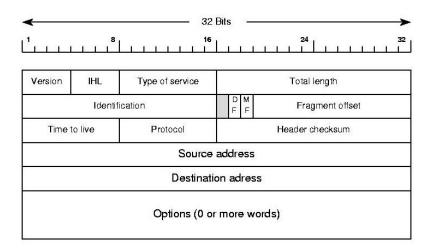
CS 349 Computer Networks Lab - Assignment 2 Rohit Pant (160101049)

- Application Online Game
- Game Played <u>www.addictinggames.com/puzzle-games/ball-in-the-hole-game.jsp</u>
- PC was connected to DigitalOcean VPN (Bangalore) while taking 2 of the readings.
- Traces https://drive.google.com/open?id=19IYJzjnTV50p89y0EsbGX3uBNBELF4vg

Ans 1. The various protocols used by the online game website for various layers were - Link Layer (Ethernet) - Ethernet frame starts with 7-Bytes Preamble. This is pattern of alternative 0's and 1's which indicates starting of the frame and allow sender and receiver to establish bit synchronization. SFD is a 1-Byte field which is always set to 10101011 and signifies the start of frame. The destination and source address are 6-byte fields which contains the MAC address of the source machine and the machine for which the message is destined. Length gives the length of the frame. Data field contains the actual data which is also known as Payload. The Frame Check Sequence field contains 32-bits hash code of data generated over the remaining fields. If the checksum computed by destination is not same as sent checksum value, data received is corrupted.

7 byte	1 byte	6 byte	6 byte	2 byte	46 to 1500 byte	4 byte
Preamble	Start Frame Delimiter	Destination Address	Source Address	Length	Data	Frame Check Sequence (CRC)

<u>Network Layer (Internet Protocol)</u> - <u>Version</u> gives the version number of the IP packet header. <u>Internet Header Length (IHL)</u> specifies the length of the IP packet header in 32 bit words. The <u>Type of Service</u>

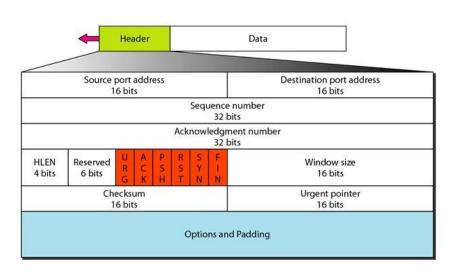


field is is often ignored by current routers but is meant to allow traffic to be prioritised. Total length is the length of the entire datagram including header and data. The <u>Identification</u> field is used to identify the original IP packet in case of fragmentation during the transmission. If an IP Packet is too large to handle, the 3-bit <u>Flag</u> field tells if it can be fragmented or not. Fragment offset tells the exact position of the fragment in the original IP Packet. <u>Time to Live or TTL</u> is a value set, which tells the network how many

routers (hops) this packet can cross. At each hop, its value is decremented by one and when the value reaches zero, the packet is discarded. The <u>Protocol</u> field identifies the transport-layer protocol which will interpret the Data section. The <u>Header Checksum</u> field is used to keep checksum value of entire header which is then used to check if the packet is received error-free. The <u>source and destination address</u> are 32-bit addresses of the sender (or source) and the receiver (or destination) of the packet. <u>Options</u> is an

optional field, which is used if the value of IHL is greater than 5 and can be used for security or debugging.

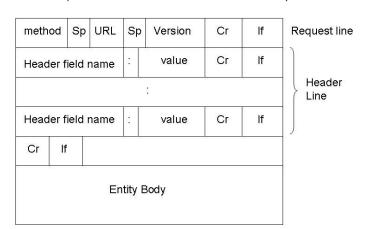
<u>Transport Layer (TCP)</u>- The <u>Source and Destination port address</u> give the source and destination port numbers. The <u>sequence number</u> of the first data octet in this segment is the next field. The acknowledgment field contains the value of the next sequence number the sender of the segment is



expecting to receive. HLEN stores the total size of a TCP header in multiples of four bytes. The Reserved field is reserved for future use and must be zero. Window size is the number of data octets beginning with the one indicated in the acknowledgment field which the sender of this segment is willing to accept. The checksum field is the 16 bit one's complement of the one's complement sum of all 16 bit

words in the header and text. The <u>urgent pointer</u> points to the sequence number of the octet following the urgent data. <u>Optional</u> TCP data can be used to include support for special acknowledgment and window scaling algorithms.

<u>Application Layer (HTTP)</u> - An <u>HTTP method</u>, a verb (like GET, PUT or POST) or a noun (like HEAD or OPTIONS), that describes the action to be performed. The request target, usually a <u>URL</u>, or the absolute



path of the protocol, port, and domain are usually characterized by the request context. The <u>HTTP version</u>, which defines the structure of the remaining message. There are numerous <u>request headers</u> available. They can be divided in several groups: <u>General headers</u>, like Via, apply to the message as a whole. Request headers, like User-Agent, Accept-Type (accepted body encoding), modify the request by specifying it further (like Accept-Language), by giving context (like Referer), or by conditionally restricting it (like

If-None) <u>Entity headers</u>, like Content-Length which apply to the body of the request. Obviously, there is no such header transmitted if there is no body in the request. <u>Bodies</u> can be broadly divided into two categories: <u>Single-resource bodies</u>, consisting of one single file, defined by the two headers: Content-Type and Content-Length. <u>Multiple-resource bodies</u>, consisting of a multipart body, each containing a different bit of information. This is typically associated with HTML Forms.

Ans 2. Several of the fields have been explained in the previous answer, so I will explain only the additional ones here.

Ethernet -

The values of the <u>source and destination MAC addresses</u> can be seen. Source and Destination both are <u>unicast</u> here meaning point-to-point communication. The type of upper layer protocol used is <u>IPv4</u>.

IPv4 -

Version is IPv4, thus leading 4 bits are 0100. Header length is 20 bytes while total datagram size is 864

```
bytes. Differentiated Services
Internet Protocol Version 4, Src: 10.19.1.5, Dst: 202.141.80.20
                                                                      Code Point (DSCP) is a packet
    0100 .... = Version: 4
                                                                      header value that can be used to
    .... 0101 = Header Length: 20 bytes (5)
   Differentiated Services
                                                                      indicate the level of service
    Total Length: 864
                                                                      requested for traffic, such as high
    Identification: 0xadbc (44476)
  ▶ Flags: 0x4000, Don't fragment
                                                                      priority or best effort delivery. ECN
    Time to live: 64
                                                                      is Explicit Congestion Notification.
    Protocol: TCP (6)
    Header checksum: 0x6422 [validation disabled]
                                                                      TTL is 64. Don't fragment flag is
    [Header checksum status: Unverified]
                                                                      set due to which a router which
    Source: 10.19.1.5
    Destination: 202.141.80.20
                                                                      normally would fragment a
```

packet larger than MTU (and potentially deliver it out of order), instead will drop the packet. Upper layer protocol is <u>TCP</u>. <u>Checksum</u> is value 0x6422. <u>Source and destination IP addresses</u> are also specified.

TCP -

The TCP packet contains the values of the <u>source and destination ports</u>, the sequence and <u>acknowledgement</u> and <u>header length</u> (which is 32 bytes in this case). The <u>PSH and ACK flags</u> are set.

```
Transmission Control Protocol, Src Port: 58150, Dst Port: 3128, Seq: 1, Ack: 1, Len: 812
Source Port: 58150
Destination Port: 3128
[Stream index: 33]
[TCP Segment Len: 812]
Sequence number: 813
[Next sequence number: 813 (relative sequence number)
[Next sequence number: 1 (relative ack number)
Acknowledgment number: 1 (relative ack number)
1000 ... = Header Length: 32 bytes (8)
Flags: 0x018 (PSH, ACK)
Window size value: 229
[Calculated window size: 29312]
[Window size scaling factor: 128]
Checksum: 0x018c [unverified]
Urgent pointer: 0

Options: (12 bytes), No-Operation (NOP), No-Operation (NOP), Timestamps
FSEQ/ACK analysis]
TCP pavload (812 bytes)
```

The Push flag tells the receiver's network stack to "push" the data straight to the receiving socket, and not to wait for any more packets before doing so. The ACK flag, which stands for "Acknowledgment", is used to acknowledge the successful receipt of a

packet. <u>Checksum</u> value is 0x018c. The <u>window size scaling factor</u> shows the number of leftward bit shifts that should be used for an advertised window size. The <u>urgent pointer</u> is not set.

HTTP -

```
Hypertext Transfer Protocol

> GET http://www.addictinggames.com/ HTTP/1.1\r\n
Host: www.addictinggames.com/ HTTP/1.1\r\n
Host: www.addictinggames.com/r\n
Proxy-Connection: keep-alive\r\n
Proxy-Connection: keep-alive\r\n
Cache-Control: no-cache\r\n
> Proxy-Authorization: Basic cm9oxXqucGFudDpUdYYIXXVuSg==\r\n
Upgrade: Insecure-Requests: 1\r\n
Upgrade: Insecure-Requests: 1\r\n
User-Agent: Mozilla/5.9 (X11; Linux x86_64) AppleWebKit/537.36 (KHTML, like Gecko) Chrome/70.0.3538.77 Safari/537.36\r\n
Accept: text/html, application/xhml+xml, application/xml;q=0.9, image/webp, image/apng, */*;q=0.8\r\n
Accept: Language: en-Us, en;q=0.9\r\n
Accept: Language: en-Us, en;q=0.9\r\n
Accept: Language: en-Us, en;q=0.9\r\n
Full request URI: http://www.addictinggames.com/]
[Full request URI: http://www.addictinggames.com/]
[HTTP request U711]
[Response in frame: 7781]
[Response in frame: 1781]
[Response in frame: 1781]
[Response in frame: 18053]
```

The request is a <u>GET request</u> to addictinggames.com using <u>HTTP version 1.1</u>. <u>Proxy connection Keep Alive</u> instructs the server to maintain the TCP connection even after sending response. <u>Cache-Control</u> is used to specify directives that must be obeyed by all caching mechanisms along the request-response chain. <u>Upgrade-Insecure-Requests</u> tells the server that the client would prefer redirection to HTTPS. <u>Pragma</u> is an implementation - specific field that may have various effects anywhere along the request-response chain. <u>Proxy Authorization</u> is added as this reading was taken while using IITG Proxy. The <u>User-Agent</u> request header contains a characteristic string that allows the network protocol peers to identify the application type. <u>Accept</u> lists the acceptable response data types. <u>Accept-encoding</u> and <u>Accept-language</u> give the content encoding, usually a compression algorithm, and the language that the client can understand. A cookie for the website containing information like recent game played and device id is also present.

Ans 3.

356 181.867703725 10.8.0.2	1.0.0.1	DNS	79 Standard query 0xc4eb A www.addictinggames.com OPT
357 181.949823106 1.0.0.1	10.8.0.2	DNS	111 Standard query response 0xc4eb A www.addictinggames.com A 104.25.168.36 A 104.25.167.36 OPT
358 181.951582775 10.8.0.2	104.25.168.36	TCP	60 44540 → 80 [SYN] Seq=0 Win=29200 Len=0 MSS=1460 SACK_PERM=1 TSval=2729824201 TSecr=0 WS=128
359 182.064137096 104.25.168.36	10.8.0.2	TCP	52 80 → 44540 [SYN, ACK] Seq=0 Ack=1 Win=29200 Len=0 MSS=1356 SACK_PERM=1 WS=1024
360 182.064201305 10.8.0.2	104.25.168.36	TCP	40 44540 → 80 [ACK] Seq=1 Ack=1 Win=29312 Len=0
361 182.064571677 10.8.0.2	104.25.168.36	HTTP	733 GET / HTTP/1.1
362 182.270889904 104.25.168.36	10.8.0.2	TCP	40 80 → 44540 [ACK] Seq=1 Ack=694 Win=30720 Len=0

The following exchange of messages (as can be seen in the picture) was observed on visiting the online games website addictinggames.com -

- 1. A DNS query was sent to get the IP Address corresponding to the domain www.addictinggames.com.
- 2. A TCP connection is initiated by performing a 3-way handshake with the destination server.
- 3. An HTTP GET request is sent to get the web page and other attachments.
- 4. The requested data is sent by the server over the established TCP connection.
- 5. Connection termination on closing browser by sending FIN TCP packets.

3-way handshake -

<u>Step 1 (SYN)</u>: In the first step, client wants to establish a connection with server, so it sends a segment with SYN (Synchronize Sequence Number) which informs server that client is likely to start communication and with what sequence number it starts its' segments.

<u>Step 2 (SYN + ACK)</u>: Server responds to the client request with SYN-ACK signal bits set. Acknowledgement(ACK) signifies the response of segment it received and SYN signifies with what sequence number it is likely to start its' segments..

<u>Step 3 (ACK)</u>: In the final part client acknowledges the response of server and they both establish a reliable connection with which they will start the actual data transfer.

Data Transfer - The client using a GET request requests the HTML content After that data exchange happens using TCP protocol with the use of various ACKs and [PSH,ACK]s. TCP's push capability

accomplishes two things: The sending application informs TCP that data should be sent immediately. The PSH flag in the TCP header informs the receiving host that the data should be pushed up to the receiving application immediately. So the server tells the client at various intervals that it has no more data to send and requests an acknowledgement immediately to which the client also replies with an ACK.

After the game has loaded, no live exchange of data occurs. Thus any actions like completing a level, pausing or restarting the game and muting the audio generates no data packets. This is due to the fact that the game I choose was quite basic and was controlled by a single Javascript file. Thus once all the resources are received from the server, no further communication with the server is required. The client only sends TCP Keep Alive packets so that the server maintains the connection.

TOTAL OF COOPTAINTS TOATTO TO . TO . TO . TO . TO	10.0.0.2	101	TO ITO KOOP ALIVE NOKE OF COLID [MOK] DEGILIZATE NOK-LITER WITH-OUTTO LONGO
20169 127.035997773 10.8.0.2	104.25.167.36	TCP	40 [TCP Keep-Alive] 59270 → 80 [ACK] Seq=11569 Ack=272519 Win=372608 Len=0
20196 127.176597541 104.25.167.36	10.8.0.2	TCP	40 [TCP Keep-Alive ACK] 80 → 59270 [ACK] Seq=272519 Ack=11570 Win=53248 Len=0
22748 173.980021109 10.8.0.2	104.25.167.36	TCP	40 [TCP Keep-Alive] 59270 → 80 [ACK] Seq=11569 Ack=272519 Win=372608 Len=0
22755 174.096236554 104.25.167.36	10.8.0.2	TCP	40 [TCP Keep-Alive ACK] 80 → 59270 [ACK] Seq=272519 Ack=11570 Win=53248 Len=0

Connection termination -

25941 214.800556931 10.8.0.2	104.25.167.36	TCP	40 59270 → 80 [FIN, ACK] Seq=11570 Ack=272519 Win=372608 Len=0
26088 214.906539154 104.25.167.36	10.8.0.2	TCP	40 80 → 59270 [FIN, ACK] Seq=272519 Ack=11571 Win=53248 Len=0
26089 214.906548053 10.8.0.2	104.25.167.36	TCP	40 59270 → 80 [ACK] Seq=11571 Ack=272520 Win=372608 Len=0

<u>Step 1 (FIN From Client)</u> – Suppose that the client application decides it wants to close the connection. (Note that the server could also choose to close the connection). This causes the client send a TCP segment with the **FIN** bit set to **1** to server and to enter the

<u>Step 2 (ACK From Server)</u> – When Server received FIN bit segment from the client, server immediately sends acknowledgement (ACK) segment to the client.

<u>Step 3 (Client waiting)</u> – The client waits for a TCP segment from the server with an acknowledgment. When it receives this segment the client waits for another segment from the server with the FIN bit set to 1.

<u>Step 4 (FIN from Server)</u> – Server sends FIN bit segment to the Sender(Client) after some time when Server send the ACK segment (because of some closing process in the Server).

<u>Step 5 (ACK from Client)</u> – When Client receive FIN bit segment from the Server, the client acknowledges the server's segment. The connection formally closes and all resources on the client side (including port numbers and buffer data) are released.

Ans 4. HTTP (Hypertext Transfer Protocol) is the set of rules for transferring files (text, graphic images, sound, video, and other multimedia files) on the World Wide Web. It is based on the client-server paradigm and uses requests and responses for data transfer. As can be seen in the traces, our website uses HTTP version 1.1. It uses it to deliver the website's HTML code, the game icons, javascript and audio to the client.

Transmission Control Protocol (TCP) – By looking at the traces we infer that TCP is responsible for the Transport Layer. Due to the 3-way handshake approach used it ensures that the receiver is ready to receive the resource. If a packet is lost in transit then a retransmission is requested using the system of acknowledgement and sequence numbers. Thus inorder and reliable data transfer is guaranteed. Connection termination too takes place with a sequence of 3 packets in our case. Generally games prefer to use UDP as a faster response is required in a live game. But here very no data is being communicated to the server as all the game resources are loaded initially. Thus TCP seems more viable as it guarantees packet delivery.

IPv4 - The IP has the job of delivering packets using the IP headers from the source to the destination.

Existing in the network layer, IPv4 connection is a hop to hop connection.

Ethernet - Ethernet is the most widely used way of connecting computers together in a local area network or LAN. Details regarding the MAC address of the source and destination are found in Ethernet header. Ethernet lying in data link layer is also responsible for error detection and correction along with flow control.

Ans 5.

	Sample 1	Sample 2	Sample 3
Time	4 P.M	11 PM	5 AM
Network	LAN (Lab)	WIFI through VPN (Room)	LAN (Room)
Throughput	29000	8591	16000
Average RTT	12 ms	2 ms	1 ms
Avg. Packet Size	634 bytes	655 bytes	297 bytes
Total Packets	23092	7305	24253
Packets Lost	22	14	7
No. of TCP Packets	16075	7225	7785
No. of UDP Packets	7014	0	1843
Number of Responses wrt one request send	1.43	1.19	1.25

Ans 6. Yes, the whole content is being sent from multiple locations. The website addictinggames.com (104.25.168.36) uses a Content Distribution Network hosted by Cloudfare having domain cdn.addictinggames.com (104.25.167.36). The base 'index.html' file and logo image were retrieved from addictinggames.com, while the other game media (including game icons, javascript) were retrieved from the CDN. The latter data is the more crucial both wrt size and frequency of use. So it is critical that it be delivered reliably and quickly. Requested content is cached (pre-saved) by a CDN's servers meaning end users will get that content by connecting to the nearest CDN server rather than waiting for their request to go directly to the origin server. This results in a significantly better performance for the client. Load balancing on multiple servers leads to reduced workload and reliability in case a server fails.