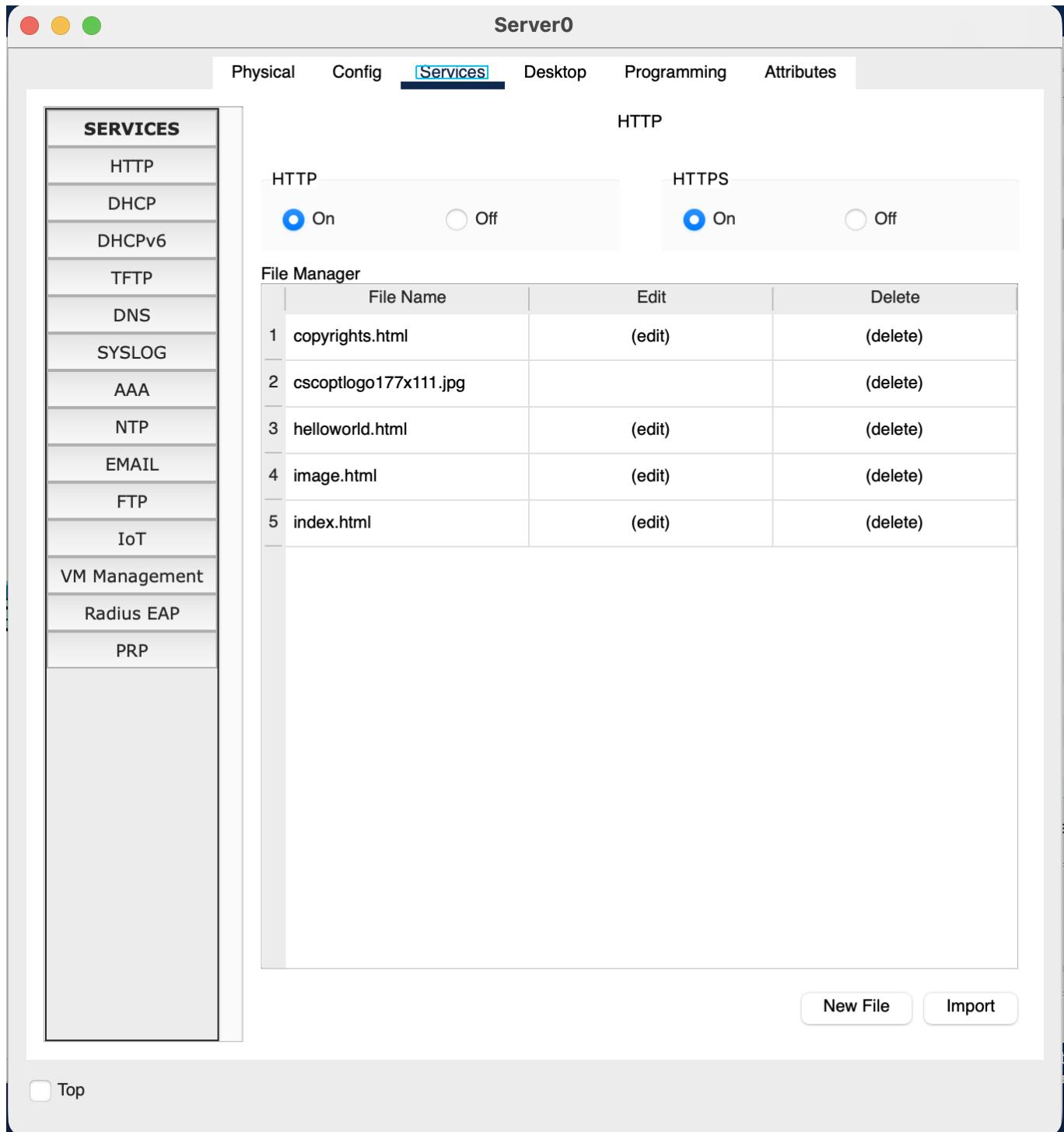
## Part 1: Lab Setup

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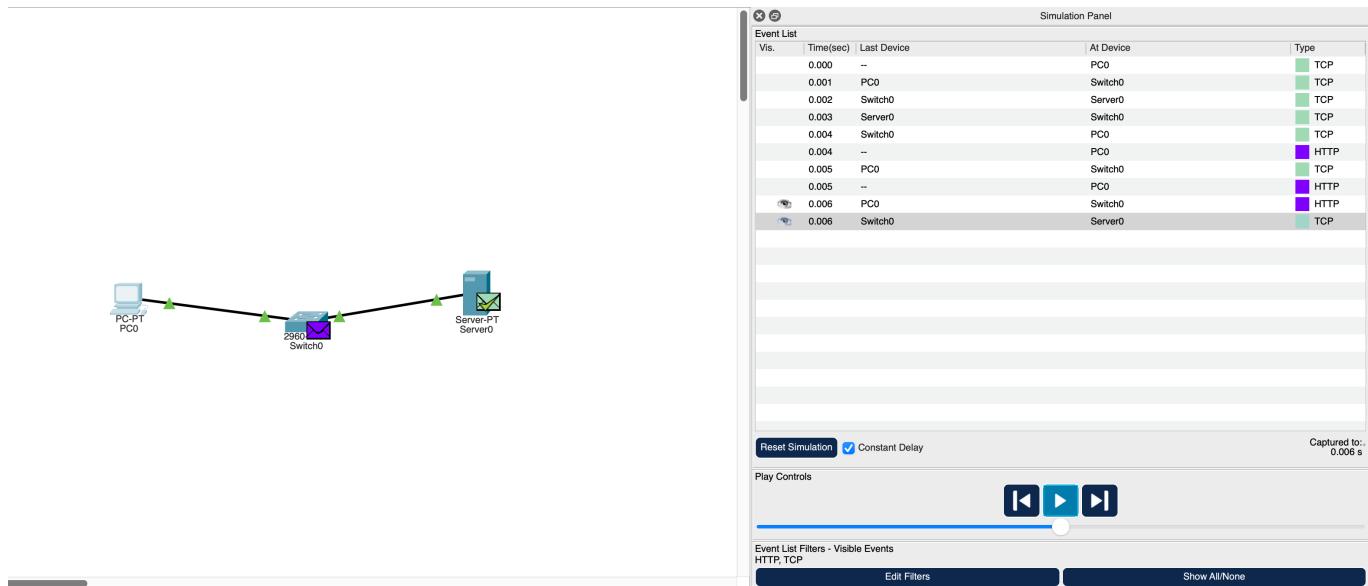
This section shows the basic network topology used in the simulation before starting all TCP experiments.







## Part 2: The 3-Way Handshake



## Lab Analysis

Packet 1 (PC → Server):

Flag = SYN

This is the first step: PC initiates the connection.

Packet 2 (Server → PC):

Flags = SYN and ACK

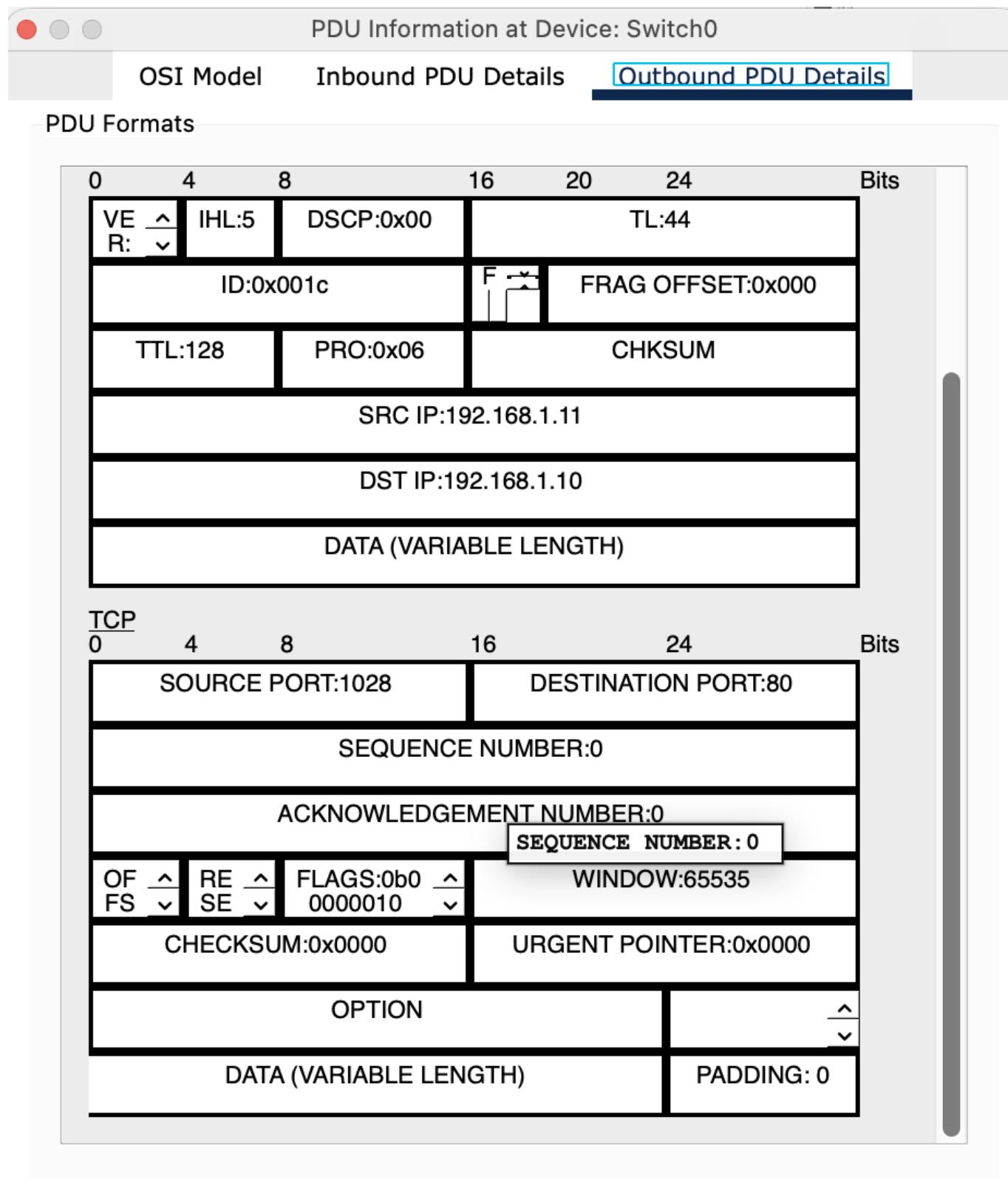
Server acknowledges the request and sends its own SYN.

Packet 3 (PC → Server):

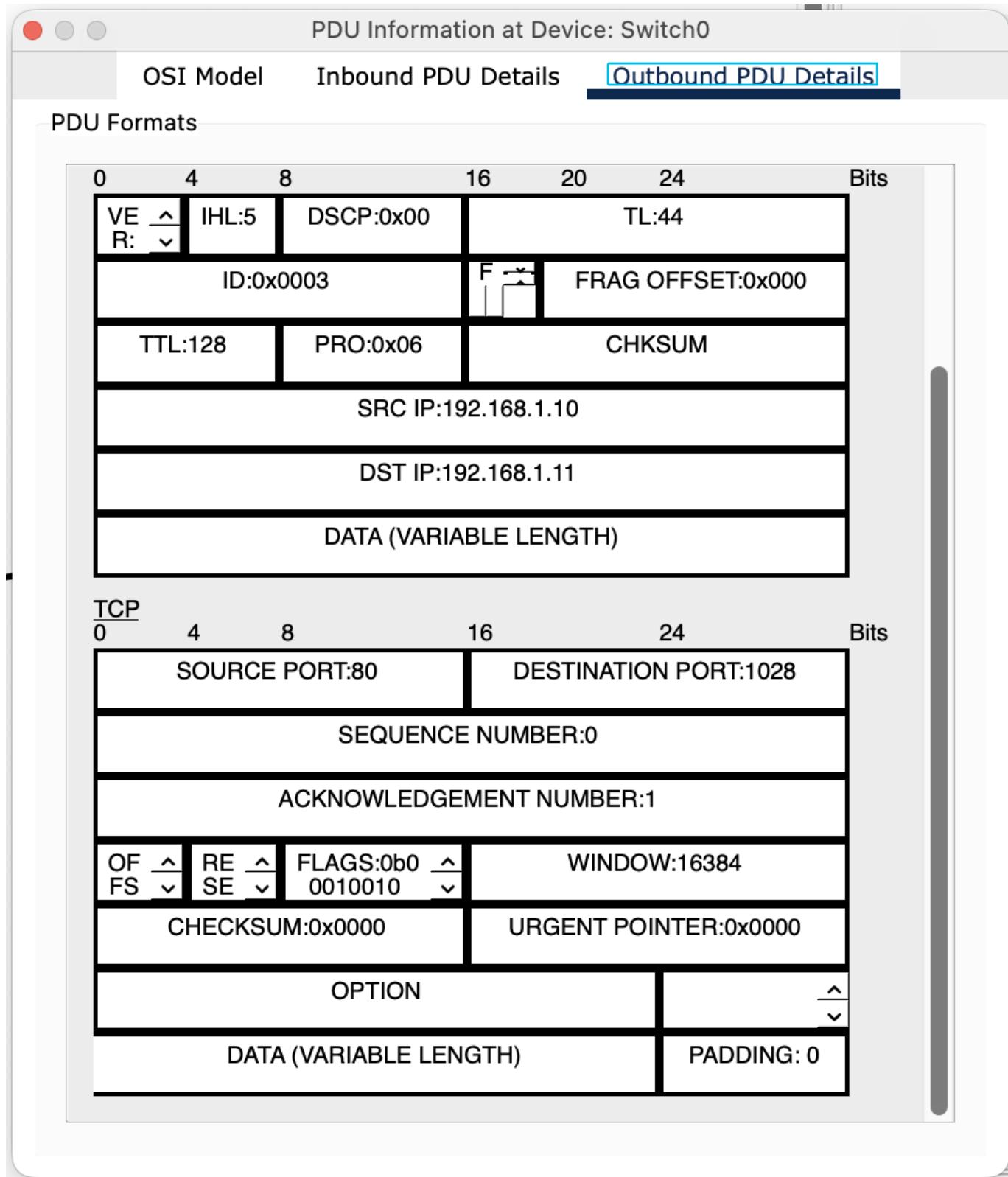
Flag = ACK

PC confirms the connection is established.

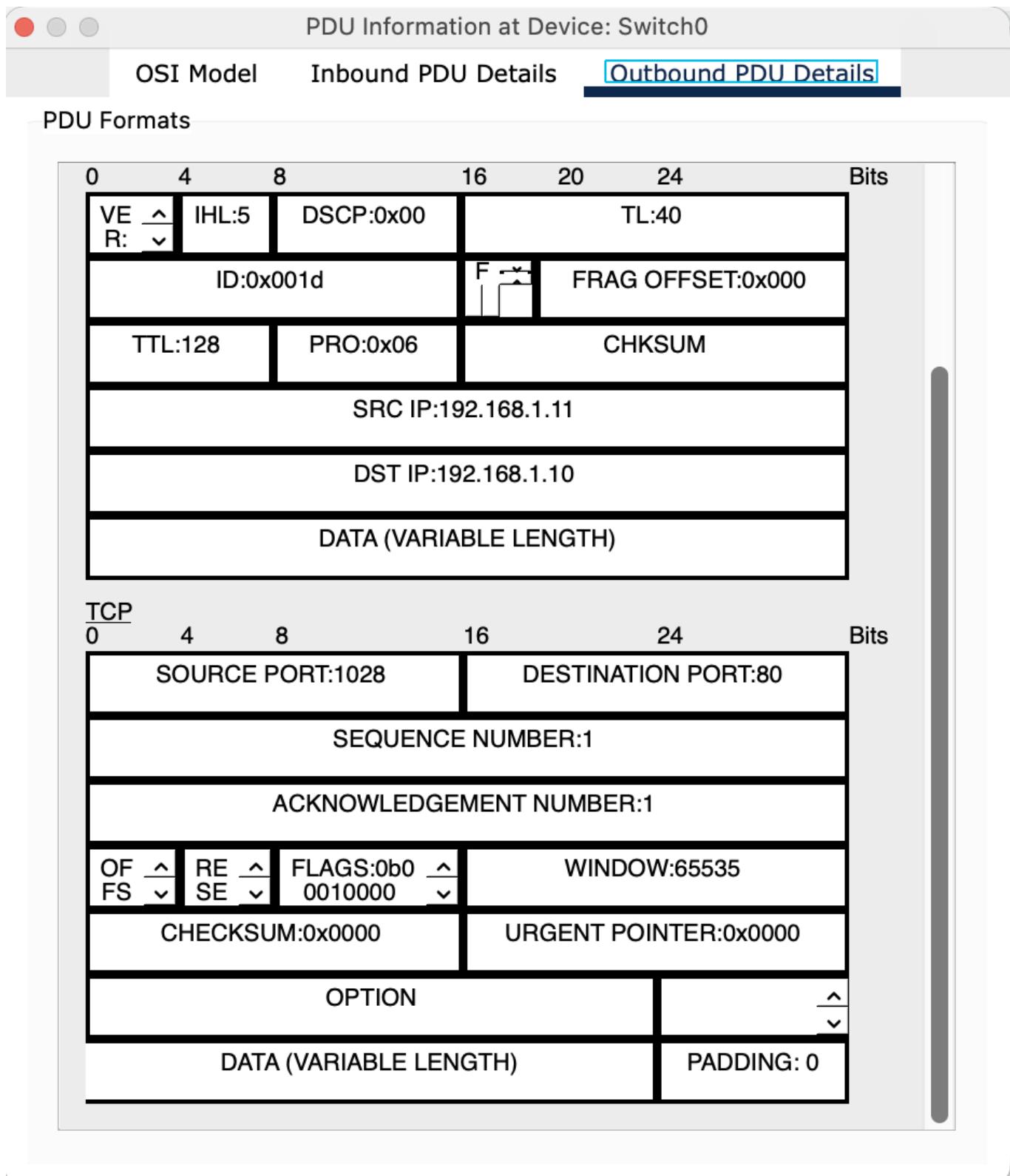
Packet 1:



Packet 2:



Packet 3:



Draw It: Sketch the 3 arrows on paper and label them (SYN, SYN-ACK, ACK).

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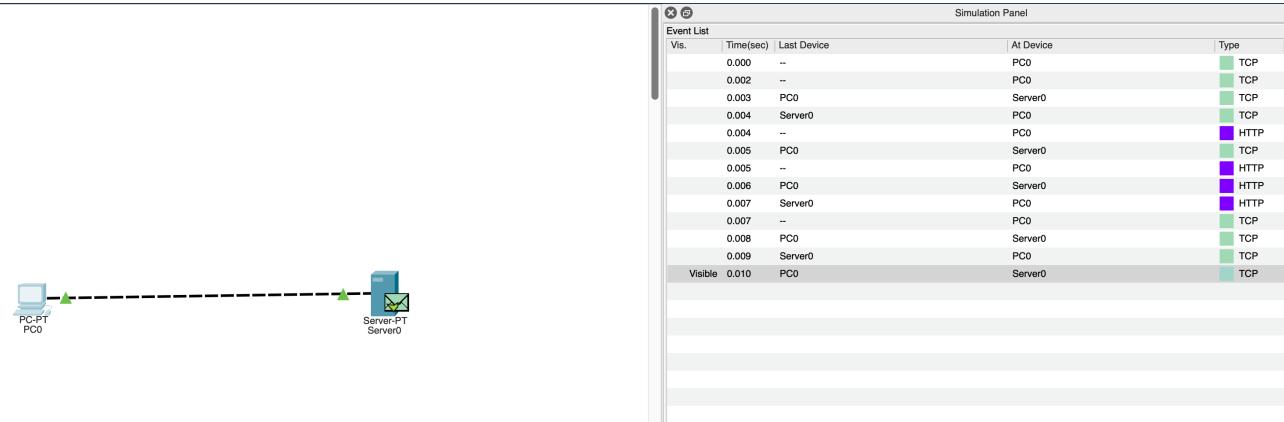
PC → Server : SYN
Server → PC : SYN-ACK
PC → Server : ACK

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Analysis

- The TCP 3-Way Handshake establishes a reliable connection before any data is exchanged.
- Once the handshake is complete, the HTTP request-response process can start.

## Part 3: The Sequence Number Detective



### Task

- Continue simulation after the TCP handshake.
- Capture the HTTP response packet from Server0 to PC1.
- Inspect the TCP Header for:
  - Sequence Number (Seq)
  - Data Length (Len)

### Steps

- Click Capture/Forward until the HTTP packet appears from Server0 → PC1.
- Open the packet details and record:
  - Sequence Number (Seq): 1
  - Data Length (Len): 369
- Predict the Acknowledgement Number:
  - Formula:

Expected ACK = Seq + Len

- Forward the next packet (PC1 → Server0).
- Confirm the Acknowledgement Number matches your prediction.



PDU Information at Device: PC0

**OSI Model      Inbound PDU Details**

At Device: PC0  
Source: PC0  
Destination: HTTP CLIENT

In Layers	Out Layers
Layer 7: HTTP	Layer7
Layer6	Layer6
Layer5	Layer5
Layer 4: TCP Src Port: 80, Dst Port: 1029	Layer4
Layer 3: IP Header Src. IP: 192.168.1.10, Dest. IP: 192.168.1.11	Layer3
Layer 2: Ethernet II Header 0030.F254.EE54 >> 0002.1710.3759	Layer2
Layer 1: Port FastEthernet0	Layer1

1. The device receives a TCP PUSH+ACK segment on the connection to 192.168.1.10 on port 80.  
 2. Received segment information: the sequence number 1, the ACK number 102, and the data length 471.  
 3. The TCP segment has the expected peer sequence number.  
 4. The TCP segment has the expected ACK number. The device pops the last sent segment from the buffer.  
 5. TCP processes payload data.  
 6. TCP reassembles all data segments and passes to the upper layer.

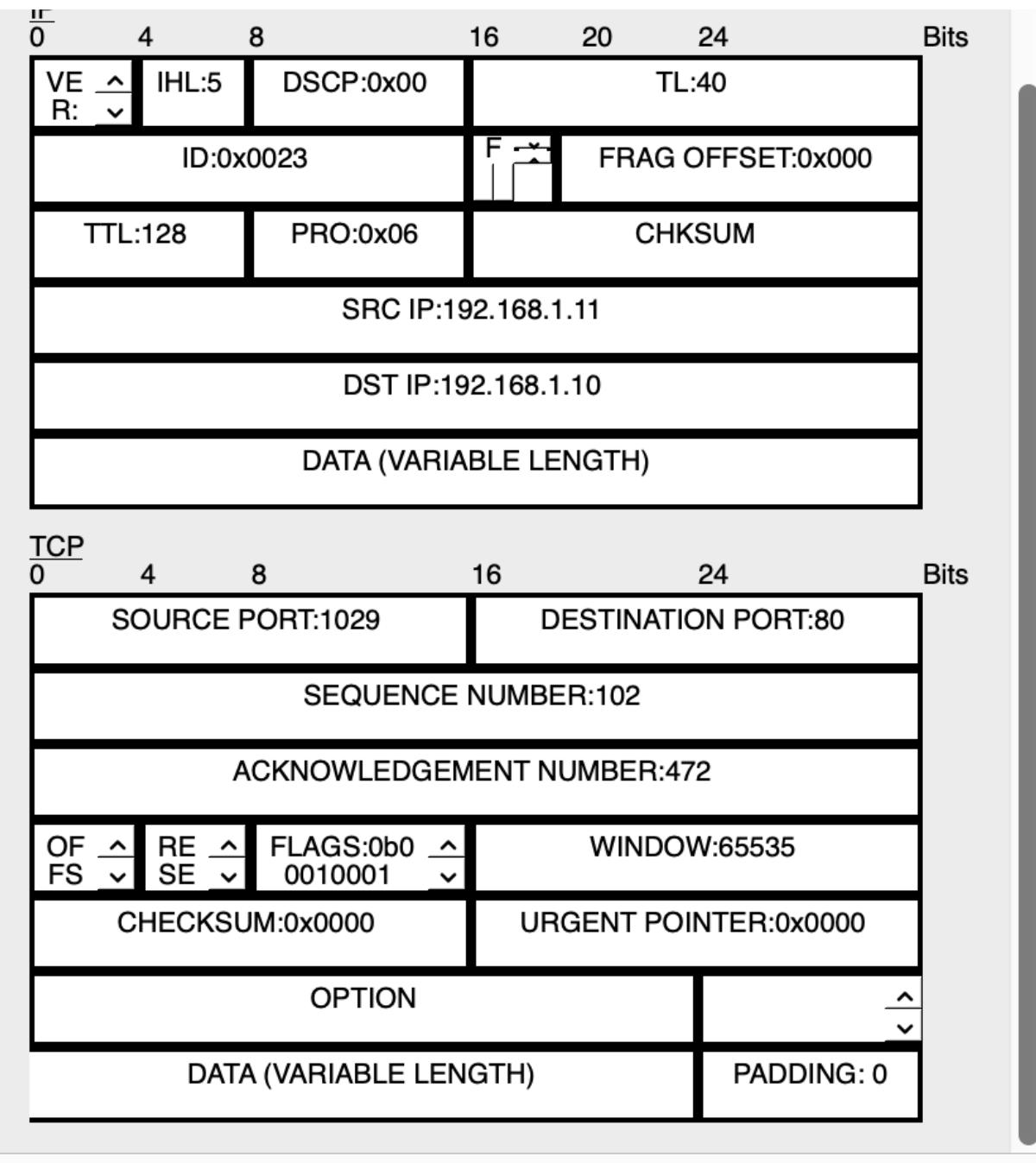
**Challenge Me**    << Previous Layer    Next Layer >>

PDU Information at Device: PC0

**OSI Model      Outbound PDU Details**

PDU Formats

SRC ADD	^	I	^	DATA (VA)	^	00.0000	^
R:0002.17	▼	Y	▼	RIABLE L	▼	00000	▼
ID							



## Analysis

- TCP uses sequence numbers to track byte order and ensure reliable delivery.
- The client acknowledges the last byte received + 1.
- This confirms successful receipt and readiness for the next segment.

## Part 4: Breaking the Network (TCP Retransmission)

### Task

- Observe how TCP handles packet loss by forcing a failure in the network.

## Steps

1. Reset Simulation completely.
2. Set filters to allow TCP packets.
3. On PC1, open the browser and request the website again (Server0).
4. Wait until the 3-Way Handshake is complete.
5. Before the HTTP packet returns, use the Delete Tool (X) to cut the cable between Server0 and Switch0.
6. Continue clicking Capture/Forward to observe events.

## Observations

- PC1 waits for the HTTP response but does not receive it.
- After a timeout, TCP automatically retransmits the packet.
- This shows TCP's reliability mechanism: it detects loss and resends data.

## Analysis

- TCP Retransmission ensures reliable delivery by resending lost packets.
- If this were UDP (e.g., live video stream):
  - The packet would not be resent.
  - UDP is connectionless and does not guarantee delivery.
  - Lost packets simply result in missing frames or data.

## Key Takeaway

- TCP sacrifices speed for reliability (resends lost data).
- UDP sacrifices reliability for speed (no retransmission).

# Part 5: The Pipeline Race

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## Scenario

- Link speed: **1 Gbps**
- Packet Transmission Time: **0.008 ms**
- Round Trip Time (RTT): **30 ms**
- Formula for Utilization:

$$U = (N \times \text{Transmission Time}) / (\text{RTT} + \text{Transmission Time})$$

## Questions & Calculations

### Q2 (Stop-and-Wait, N=1):

$$U = (1 \times 0.008) / (30 + 0.008) \approx 0.000266 \text{ (or } 0.0266\%)$$

### Q3 (Pipelining, N=3):

$$U = (3 \times 0.008) / (30 + 0.008) \approx 0.000799 \text{ (or } 0.0799\%)$$

#### **Q4 (Improvement Factor):**

$$\text{Factor} = U(N=3) / U(N=1) = 0.000799 / 0.000266 \approx 3.0$$

#### Results Table

Scenario	N	Utilization (U)	Percentage
Stop-and-Wait	1	0.000266	0.0266%
Pipelining	3	0.000799	0.0799%
Improvement Factor	-	~3x	3 times better

#### Analysis

- **Stop-and-Wait** uses the link inefficiently because the sender must wait for every ACK before transmitting the next packet.
- **Pipelining** sends several packets at once, which boosts link utilization.
- With even 3 packets in the pipeline, the utilization becomes roughly three times higher.

## Part 6: Flow Control & Teardown

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#### Steps Performed

1. Reconnect cables between **Server0** and **Switch0**.
2. Load the webpage successfully from **PC1**.
3. Opened the **last TCP packet** received by PC1.
4. Found the **Window Size** field in the TCP header.
5. Close the browser.
6. Observe the **FIN packets** exchanged during connection termination.

#### Key Findings

1. Flow Control (Window Size)
  - **The Window Size** tells the server how many bytes the PC is currently able to receive.
  - Example:
    - If the window size is 4096, PC1 is saying:  
"My buffer has room for 4096 more bytes, send only this much before waiting."
  - This prevents the sender from overwhelming the receiver.
2. Connection Teardown (4-Step TCP Closing) TCP closes connections gracefully using a four-step FIN handshake:
3. PC1 → Server0: FIN

4. Server0 → PC1: ACK
5. Server0 → PC1: FIN
6. PC1 → Server0: ACK

This ensures both directions of communication are properly closed.

## Analysis

- **Flow Control:** protects the receiver from buffer overflow by regulating how much data can be sent.
  - **Teardown:** ensures both sides finish communicating cleanly, preventing data loss.
  - Unlike UDP (no formal closing), TCP guarantees an orderly shutdown.
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