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Software Defined Networking (SDN) emulation with Mininet and OpenFlow

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Abstract

The world has been considered a digital society with the intervention of the Internet. These computer networks form a strong foundation in the manner we access information, communicate, and handle data, providing convenience in various aspects of life. This technology relies strongly on the operations of traditional IP networks and despite being widely adopted, they do get complex and hard to manage in terms of configuration according to predefined policies and reconfiguration in response to load, faults, and changes. On top of that, conventional networks are vertically integrated (the control plane and data plane are bundled together). Emerging technologies such as Software-Defined Networking (SDN) provide a convenient approach in handling computer networks. SDN brings a proposed paradigm which changes state of affairs by separating the network's control logic from the data plane. This approach enables centralisation of network control and ability to program the network improving network availability, scalability, and management. This report presents a practical use-cases of SDN. It starts by utilising concepts of SDN in both a wireless and wired network scenario, an Ad-Hoc Network for emergency scenario, and multi-cast video streaming. We measure and test out performance metrics such as throughput and jitter for communications under Session Initiation Protocol (SIP) and UDP(connectionless) communication in client-server scenario. All activities are carried out on a single machine running Linux with underlying applications such as Mininet, Open Network Operating System(ONOS) and Wireshark.

1 Introduction

Since the conception and realisation of the Advanced Research Projects Agency Networks(ARPANET) (Lukasik 2011), developments and innovations in computer networks have seen significant growth in size and requirements and navigating traditional network switches has become a challenge. In traditional networks, the control plane operation has a distributed infrastructure that requires protocols such as OSPF, STP, EIGRP, to operate independently on network devices. The forwarding decision which is the process of determining how to send a packet from one device (such as a router or switch) to another device within a network, based on certain criteria is performed by these network devices. This decision is made based on the information available in the routing table (for routers) or the switching table (MAC table) (for switches). It is agreed that the network devices connect but there is no centralised machine to manage or summarise the whole network (Haji et al. 2021). Also, the distributed control and transport protocols running inside routers and switches make it complex to express desired high-level network policies since network operators must configure each network device separately (Kreutz et al. (2015)). This main issue is rectified by SDN, which introduces the needed network control, flexibility, and programmability that network operators want.

1.1 Software-Defined Networking (SDN) and OpenFlow

Software-Defined Networking (SDN) is a computer technology that changes the limitations of current computer network infrastructure. SDN (Lara et al. 2014) decouples the control and data plane of a network leaving the switch with the simple function of forwarding packets based on a set of rules. By doing so, it breaks the vertical integration of conventional networks as the control logic is separated from the data plane i.e. the underlying switches and routers that forward the traffic (Kreutz et al. 2015). To make possible the communication between the controller and the network devices, OpenFlow (Lara et al. 2014) is used in conjunction with SDN. OpenFlow standardises the communication between the switches and the software-based controller in an SDN architecture. Researchers found it difficult to test out new ideas in current hardware because the source code of the software running on the network device cannot be modified making the infrastructure 'rigid'. As a result, a standardised protocol such as OpenFlow, to control the flow table of switches through software was provided by identifying common features in the flow tables of the ethernet switches (Lara et al. 2014).

The figures below show a comparison of the overview for traditional and SDN network architecture.

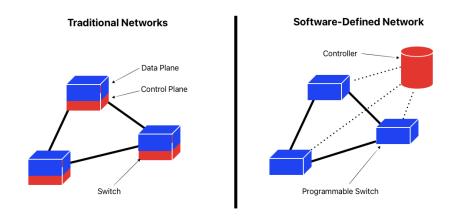


Figure 1: Architecture for Traditional and SDN networks

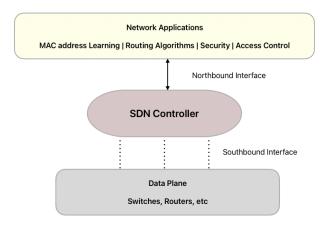


Figure 2: SDN architecture

The underlying network applications or packages which are the software for the logical operations within the network are installed and activated on the SDN controller. These applications are known as the Northbound APIs or interfaces as shown in Fig. 2. The Southbound interface consists of the protocols that facilitate communication between the remote SDN controller and the network devices, for example, the OpenFlow (Mokoena et al. 2023) protocol. Examples of such network applications can be STP, routing algorithms, and access control and examples of an SDN controller can be the ONOS controller. The prospect of such architecture is noticed in terms of ease of network management, programmability and openness to innovations and virtualisation it presents.

1.2 Mininet

For the scope of this paper, we aim to determine and investigate the convenience of SDN networks in a variety of network scenarios and requirements. To test out the efficiency, reliability, cost, and flexibility of the SDN approach in our network designs, the experiments are carried out on a network emulator known as Mininet. It is one of the many network simulation tools (another example is OMNETT++) that have been developed to virtualise and test network performance (Haji et al. 2021, Mokoena et al. 2023).

Mininet (de Oliveira et al. 2014) provides a system that allows rapid prototyping of large networks and creates scalable software-defined networks using a lightweight virtualisation mechanism. This alleviates the downside of conventional networking by providing a centralised view of the network and paves the way for more controllability in managing the way networks operate regardless of their size or their complexity.

Since research on this topic is still in progress, there are not many devices such as routers and switches that implement SDN functionalities, moreover, the existing ones are very expensive. Thus, in order to enable researchers to perform experiments and test novel features of this new paradigm in practice at a low financial cost, one solution is to use virtual network emulators such as Mininet. It creates SDN elements, customises them, shares them among other networks, and performs interactions (de Oliveira et al. 2014).

1.3 Emulation environment specifications

For this experiment, we utilise a microcomputer iMac with the following specifications: Processor 3 GHz 6-Core Intel Core i5, 16GB of ram, running the macOS 15.1, and VirtualBox Oracle VM version 7.1.2. In this microcomputer, under the management of VirtualBox, we installed the following guest operating systems: Mininet emulator version 2.0 on Linux operating system Ubuntu 64bits with 4GB of RAM; ONOS controller version 4.2.14.

2 Task 1 - WiFi Network Emulation

In this task, we emulate a wireless network designed for a floor in the new building. For this purpose, we emulate 3 stations and 5 access points. The stations may represent a smart hand-held device which can vary from a smartphone to a laptop, UE or any WiFi-compatible device. The stations carry a Class C private IP address of the same network. Access points are connected using a physical link facilitating a linear topology. The new building would create a minimalistic noise threshold of -91dBm. The access points are positioned strategically within the floor to make space for a signal dead zone (red spot) and 3 stations will be in a mobility state to emulate a real-life network scenario.

2.1 Wireless Communication

Briefly on wireless communication, it is a form of computer communication which is based on digital signals that are transmitted from one end and received at another end over the radio frequency spectrum. Medium Access Control or MAC (Murthy & Manoj 2004) is a mechanism devised to manage how these radio channels are accessed by wireless capable devices for wireless communication and has two main techniques; Contention-based & Contention-free MAC.

A Python script will contain the code for this configuration and upon completion, we start the network using Mininet on a Linux terminal and view the network emulation on the Mininet-GUI feature while having to observe access point association, full connectivity between all nodes in the network on the Mininet CLI, and briefly understand how communication channels are used in a wireless network.

The figures below show the outline of the floor-plan in context and position of APs and stations in the design.

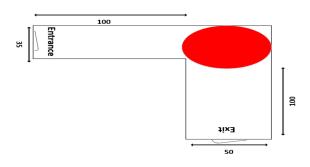


Figure 3: Floor-plan

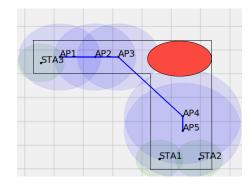


Figure 4: Floor-plan with network topology

2.2 Design and Configuration

Both figures show the outline of the floor plan in context with the dimensions included in Fig. 3 while Fig. 4 is a snapshot of the Mininet GUI with the positions of the stations and access points and the range of coverage of all devices. The 'red-spotted' zone as seen remains free of WiFi signals as required in the design. Table 1 contains the details of stations and access points which are used for the network configuration in a Python script. All three stations share the same network address and have full authenticated access to each access point.

The access points are positioned strategically and linked serially to ensure the required zones within the floor have strong WiFi frequency coverage and support full connectivity of the stations. To emulate a real networking scenario, mobility is included for the stations, details are shown in Table 2. Mobility attributes in terms of velocity or speed of motion for each station are specified. The mobility attributes are added to the Python code for the network configuration which is captured in the appendix section of this document.

DEVICE	MAC	IPv4	(x,y)	SSID	PASSWORD	RANGE	CHANNEL
STA1	00:00:00:00:00:10	192.168.50.11/24	15,115	n/a	n/a	20	n/a
STA2	00:00:00:00:00:11	192.168.50.12/24	20,130	n/a	n/a	20	n/a
STA3	00:00:00:00:00:12	192.168.50.13/24	140,10	n/a	n/a	20	n/a
AP1	00:00:00:00:00:00	n/a	30,117.5	AP1	n/a	35	1
AP2	00:00:00:00:00:01	n/a	60,117.5	AP2	n/a	35	1
AP3	00:00:00:00:00:02	n/a	80,117.5	AP3	n/a	35	1
AP4	00:00:00:00:00:03	n/a	135,55	AP4	n/a	50	1
AP5	00:00:00:00:00:04	n/a	135,40	AP5	n/a	50	1

Table 1: Details of stations and access points

DEVICE	START LOCATION	END LOCATION	START-STOP TIME	MOVING SPEED(min-max)
STA1	15,115	115,10	10s-20s	min_v=1, max_v=5
STA2	20,130	150,10	30s-60s	$\min_{v=5, \max_{v}=10}$
STA3	140,10	15,120	25s-60s	$\min_{v=2, \max_{v}=7}$

Table 2: Mobility information for stations

2.3 Results and Analysis

Commencing the test, we prepare the Python script for the network topology with all necessary libraries then run it with Mininet on the Linux terminal. The command sudo ./<pycode filename> is used to start the network using the configuration from the Python script on Mininet. Mininet creates the SDN elements such as the controller, stations, and access points and sets up the topology as described in the Python script.

The controller used in this task is a default one used by Mininet since it has not been specified. The access points are started and operated based on instructions from the controller. Fig. 5 & 6 show the Mininet GUI view of the topology prior to mobility and after mobility respectively. All stations are within strong WiFi signal coverage and successfully communicate with each other as shown in Fig. 8.

The ping -c 3 command was utilised to send out 3 Internet Control Message Protocol(ICMP) (Murthy & Manoj 2004) packets between specified stations to determine full connectivity within the network. This

uses a series of request & reply messages to establish communication between stations. The results show all 3 packets were transmitted and received successfully, with no packet loss, and took an average time of 2000 milliseconds for all communications which verifies the optimal performance of the network.

Prior mobility, we see that stations 1 & 2 would likely be associated with AP1 and station 3 with AP5 due to their close proximity to these APs but when in mobility state, the stations will lose association with access points, go through a handoff operation, and get associated with the access point in close proximity at the end of mobility state. To confirm this, we use the <station name> iwconfig command on the Mininet CLI to pull the wireless interface information for each station as illustrated in Fig. 7. This same activity could be carried out on each station directly when accessed with the xterm <station name> command on Mininet CLI.

At the end of this experiment, we experience the sort of flexibility SDN brings into wireless network configuration and management, cost of deployment, quick response time, and overall summary of the network compared to the conventional methods of networking.

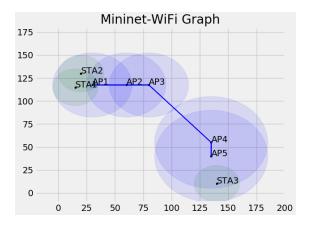


Figure 5: Prior mobility



Figure 7: Access points connected after mobility

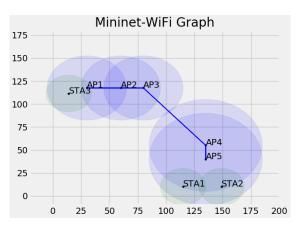


Figure 6: After mobility

```
mininet-wifi> STA1 ping STA2 -c 3
PING 192.168.50.12 (192.168.50.12) 56(84) bytes of data.
64 bytes from 192.168.50.12: icmp_seq=1 ttl=64 time=35.4 ms
64 bytes from 192.168.50.12: icmp_seq=2 ttl=64 time=10.6 ms
64 bytes from 192.168.50.12: icmp_seq=3 ttl=64 time=10.6 ms
64 bytes from 192.168.50.12: icmp_seq=3 ttl=64 time=67.6 ms
--- 192.168.50.12 ping statistics ---
3 packets transmitted, 3 received, 0% packet loss, time 2013ms
rtt min/avg/max/mdev = 10.621/37.906/67.666/23.354 ms
mininet-wifi> STA2 ping STA3 -c 3
PING 192.168.50.13 (192.168.50.13) 56(84) bytes of data.
64 bytes from 192.168.50.13: icmp_seq=1 ttl=64 time=95.3 ms
64 bytes from 192.168.50.13: icmp_seq=2 ttl=64 time=10.4 ms
64 bytes from 192.168.50.13: icmp_seq=3 ttl=64 time=9.43 ms
--- 192.168.50.13 ping statistics ---
3 packets transmitted, 3 received, 0% packet loss, time 2005ms
rtt min/avg/max/mdev = 9.430/38.417/95.370/40.274 ms
mininet-wifi> STA1 ping STA3 -c 3
PING 192.168.50.13 (192.168.50.13) 56(84) bytes of data.
64 bytes from 192.168.50.13: icmp_seq=1 ttl=64 time=102 ms
64 bytes from 192.168.50.13: icmp_seq=2 ttl=64 time=12.9 ms
64 bytes from 192.168.50.13: icmp_seq=2 ttl=64 time=19.47 ms
--- 192.168.50.13 ping statistics ---
3 packets transmitted, 3 received, 0% packet loss, time 2011ms
rtt min/avg/max/mdev = 9.477/41.814/102.984/43.277 ms
```

Figure 8: Full node connectivity

3 Task 2 - Adhoc Network Emulation

In this task, we emulate an AdHoc network scenario at an emergency gathering car park at the new building when response units communicate via AdHoc services. We evaluate some ad-hoc protocols to find which is best for the given situation. For this purpose, 3 stations are emulated and the ad-hoc protocols which will be tested are the 'batmand', 'batman_adv', and 'olsrd' protocols. ICMP streams are initiated between the stations closest to each other and 3 VoIP for each ad-hoc protocol tested. For analysis of connection and network performance, we use a network protocol analytic tool known as Wireshark to inspect and capture the traffic of a network in real time. It shows details about protocols and can be used to generate a graphical or visualised view of network traffic.

3.1 AdHoc Networks

AdHoc (Murthy & Manoj 2004) networks are quick-to-deploy and dynamic forms of networks that can be designed for a particular use case. It is a wireless network however individual nodes can act as hops for forwarding packets and can move randomly within the local network. AdHoc networks are less reliant on infrastructure and are efficient in setting up wireless communications in cases of emergencies or quick deployment.



Figure 9: Adhoc network with 3 stations

3.2 Design and Configuration

A Python script is used for the configuration of the network where all 3 stations are created and configured with a protocol to support communication over the AdHoc network. Fig. 9 depicts the floor plan of the new building and emergency gathering car park with 3 stations in close proximity. Table 3 contains the station names, IPv6 and MAC addresses, their positions, range of radio frequency coverage, and wireless configuration with details about those fields below;

- A_HEIGHT: the height of the antenna typically in physical distance(metres) between the antenna and the ground or reference point
- A_GAIN: the measure of how good an antenna converts input power into radio waves in a specific direction. Usually expressed in decibels (dB)
- SSID: Service Set Identifier, unique name used to identify a wireless local area network(WLAN)
- HT_CAP.: High Throughput Capability, relates to 802.11n/ac standards to support high data rate and better performance

NAME	IPv6	MAC	POSITION	RANGE	A_HEIGHT	A_GAIN	SSID	HT_CAP
STA1	2024::11	00:00:00:00:01:11	20,10,0	30	1	5	adhocUH	HT40+
STA2	2024::12	00:00:00:00:01:12	$45,\!10,\!0$	30	2	6	adhocUH	HT40+
STA3	2024::13	00:00:00:00:01:13	70,10,0	30	3	7	adhocUH	HT40+

Table 3: Adhoc stations configuration details

3.3 Results and Analysis

After completion of the script, the network is started on the Linux terminal with the sudo ./<python filename> command. For this experiment, we aim to establish a successful connection between the closest stations in our setup i.e. between STA1 and STA2 and test out a VoIP connection between the stations under three different adhoc protocols.

The Session Initiation Protocol(SIP) which is a protocol used to establish, modify, and terminate multimedia sessions over IP networks will be initiated on one station as a server and on the other as the client. The SIP commands are run on each station directly accessing them with xterm <station name> on Mininet CLI.

- on server(STA1): sipp -sn -uas
- \bullet on client(STA2): sipp -sn uac <STA1 IPv4> -timeout 120

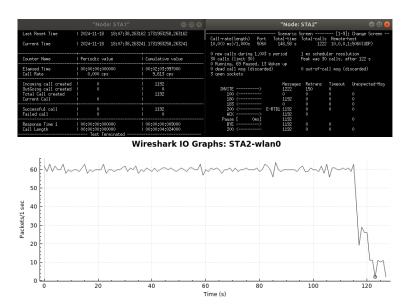


Figure 10: SIP connection & throughput under BATMAN_ADV protocol

The sipp -sn -uas command sets up a User Agent Server on STA1 to respond to incoming SIP requests such as an INVITE message with appropriate SIP responses such as 180 RINGING, 200 OK.

The sipp -sn uac <STA1 IPv4> -timeout 120 command sets up STA2 as a User Agent Client to send INVITE messages to initiate SIP calls and wait for response such as 180 RINGING and 200 OK.

We capture and inspect the communication on Wireshark by running another instance of 'xterm' on the client station on the Mininet CLI, then enter the command sudo wireshark to launch Wireshark. After it launches, the interface for the client station is selected to view the detailed packet information and the throughput graph is generated by clicking on the 'Statistics' tab in Wireshark and then selecting 'I/O Graph' from the dropdown list.

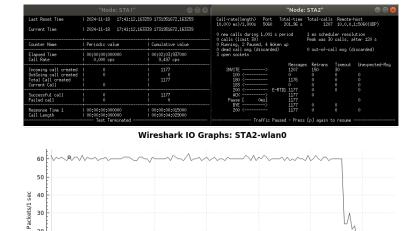


Figure 11: SIP connection & throughput under BATMAND protocol

The figures 10, 11 & 12 show the statistics from the SIP connection which lasts for 120 seconds on both server and client terminal CLI and the throughput graph from Wireshark.

Under the 'batman_adv' protocol, the connection statistics on the server(STA1) shows;

• Cumulative Call Rate of 9.613 calls per second

10

- 1192 Successful Calls created
- 0 Failed Calls

Under the 'batmand' protocol, the connection statistics on the server(STA1) shows;

- Cumulative Call Rate of 9.497 calls per second
- 1177 Successful Calls created
- 0 Failed Calls

Under the 'olsrd' protocol, the connection statistics on the server(STA1) shows;

- Cumulative Call Rate of 9.531 calls per second
- 1182 Successful Calls created
- 0 Failed Calls

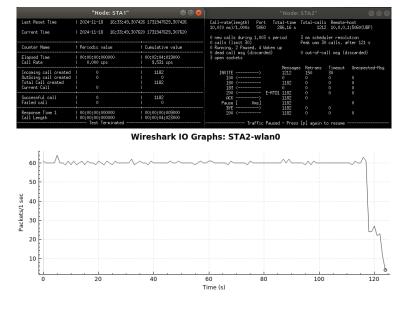


Figure 12: SIP connection & throughput under OLSRD protocol

At the end of this experiment, we have achieved a functioning ad-hoc network with successful VoIP connectivity under different ad-hoc protocols. Analysing the results from the statistics shows the best protocol to be the 'batman_adv' protocol since it achieved the highest call rate however, we notice slight differences among the 3 protocols tested in terms of network performance, connections maintained, and average bandwidth maintained during transmission period.

4 Task 3 - SDN Simulation

This task activity utilises the concept of Software Defined Networking in conjunction with OpenFlow on the local machine to manage our network. The experiment entails creating a network that connects an old building to a new building at the University of Hertfordshire. The task emulates 3 servers housed in the old building and 2 hosts in the new building. We utilise 5 switches for the underlying data forwarding and an SDN controller for the purpose of this activity. We also undertake a User Datagram Protocol(UDP) transmission between two nodes for a specified duration, port number, and bandwidth.

A UDP transmission is a type of communication protocol which unlike Transmission Control Protocol(TCP)(Conrad et al. 2012) the communication does not require the three-way handshake(SYN - SYN_ACK - ACK) to establish a secure connection.

- SYN sender sends synchronisation message to receiver
- SYN_ACK receiver sends synchronisation-acknowledgement message to sender
- ACK sender sends acknowledgment message to receiver to establish communication

The sender sends out packets without waiting for an acknowledgement from the recipient. This form of communication does check errors and may encounter loss of packets during transmission but in turn, achieves higher transmission speeds and supports transfer of large size data.

4.1 ONOS Controller

ONOS stands for Open Network Operating System. ONOS provides the control plane for a software-defined network (SDN), managing network components, such as switches and links, and running software programs or modules to provide communication services to end hosts and neighbouring networks. It was developed by the ONOS Project, which is led by a collaboration of various universities and industry partners, including Intel and Ericsson. ONOS controller helps to attain the advantages of SDN by achieving network programmability, high performance and scalability, support for OpenFlow protocols, support virtualisation of networks, fault-tolerant and high availability, and support for heterogeneity of network devices.

4.2 Design and Configuration

Table 4 contains the configuration information for hosts and servers in the network which is used in the Python script for the network configuration. After completing the Python script, the ONOS controller is built and started on a Linux terminal with the commands bazel build onos and runsdn. When the ONOS server is ready, we log in with onos localhost command from a new terminal where we have access to the controller and can activate some networking applications for the network to operate how we desire. These commands;

- app activate org.onosproject.openflow
 - app activate org.onosproject.fwd

are used to activate flows and reactive routing on the controller which is necessary for the transfer of packets across the network. Now, on a new terminal we run the network configuration file with the command; sudo mn --custom <python filename> --controller remote,ip=<0NOS controller IP> --topo <topo name> which specifies the controller to be used for the network.

We access the ONOS GUI at http://localhost:8181/onos/ui to view the topology of the network which is shown in Fig. 13. This is available only after all nodes can have full connectivity to each other, thus

servers and hosts can communicate successfully. The hosts are made visible on the ONOS GUI by pressing the "H" key on the keyboard.

NAME	IPv4	MAC ADDRESS
H1	192.170.50.11	00:00:00:00:15:98
H2	192.170.50.12	00:00:00:00:15:99
SERVER1	20.0.0.2	00:00:00:00:16:00
SERVER2	40.0.0.2	00:00:00:00:16:01
SERVER3	60.0.0.2	00:00:00:00:16:02

Table 4: Network configuration details

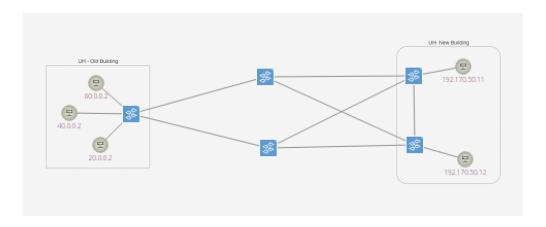


Figure 13: Network topology from ONOS GUI

As seen in the configuration details in Table 4, the hosts and servers are in different networks. Without any routing configured on the nodes, nodes will not be able to communicate with nodes in a different network. To solve this issue, we configured static routes on each node with information on the other network. The command below is a generic one of the routing configured on each node;

• ip route add <network address>/<netmask> dev <interface>

This can be carried out on the Mininet CLI, the ip route command preceded by the node name or directly on the node terminal CLI accessed with 'xterm' on Mininet. Once the route configurations are done, we can expect to have full connectivity in the network and proceed to the UDP transmission. To carry out this activity, a terminal CLI for both nodes(Server & Host1) is launched using the 'xterm' command on Mininet. The 'iperf' is a command which can be used to stress-test a network to determine the communication performance of the network.

On the server, we enter the command;

• iperf -s -u -p <port> -i 1: initiate the server to listen for UDP connections on port specified for every second

And on client, we enter command;

• iperf -c <server IP> -u -p <port> -t <time> -b <bandwidth> : start the UDP connection by the client to the server, at port, duration, and bandwidth specified

4.3 Results and Analysis

After we activate the network apps on the ONOS controller and start the network on Mininet, we check the connectivity of nodes in our network prior to configuring static routes on each node. Fig. 14 has images of the results from a 'pingall' command in Mininet.

```
mininet-wifi> pingall
*** Ping: testing ping reachability
H1 -> H2 X X X
H2 -> H1 X X
Srv1 -> X X X
Srv2 -> X X X
Srv3 -> X X X
$rv3 -> X X X
*** Results: 90% dropped (2/20 received)
```

```
nininet-wifi> pingall
*** Ping: testing ping reachability
H1 -> H2 Sv1 Sv2 Sv3
H2 -> H1 Sv1 Sv2 Sv3
Sv1 -> H1 H2 Sv2 Sv3
Sv2 -> H1 H2 Sv1 Sv3
Sv2 -> H1 H2 Sv1 Sv3
Sv3 -> H1 H2 Sv1 Sv2
*** Results: 0%_dropped (20/20 received)
```

Figure 14: Connectivity tests before & after reactive routing activation

As shown, 90% of packets dropped from testing connectivity prior to configuring the IP routes on nodes due to the nodes having no idea how to reach other networks outside their original network. Once the nodes have routing information to reach other networks, full connectivity is achieved as shown from 0% packets dropped.

Then commenced the UDP communication between Server1 & Host1 for 600 seconds, with a bandwidth of 100 megabytes per second, and on port 5656 after full connectivity of the network is established. The results from the UDP test in Fig. 15 showed successful connection between Server1 & Host1 with the following statistics;

- connection time of 600 seconds
- total jitter(time interval of packet arrival) of 0.161 milliseconds
- transferred total of 7.32 gigabytes of data
- maintained bandwidth of 105 megabytes per second
- 5349875 total datagrams sent and 250 datagrams received out-of-order

Figure 15: UDP connection between Server & Client

In general, UDP-based communications are faster compared to TCP-based connections. This is due to the lack of error checks and the acknowledgement technique used by TCP to establish security and less error in communications.

No.	Time	Source	Destination	Protocol	Lengtr Info
Г	3 0.350317428	192.170.50.11	20.0.0.2	UDP	1512 57081 → 5656 Len=1470
	4 0.350937563	192.170.50.11	20.0.0.2	UDP	1512 57081 → 5656 Len=1470
	5 0.350958881	192.170.50.11	20.0.0.2	UDP	1512 57081 → 5656 Len=1470
11	6 0.350969293	192.170.50.11	20.0.0.2	UDP	1512 57081 → 5656 Len=1470
	7 0.350980321	192.170.50.11	20.0.0.2	UDP	1512 57081 → 5656 Len=1470

Figure 16: Wireshark capture on H1-eth0 interface

This snippet from Wireshark gives details on how the UDP protocol works. The transfer is one-way(from Host1 to Server1) with no acknowledgement from Server1. This allows for high-rate data transfer and is beneficial in applications where data transfer is needed in real-time.

SDN supports the building and testing of such network scenarios with no reliance on infrastructure or hardware equipment, reduces expenses, and provides faster and more convenient control of the entire network.

5 Task 4 - Multicast Video Stream Service

The final experiment emulates a local area network(LAN) where we test the multicast video streaming over the network for the new building to facilitate learning. Multicasting (Toh 1997) in networking is a type of communication which is also known as one-to-many communication. In a connectionless network, multicast basically involves replicating packets and distributing them to multiple receivers. A multicast group refers to a set of senders and receivers who are involved in a multi-party communication session. It is always advantageous to use multicast rather than multiple unicast communication since it provides an efficient means of communication among selected nodes within a large network (Murthy & Manoj 2004).

5.1 Design and Configuration

The network to be tested consists of 2 switches, a video source acting as the server, and three host devices which could be any ethernet-enabled device. Table 5 contains the name and IP address for configuration on host devices. The host devices will have the same network address and the configuration to allow for multicast communication is activated on video source(VS), host1(H1), and host2(H2) which creates a multicast group for these devices.

DEVICE NAME	IP ADDRESS
H1	50.10.10.10
H2	50.10.10.11
Н3	50.10.10.12
VS	50.10.10.20

Table 5: Network configuration details

```
oot@ubuntu:~/Desktop/git/first-repo/wireless# echo 0 > /proc/sys/net/ipv4/icmp
_echo_ignore_broadcasts
oot@ubuntu:~/Desktop/git/first-repo/wireless# ip route add 224.0.0.0/8 dev VS-
eth0
root@ubuntu:~/Desktop/git/first-repo/wireless# ip route
50.0.0.0/8 dev VS-eth0 proto kernel scope link src 50.10.10.20
224.0.0.0/8 dev VS-eth0 scope link
                                                                          8 0 6
oot@ubuntu:~/Desktop/git/first-repo/wireless# echo 0 > /proc/sys/net/ipv4/icmp
_echo_ignore_broadcasts
-oot@ubuntu:~/Desktop/git/first-repo/wireless# ip route add 224.0.0.0/8 dev H1
eth0
oot@ubuntu:~/Desktop/git/first-repo/wireless# ip route
0.0.0.0/8 dev H1-eth0 proto kernel scope link src 50.10.10.10
24.0.0.0/8 dev H1-eth0 scope link
oot@ubuntu:~/Desktop/git/first-repo/wireless# echo 0 > /proc/sys/net/ipv4/icmp
_echo_ignore_broadcasts
oot@ubuntu:~/Desktop/git/first-repo/wireless# ip route add 224.0.0.0/8 dev H2-
root@ubuntu:~/Desktop/git/first-repo/wireless# ip route
50.0.0.0/8 dev H2-eth0 proto kernel scope link src 50.1<u>0.10.11</u>
24.0.0.0/8 dev H2-eth0 scope link
```

Figure 17: Activating multicast for required nodes

Fig. 16 shows the steps in activating the multicast on the required nodes. This can be done directly on the Mininet or at the node terminal when accessed with 'xterm' command on Mininet. The commands;

- echo 0 > /proc/sys/net/ipv4/icmp_ignore_broadcasts: allow node response to Internet Control Message Protocol(ICMP) echo request(ping request) broadcasted to every host on the network
- ip route add 224.0.0.0/8 dev <interface>: configures route that ensures multicast traffic for 224.0.0.0/8 address range is sent through the interface specified

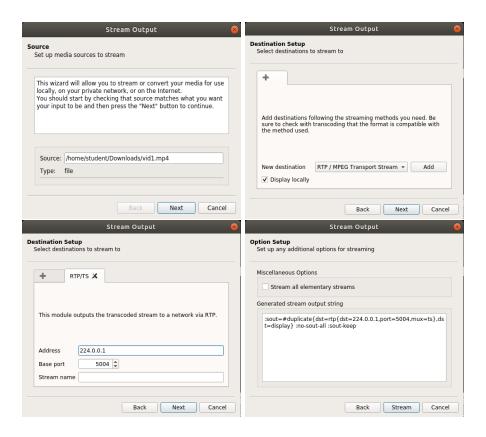


Figure 18: Multicast streaming setup at video source

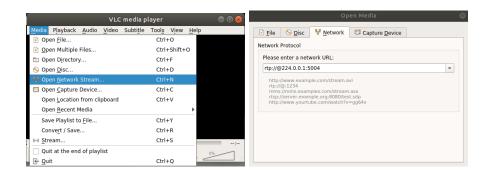


Figure 19: Multicast video connection at multicast host

After routing configuration, the video stream output is initiated from the video source on VLC through the terminal CLI of the video source, the command vlc-wrapper launches VLC where we select the video file,

transfer protocol, and provide the network to which the streaming will be sent across. On the host terminal CLI, VLC is launched with the same command and the multicast video stream is connected using the URL rtp://@224.0.0.1:5004. Fig. 17 and 18 show these steps taken out respectively.

5.2 Results and Analysis

Once all configurations are completed, the video as seen in Fig. 20 has been streamed from the source across the multicast network where the two hosts(1 & 2) can have the video output on the VLC player.

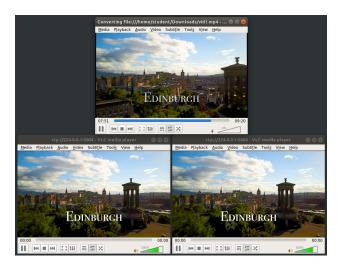


Figure 20: Multicast video stream from source to hosts

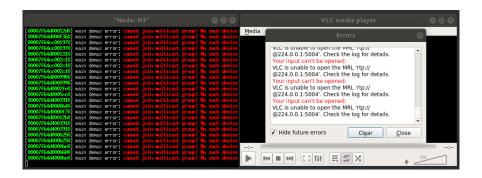


Figure 21: Multicast video stream at host 3

The third host who is not part of the multicast group cannot stream the video as shown in Fig. 21. This confirms that the streaming is a multicast stream and only host devices within the multicast group can get access to the video stream. Fig. 22 shows a snippet of a Wireshark capture where we can see from the Ethernet section of the packet that the destination MAC address is of a multicast type(IPv4mcast_01).

During this process, we encounter a protocol known as the Real-Time Transfer Protocol(RTP) which is used for transmission of media files(audio or graphics) over an IP network. Since it is also a form of UDP connection it does not require acknowledgement in setting up the communication which makes it able to support voice and video communications over IP and online gaming.

```
Time
                                                     Destination
                                                                                Protocol Length Info
                          Source
  7484 53.404170051
                          50.10.10.20
                                                     224.0.0.1
                                                                                UDP
                                                                                            1370 55011 → 5004 Len=1328
  7485 53.430292161
                          50.10.10.20
                                                     224.0.0.1
                                                                                UDP
                                                                                            1370 55011 → 5004 Len=1328
  7486 53.441390481
                                                     224.0.0.1
                                                                                UDP
                                                                                            1370 55011 → 5004 Len=1328
                          50.10.10.20
Frame 3110: 1370 bytes on wire (10960 bits), 1370 bytes captured (10960 bits) on interface 0 Ethernet II, Src: b6:fb:57:77:ea:3c (b6:fb:57:77:ea:3c), Dst: IPv4mcast_01 (01:00:5e:00:00:01)
  Destination: IPv4mcast_01 (01:00:5e:00:00:01)
Source: b6:fb:57:77:ea:3c (b6:fb:57:77:ea:3c)
    Type: IPv4 (0x0800)
Internet Protocol Version 4, Src: 50.10.10.20, Dst: 224.0.0.1
User Datagram Protocol, Src Port: 55011, Dst Port: 5004
Data (1328 bytes)
```

Figure 22: Packet details captured on Wireshark

6 Conclusion

At the end of the experiments, it is confirmed that SDN provides greater advantages to network developers or operators in the manner of network control and also a summary of the network compared to the traditional way of computer networking. There are also vast improvements in the deployment of networks, configuration, fault tolerance, and scalability as demonstrated in the tasks.

The table below shows a comparison of some features between traditional and software-defined networking.

	Feature	Traditional Networking	Software-Defined Networking
Ì	Control plane	Distributed across devices	Centralised & managed by SDN controller
	Network Configuration	Manual configuration	Configuration via software and APIs
	Agility & Flexibility	Difficult to adapt quickly	Quick reconfiguration
	Cost Efficiency	High cost of infrastructure	Low cost due to virtualisation
	Programmability	Difficult to program	Easy to program via SDN controller

To conclude, we point out the achievements of software-defined networking in the emulation of the wireless and ad-hoc network, in testing the Voice over IP(VoIP) communication, and the multicast streaming service over a network. The configuration and deployment of the networks require less time and effort, are at a very low cost, and can be easily managed due to the agility and flexibility offered. Researchers and network designers can test out new ideas and innovations while network concepts can be taught in an education setting in a very efficient and convenient way.

7 References

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8 Appendix

a Task 1 code

```
import sys
from mininet.node import Controller
from mininet.log import setLogLevel, info
from mn_wifi.cli import CLI
from mn_wifi.net import Mininet_wifi
def topology():
    info("**creating network**\n")
    net = Mininet_wifi()
    info("**creating nodes**\n")
    ap1 = net.addAccessPoint('ap1', mac='00:00:00:00:00', ssid='AP1', mode
       ='g', channel='1', position='30,117.5', band='5', range=35)
    ap2 = net.addAccessPoint('ap2', mac='00:00:00:00:00:01', ssid='AP2', mode
       ='g', channel='1', position='60,117.5', band='5', range=35)
    ap3 = net.addAccessPoint('ap3', mac='00:00:00:00:00:02', ssid='AP3', mode
       ='g', channel='1', position='80,117.5', band='5', range=35)
    ap4 = net.addAccessPoint('ap4', mac='00:00:00:00:00:03', ssid='AP4', mode
       ='g', channel='1', position='135,55', band='5', range=50)
    ap5 = net.addAccessPoint('ap5', mac='00:00:00:00:00:04', ssid='AP5', mode
       ='g', channel='1', position='135,40', band='5', range=50)
    sta1 = net.addStation('iphone5', mac='00:00:00:00:00:10', iphone5')
       = '192.168.50.11/24', position = '15,115', range = 20, min_v=1, max_v=5)
    sta2 = net.addStation('ipad', mac='00:00:00:00:00:11', ip
       = '192.168.50.12/24', position = '20,130', range=20, min_v=5, max_v=10)
    sta3 = net.addStation('macbook', mac='00:00:00:00:00:12', ip
       = 192.168.50.13/24, position = 140.10, range = 20, min_v=2, max_v=7
    c0 = net.addController('c0')
    net.setPropagationModel(model="logDistance", exp=5)
    info("**configuring wifi nodes and mobility**\n")
    net.configureWifiNodes()
    net.addLink(ap1, ap2)
    net.addLink(ap2, ap3)
    net.addLink(ap3, ap4)
    net.addLink(ap4, ap5)
    net.plotGraph(min_x=-20, min_y=-10, max_x=200, max_y=180)
    net.startMobility(time=0)
    net.mobility(sta1, 'start', time=10, position='15,115')
    net.mobility(sta1, 'stop', time=20, position='115,10')
    net.mobility(sta2, 'start', time=30, position='20,130')
```

```
net.mobility(sta2, 'stop', time=60, position='150,10')
    net.mobility(sta3, 'start', time=25, position='140,10')
    net.mobility(sta3, 'stop', time=60, position='15,110')
    net.stopMobility(time=120)
    info("**starting network**\n")
    net.build()
    c0.start()
    ap1.start([c0])
    ap2.start([c0])
    ap3.start([c0])
    ap4.start([c0])
    ap5.start([c0])
    info("**running CLI**\n")
    CLI(net)
    info("**stopping network**\n")
    net.stop()
if _-name_- = '_-main_-':
    setLogLevel('info')
    plot = False if '-p' in sys.argv else True
    topology()
```

b Task 2 code

```
import sys
from mininet.log import setLogLevel, info
from mn_wifi.link import wmediumd, adhoc
from mn_wifi.cli import CLI
from mn_wifi.net import Mininet_wifi
from mn_wifi.wmediumdConnector import interference
def topology (args):
    "Create a network"
    net = Mininet_wifi(link=wmediumd, wmediumd_mode=interference)
    info("*** Creating nodes\n")
    sta1 = net.addStation('STA1', ip6='2024::11', mac='00:00:00:00:01:11',
       position = '20,10,0', range=30, antennaGain=5, antennaHeight=1)
    sta2 = net.addStation('STA2', ip6 = '2024::12', mac = '00:00:00:00:01:12',
       position = '45,10,0', range=30, antennaGain=6, antennaHeight=2)
    sta3 = net.addStation('STA3', ip6='2024::13', mac='00:00:00:00:01:13',
       position = '70,10,0', range=30, antennaGain=7, antennaHeight=3)
    net.setPropagationModel(model="logDistance", exp=4)
```

```
info("*** Configuring nodes\n")
    net.configureNodes()
    info("*** Creating links\n")
    #testing 'batman_adv', 'batmand', 'olsrd' manet protocol
    net.plotGraph(\min_{x=-20}, \min_{y=-40}, \max_{x=120}, \max_{y=90})
    net.addLink(sta1, cls=adhoc, intf='STA1-wlan0', ssid='adhocUH', mode='g',
       channel=5, ht_cap='HT40+', proto='batman_adv')
    net.addLink(sta2, cls=adhoc, intf='STA2-wlan0', ssid='adhocUH', mode='g',
       channel=5, ht_cap='HT40+', proto='batman_adv')
    net.addLink(sta3, cls=adhoc, intf='STA3-wlan0', ssid='adhocUH', mode='g',
       channel=5, ht_cap='HT40+', proto='batman_adv')
    info("*** Starting network\n")
    net.build()
    info("*** Running CLI\n")
    CLI(net)
    info("*** Stopping network\n")
    net.stop()
if __name__ = '__main__ ':
    setLogLevel('info')
    topology (sys.argv)
    Task 3 code
\mathbf{c}
```

```
from mininet.topo import Topo
from mininet.node import CPULimitedHost, Host, Node
from mininet.node import OVSSwitch
from mininet.topo import Topo
class MyTopo(Topo):
        "Simple topology with VLAN support."
        def = init_{-}(self):
                "Create custom topology."
                # Initialize topology
                Topo. _{-i} nit _{-} (self)
                h1 = self.addHost('H1', mac='00:00:00:00:15:98', ip
                    = '192.170.50.11/24'
                h2 = self.addHost('H2', mac = '00:00:00:00:15:99', ip
                    = 192.170.50.12/24
```

```
SERVER1 = self.addHost( 'Sv1', mac = '00:00:00:00:16:00', ip
                   = 20.0.0.2/8
               SERVER2 = self.addHost( 'Sv2', mac = '00:00:00:00:16:01', ip
                   = 40.0.0.2/8
               SERVER3 = self.addHost( 'Sv3', mac='00:00:00:00:16:02', ip
                   = 60.0.0.2/8
               Switch1 = self.addSwitch( 'Switch1', cls=OVSSwitch)
               Switch2 = self.addSwitch( 'Switch2', cls=OVSSwitch)
               Switch3 = self.addSwitch('Switch3', cls=OVSSwitch)
               Switch4 = self.addSwitch( 'Switch4', cls=OVSSwitch)
               Switch5 = self.addSwitch( 'Switch5', cls=OVSSwitch)
               self.addLink( h1, Switch4)
                self.addLink( h2, Switch5)
               self.addLink(SERVER1, Switch1)
               self.addLink(SERVER2, Switch1)
               self.addLink(SERVER3, Switch1)
               self.addLink(Switch1, Switch2)
               self.addLink(Switch1, Switch3)
               self.addLink( Switch2, Switch4)
               self.addLink(Switch2, Switch5)
               self.addLink(Switch3, Switch4)
               self.addLink(Switch3, Switch5)
                self.addLink(Switch4, Switch5)
topos = { 'mytopo': ( lambda: MyTopo() ) }
```

d Task 4 code

```
from mininet.topo import Topo
from mininet.net import Mininet
from mininet.node import Node
from mininet.log import setLogLevel, info
from mininet.cli import CLI
class MyTopo (Topo ):
    "Simple topology example."
    def _-init_-(self):
        "Create custom topo."
        # Initialize topology
        Topo. _{-i} nit _{-} ( self )
        # hosts and switches
        h1 = self.addHost('H1', ip = '50.10.10.10/8')
        h2 = self.addHost('H2', ip = '50.10.10.11/8')
        h3 = self.addHost('H3', ip = '50.10.10.12/8')
        h4 = self.addHost('vidSrc', ip = '50.10.10.20/8')
        s1 = self.addSwitch('s1')
        s2 = self.addSwitch('s2')
        self.addLink(h1, s2)
        self.addLink(h2, s2)
        self.addLink(h3, s2)
        self.addLink(s1, s2)
        self.addLink(h4, s1)
topos = { 'mytopo': ( lambda: MyTopo() ) }
```