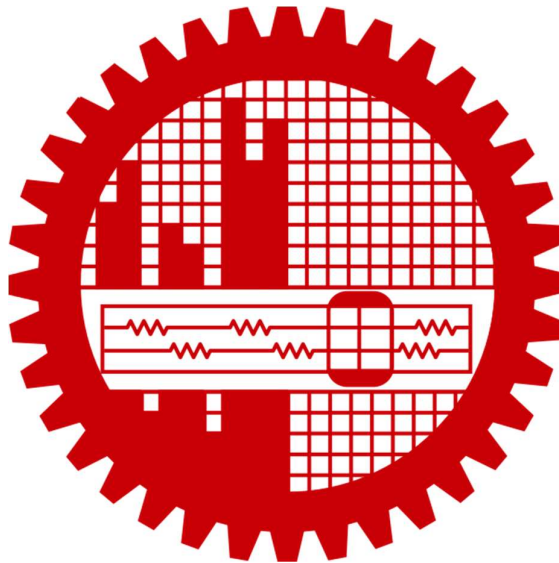


# **NS2 Project Report**

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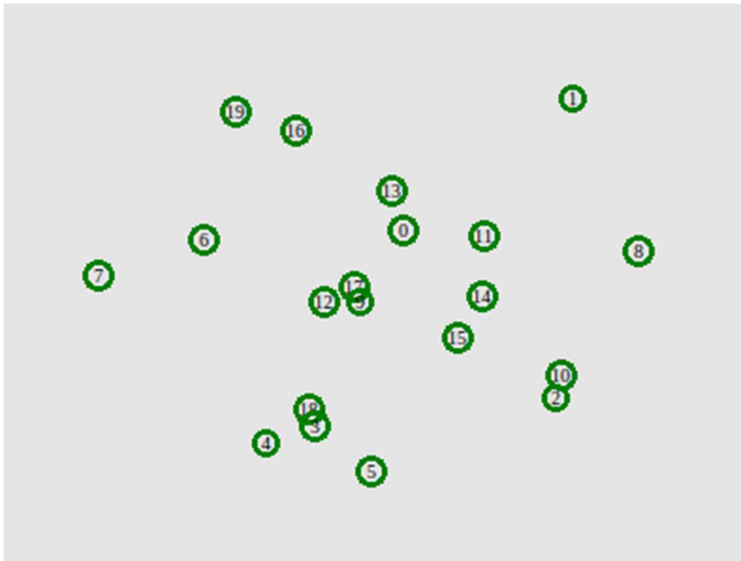
**Department of Computer Science and Engineering**  
**Bangladesh University of Engineering and Technology**

## Introduction

NS2 is an open-source event-driven simulator designed specifically for research in communication networks. In this project, I have to vary different parameters such as number of nodes, number of flows, number of packets per second, speed of nodes and measure some metrics and plot them in some graphs. I also had to modify some mechanisms of Retransmission Timeout calculation and compare the modified version with the original implementation of NS2. The paper from which I have taken idea from is [The Peak-Hopper: A New End-to End Retransmission Timer for Reliable Unicast Transport](#)

## Simulated Networks

- Wireless 802.11 (dynamic) network using AODV routing in Random topology
- Wireless 802.15.4 (dynamic) network using DSDV routing in Random Topology



## Parameters and Metrics

### Parameters

1. Number of nodes (20, 40, 60, 80, 100)
2. Number of flows (10, 20, 30, 40, 50)
3. Number of packets per second (100, 200, 300, 400, 500)
4. Speed of nodes (5, 10, 15, 20, 25)

### Metrics

1. Network Throughput
2. End-to-End Delay
3. Packet Delivery Ratio

4. Packet Drop Ratio
5. Energy Consumption

### Overview of the proposed algorithm

The PeakHopper algorithm essentially runs two RTO algorithms in parallel. One algorithm (Short-Term History RTO) monitors the present and short-term history in order to respond to RTT increases. The other algorithm (Long-Term History RTO) simply decays the current value of RTO, and can therefore be said to represent the long-term history.

We describe the details of how the PH-RTO is calculated in five steps.

$$\delta = \frac{RTT_{sample} - RTT_{previous}}{RTT_{previous}} \quad (\text{Step 1})$$

$$D = 1 - \frac{1}{F * S} \quad (\text{Step 2})$$

$$B \leftarrow \min(\max(2 * \delta, D * B), B_{\max}) \quad (\text{Step 3})$$

$$RTT_{\max} = \max(RTT_{sample}, RTT_{previous}) \quad (\text{Step 4})$$

$$RTO \leftarrow \max(D * RTO, (1 + B) * RTT_{\max}, RTO_{\min}) \quad (\text{Step 5})$$

1. Having collected a new RTT sample,  $RTT_{sample}$ , we compare this value to the previous RTT sample collected,  $RTT_{previous}$ , as shown in Step 1. We call the normalized change between these two samples delta. This is the measure of the short-term changes in RTT.
2. In Step 2, we further define a decay factor,  $D$ .  $D$  determines how rapidly the RTO is decayed. We also introduce a fader variable,  $F$ , which controls the speed of this decay (a high  $F$  gives a slow decay and vice versa).
3. In Step 3, we calculate a booster variable  $B$ . The booster variable determines how high the RTO should hop when a large RTT increase has been detected.
4. In Step 4, we set  $RTT_{\max}$  to the maximum of the new RTT sample,  $RTT_{sample}$ , and the previous RTT sample,  $RTT_{previous}$ .  $RTT_{\max}$  is used to represent the short-term history of the RTT.
5. In step 5, We set RTO to the maximum of a long-term history (represented by the term  $D * RTO$ ) and the short-term history (represented by the term  $((1+B) * RTT_{\max})$ .
6. In final step (Step 6), we ensure that the RTO does not fall below the minimum allowed RTO.

### Modifications:

I have modified the RTO calculation algorithm in NS2. The relevant files were tcp.h and tcp.cc.

In tcp.h new variables are declared:

```
293     double minrtt_; /* min value of an rtt
294     //Shahrukh Changes
295     double B; //save rakhar jonno
296
297     double RTTsample;
298     double RTTprev;
299     double F;
300     double S;
301     double RTO;
302
303     //Shahrukh Change done
304     int to_resetRTO; /* Up backoff RTO after
```

In rtt\_timeout() function new mechanism is implemented.

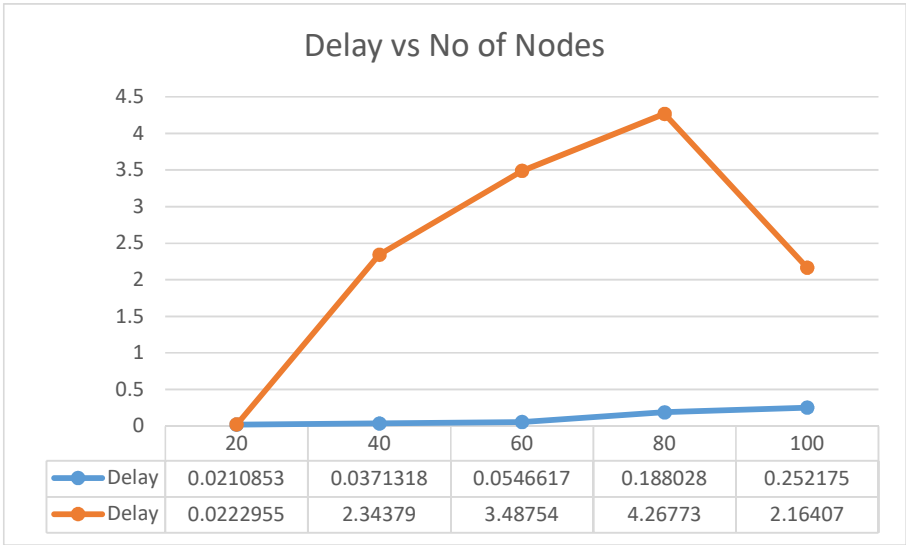
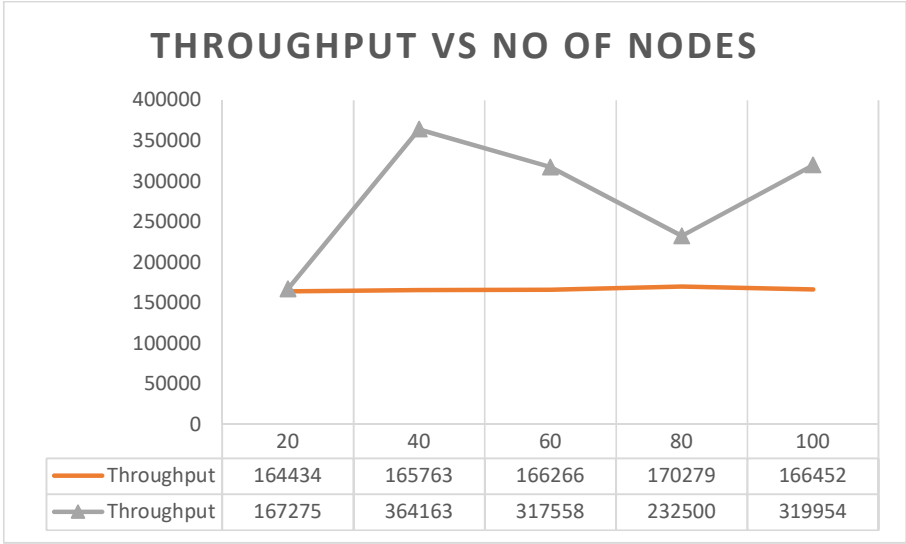
```
double TcpAgent::rtt_timeout()
{
    //Shahrukh Changes
    int peakhopper=1;
    if(peakhopper==1)
    {
        double delta=0;
        if (RTTprev!=0)
        {
            delta=(RTTsample-RTTprev)/RTTprev;
            if(ts_option_)
            {
                S=t_seqno_ - highest_ack_;
            }
            else
            {
                S=1;
            }
            double D;
            if (S==0)
            {
                D =1;
            }
        }
        else
    }
```

```
551     {
552         D = 1;
553     }
554
555     else
556     {
557         D = 1 - (1/(F*S));
558
559         B=fmax(delta, D*B);
560         double RTTmax=fmax(RTTsample, RTTprev);
561         RT0=fmax(D*RT0, (1+B)*RTTmax);
562         RT0=fmax(RT0, minrto_);
563         return RT0;
564     }
565
566     //Change Done by Shahrukh
567
568
569
```

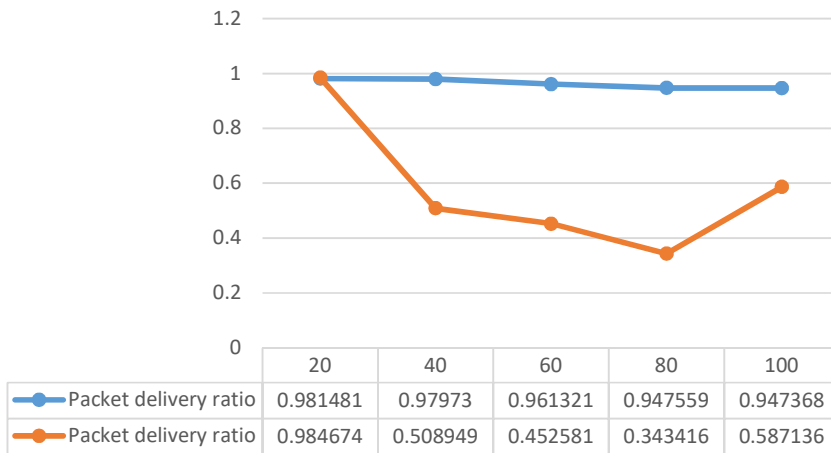
**Graphs:**

802.11 network:

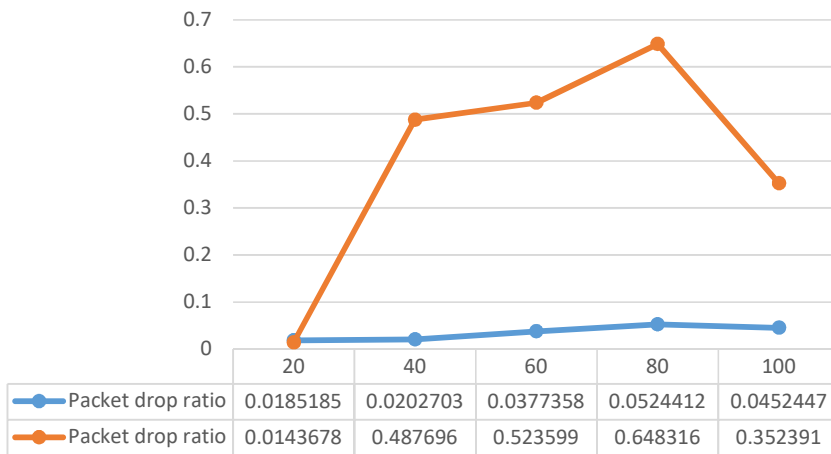
**Varying no of nodes:**



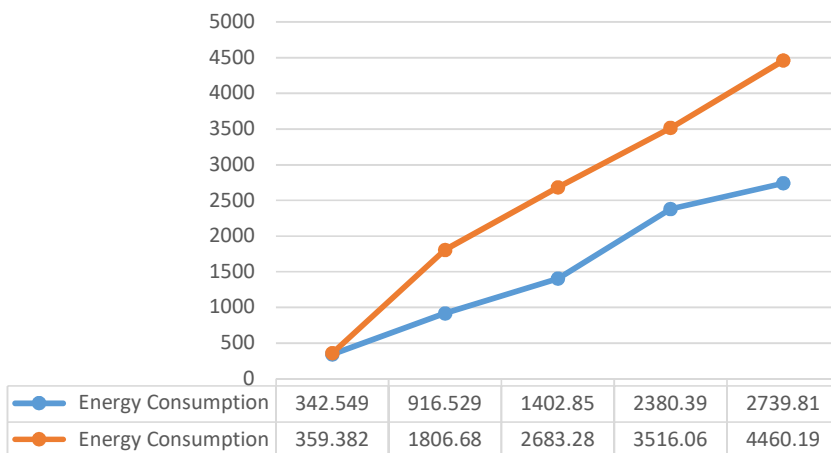
### Delivery Ratio vs no of nodes



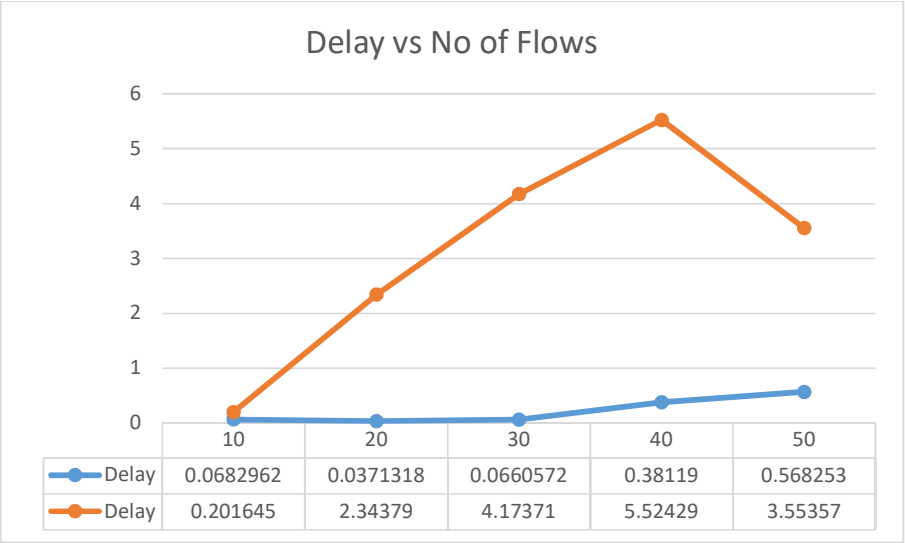
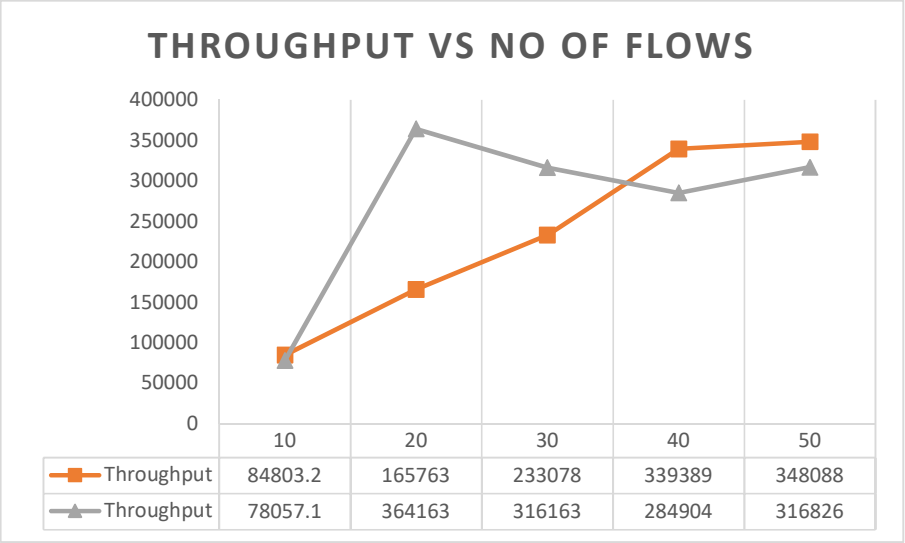
### Drop Ratio vs No of nodes



### Energy Consumption vs no of nodes

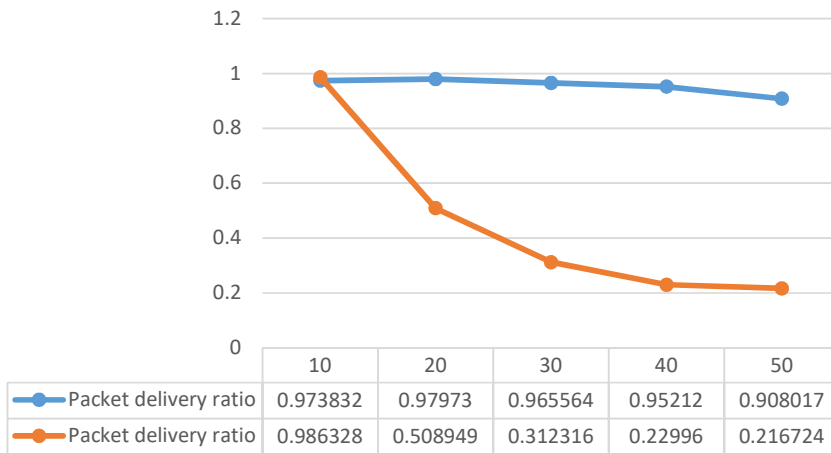


Varying no of flows:

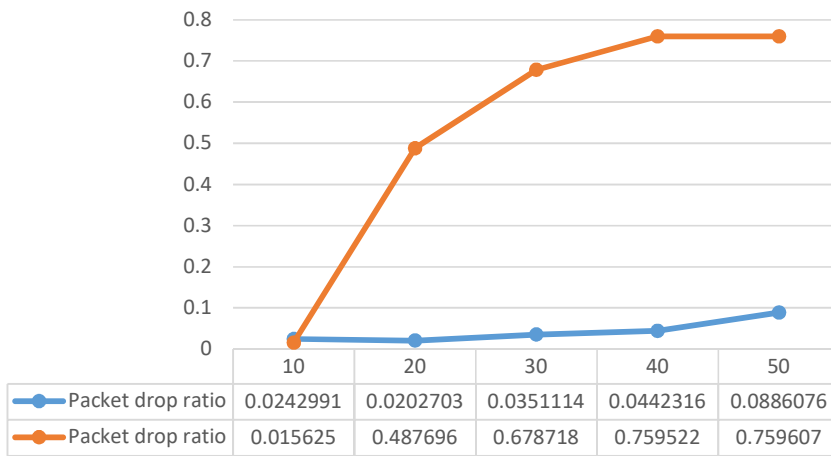




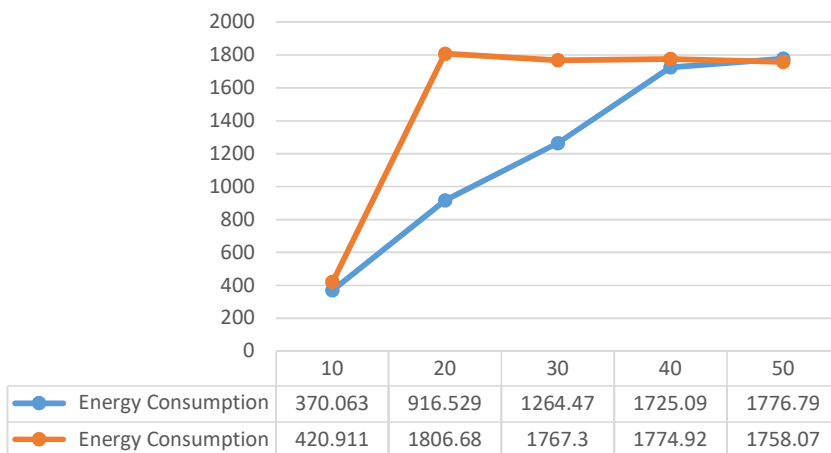
### Delivery Ratio vs No of Flows



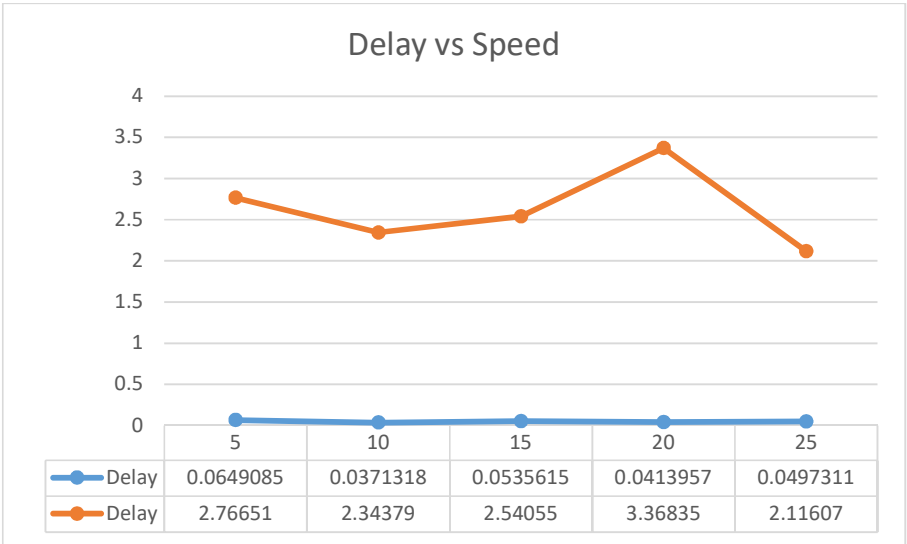
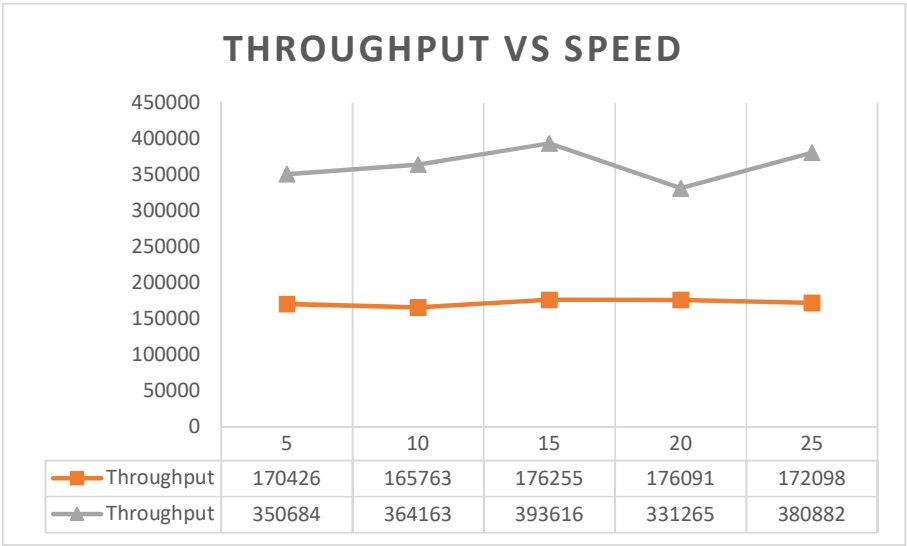
### Drop Ratio vs no of flows



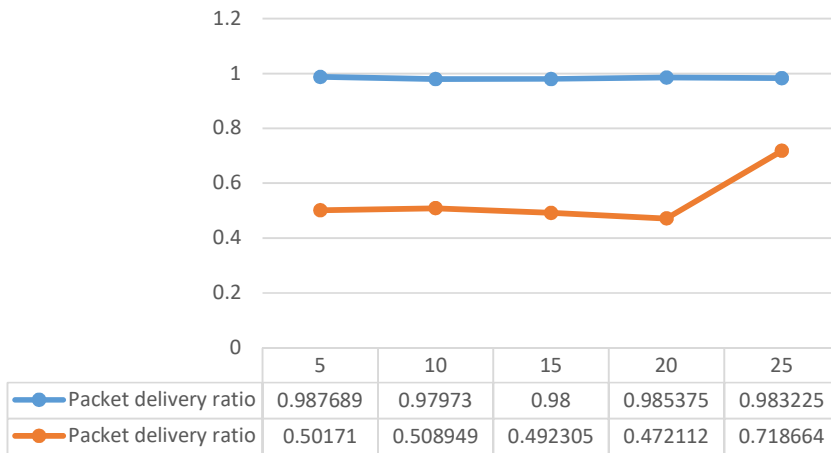
### Energy Consumption vs no of flows



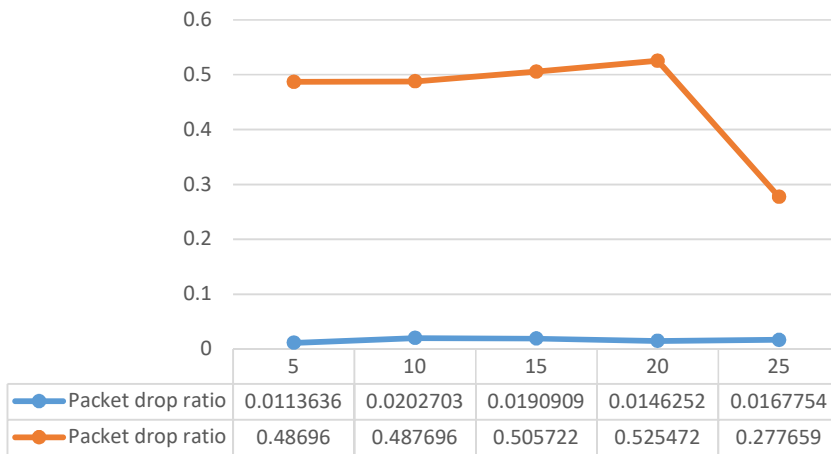
Varying speed:



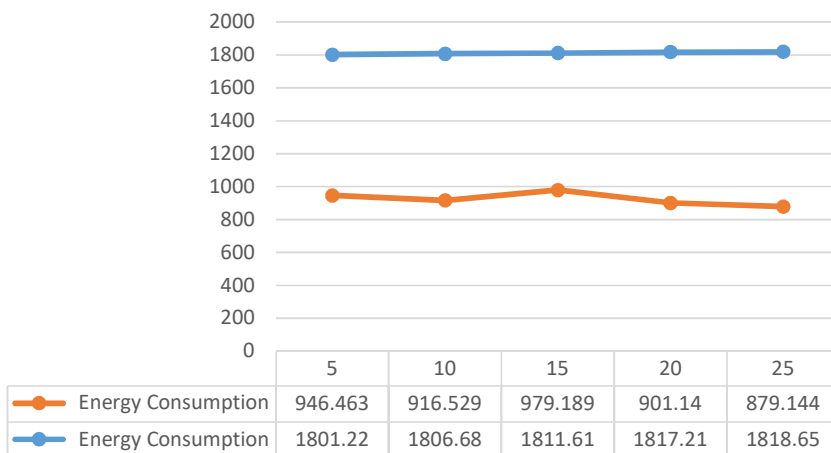
Delivery Ratio vs Speed



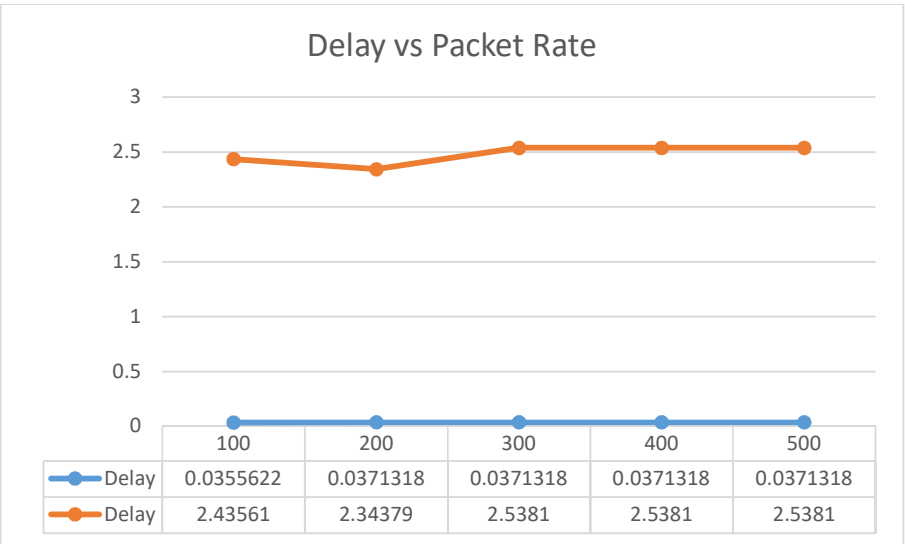
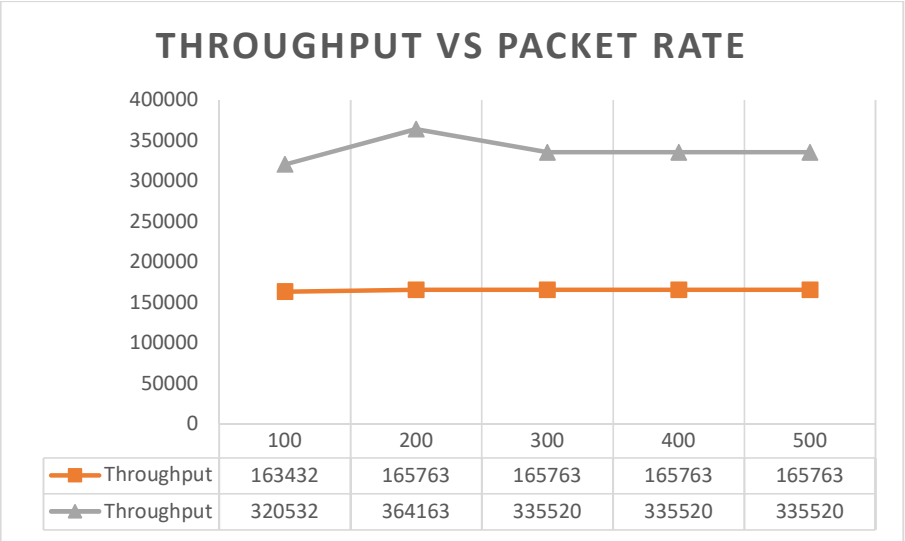
Drop Ratio vs Speed

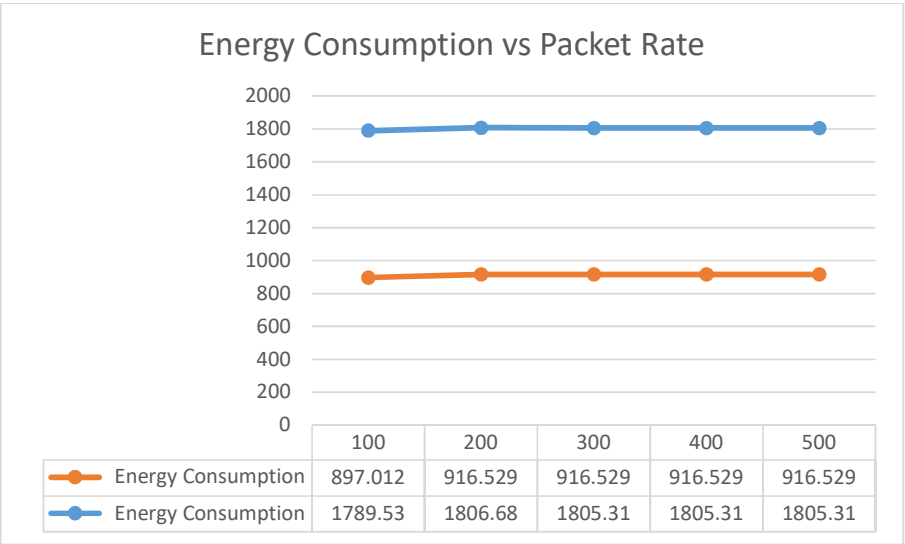
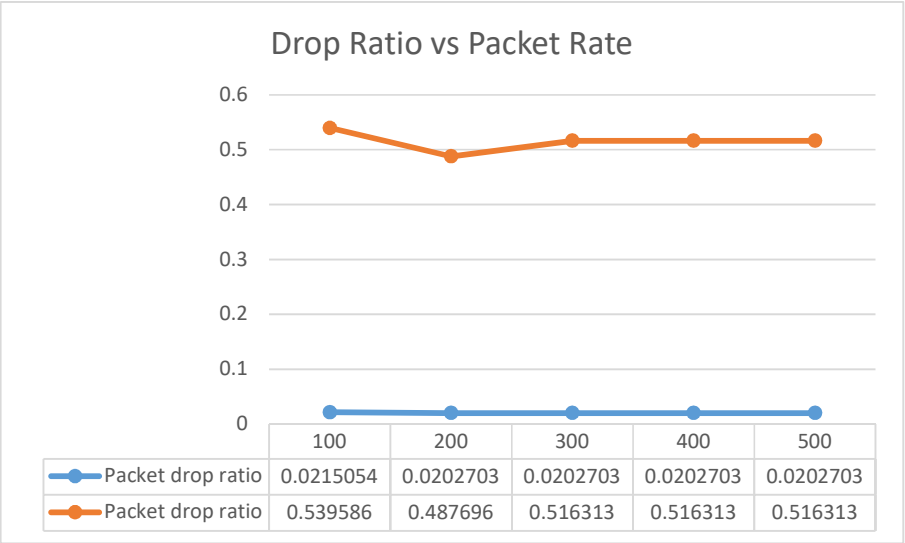
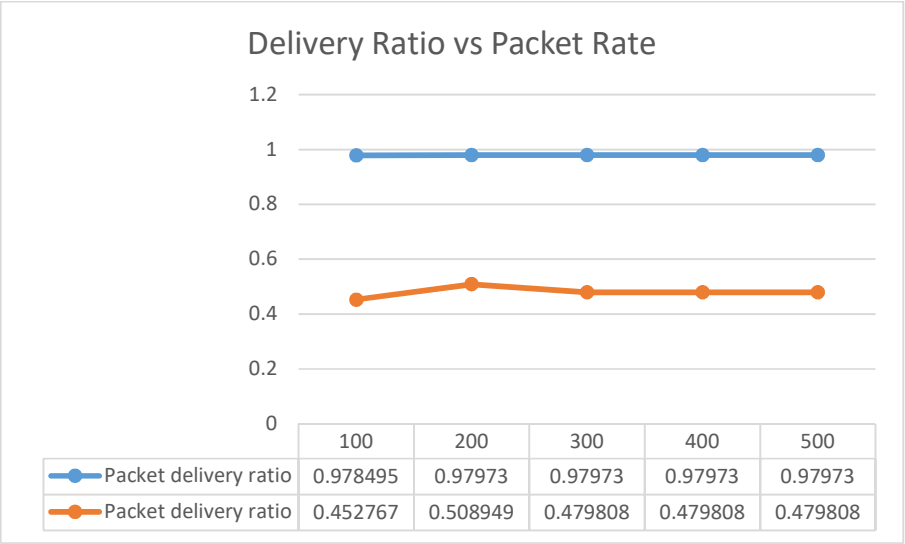


Energy Consumption vs Speed



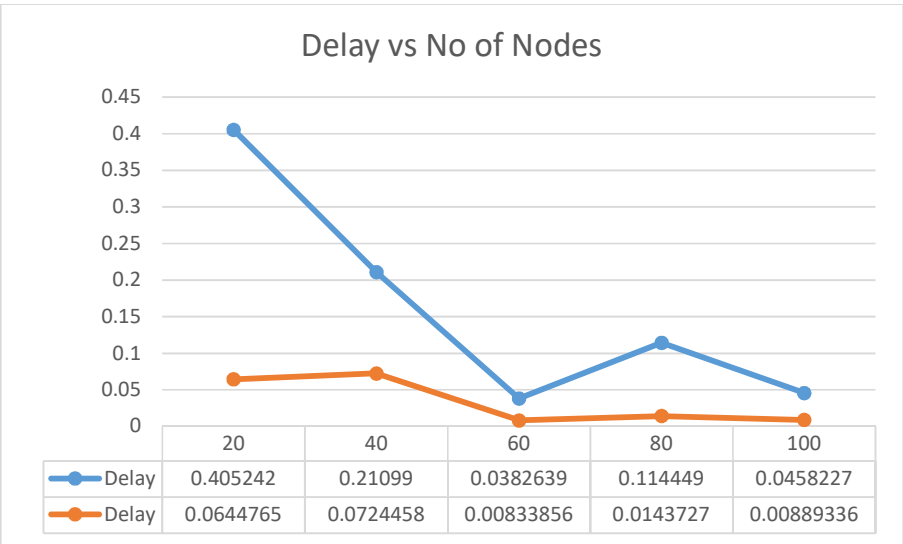
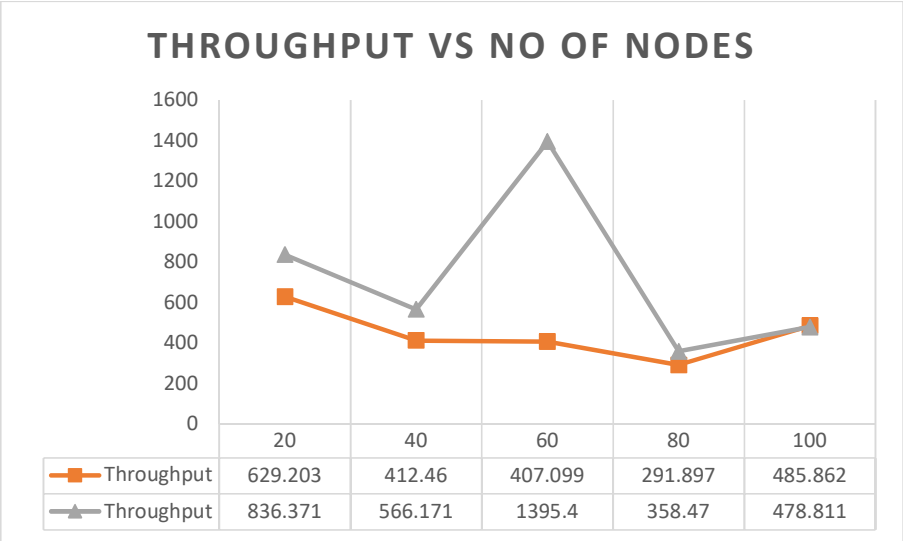
Varying packet per second:



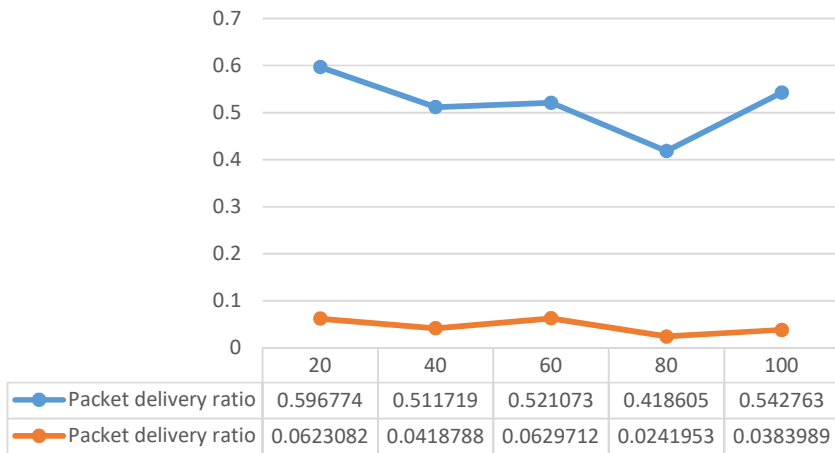


802.11 network:

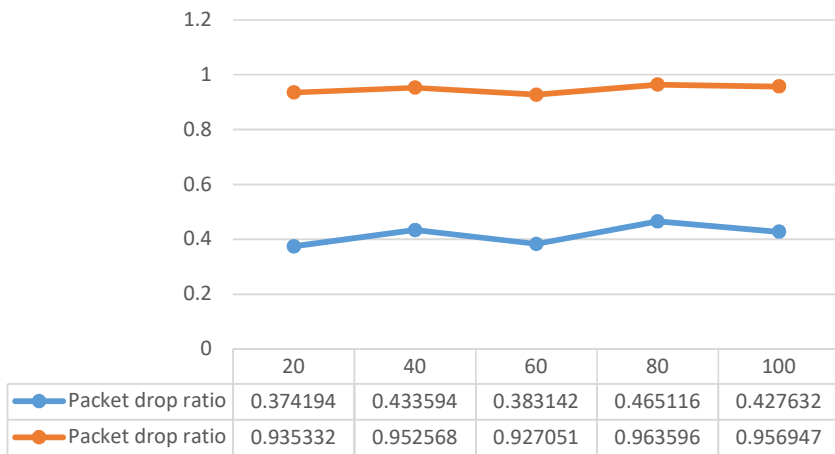
Varying no of node:



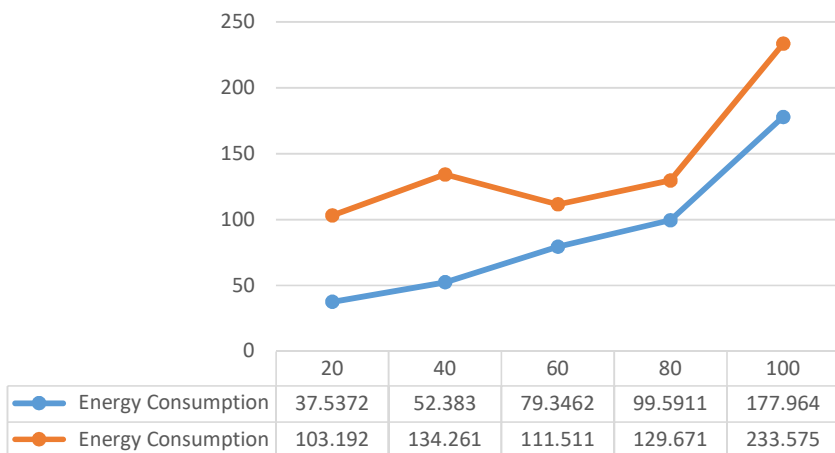
### Delivery Ratio vs no of nodes



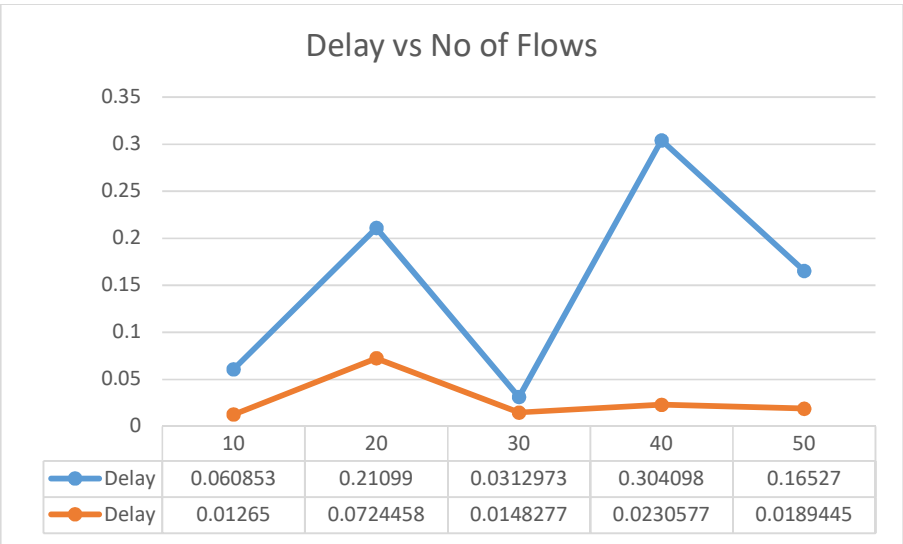
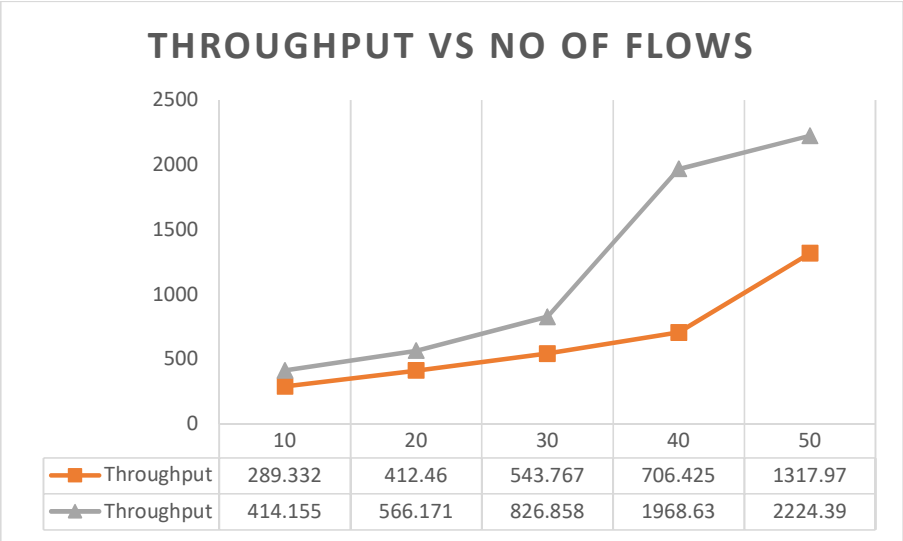
### Drop Ratio vs No of nodes



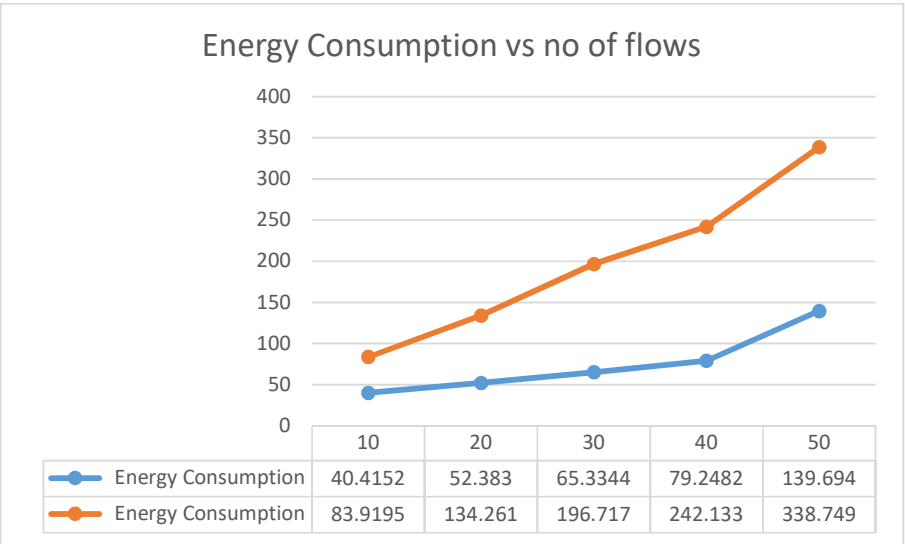
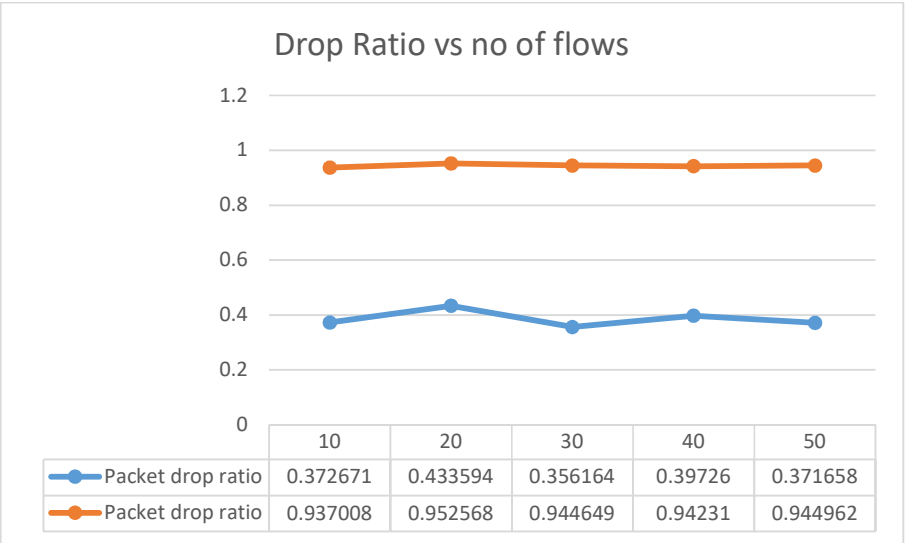
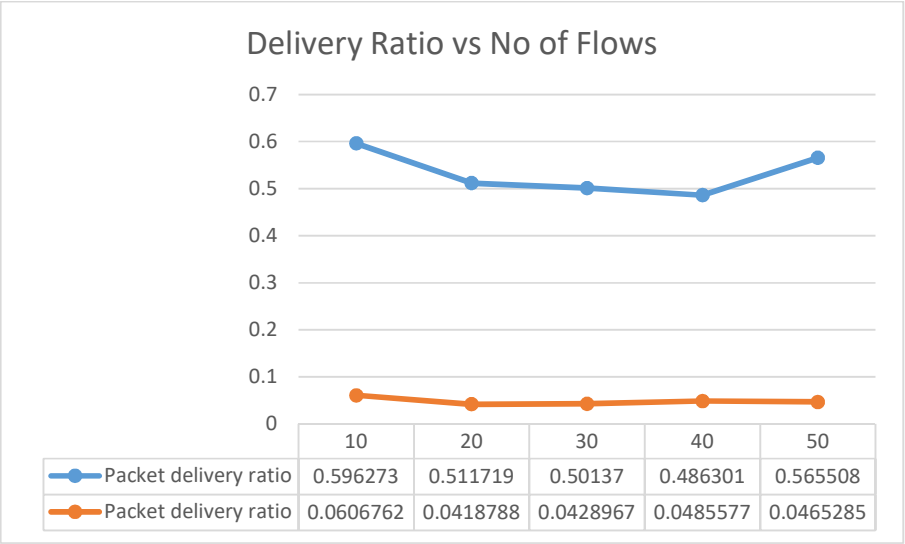
### Energy Consumption vs no of nodes



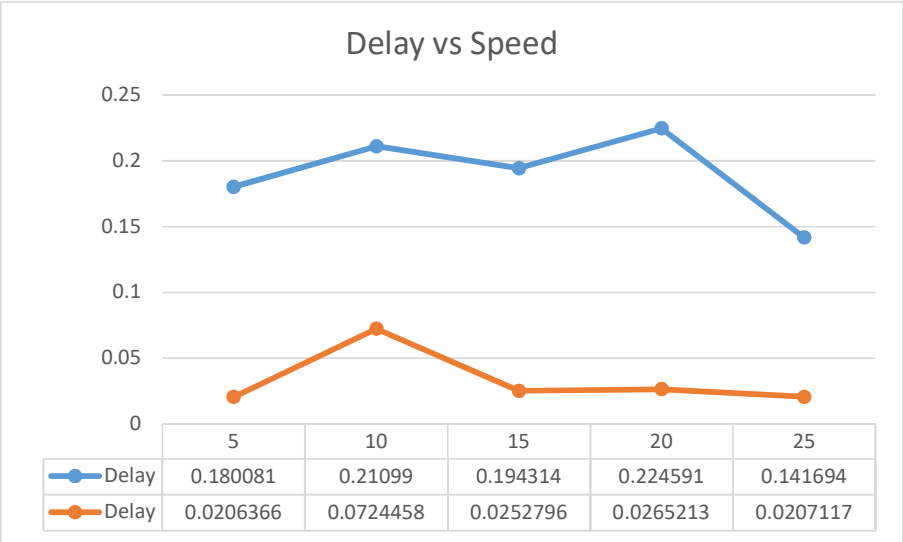
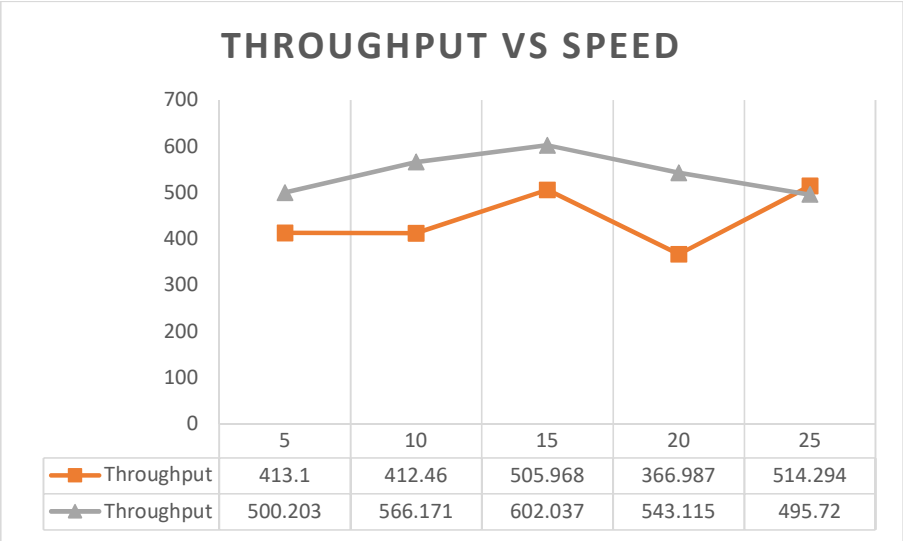
Varying no of flows:



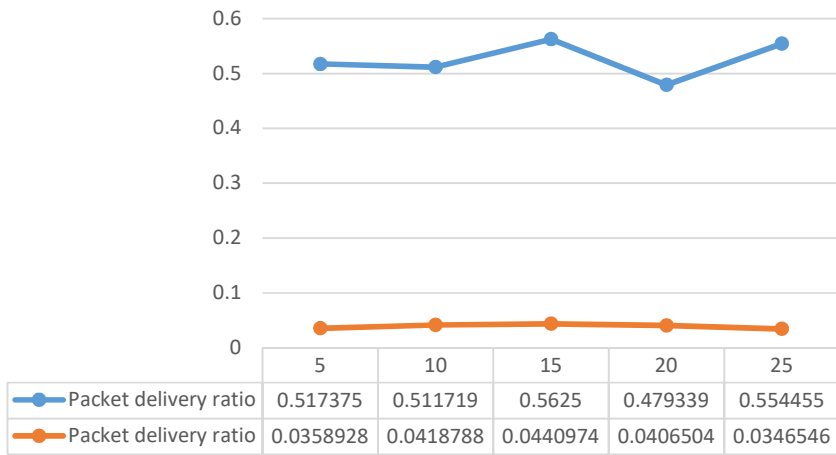




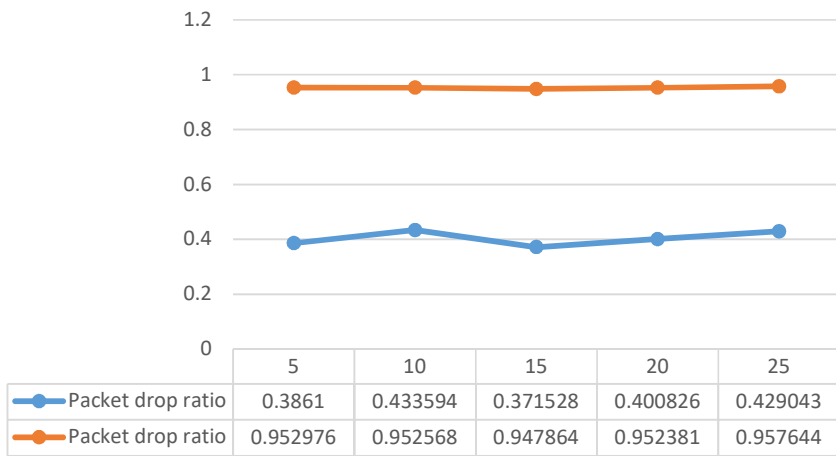
Varying speed of nodes:



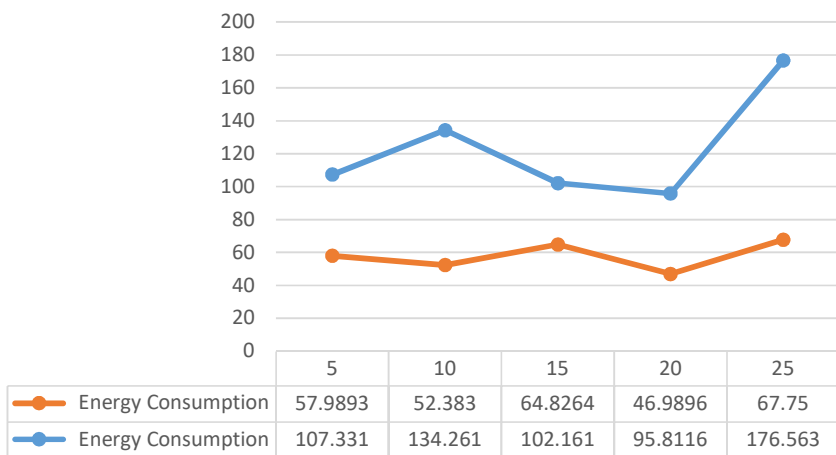
Delivery Ratio vs Speed



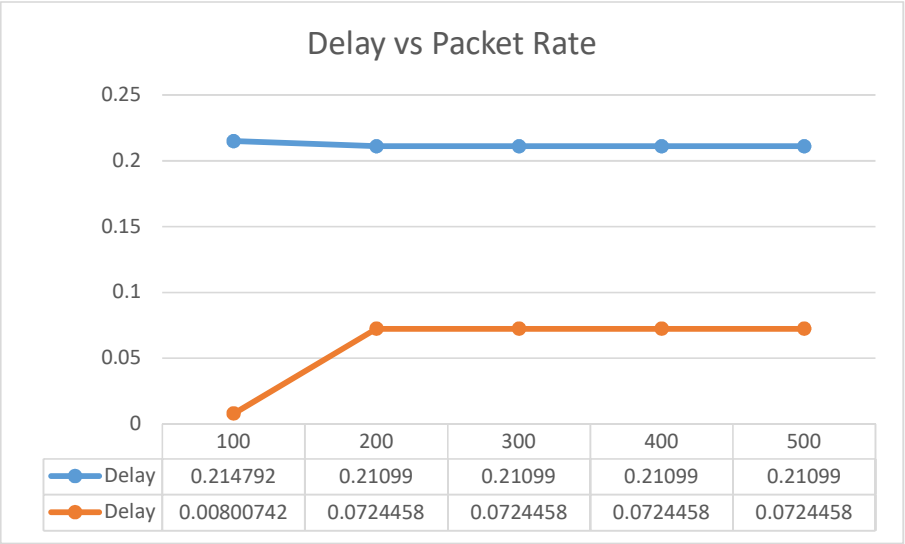
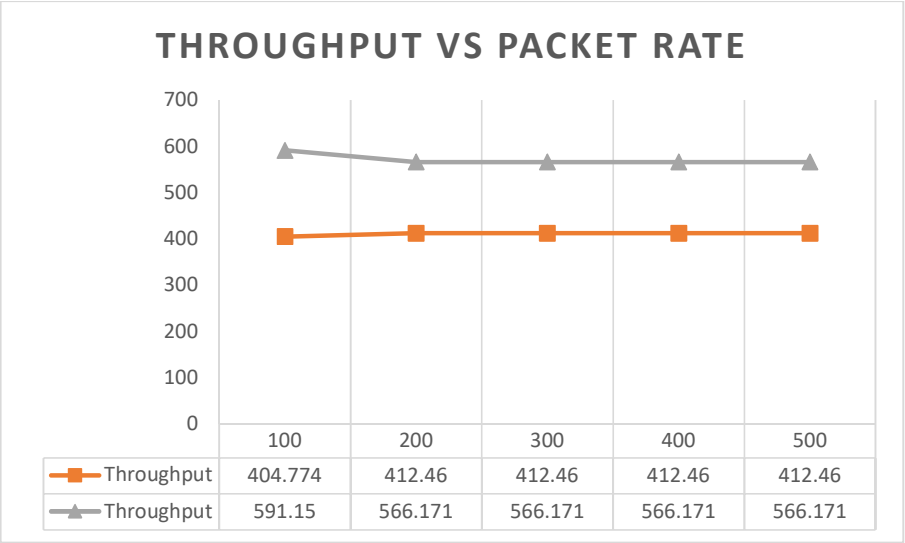
Drop Ratio vs Speed

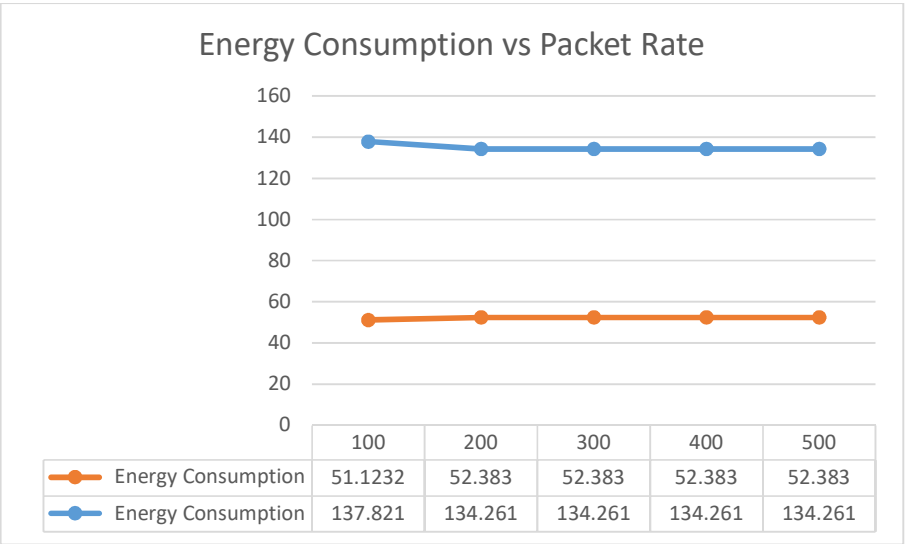
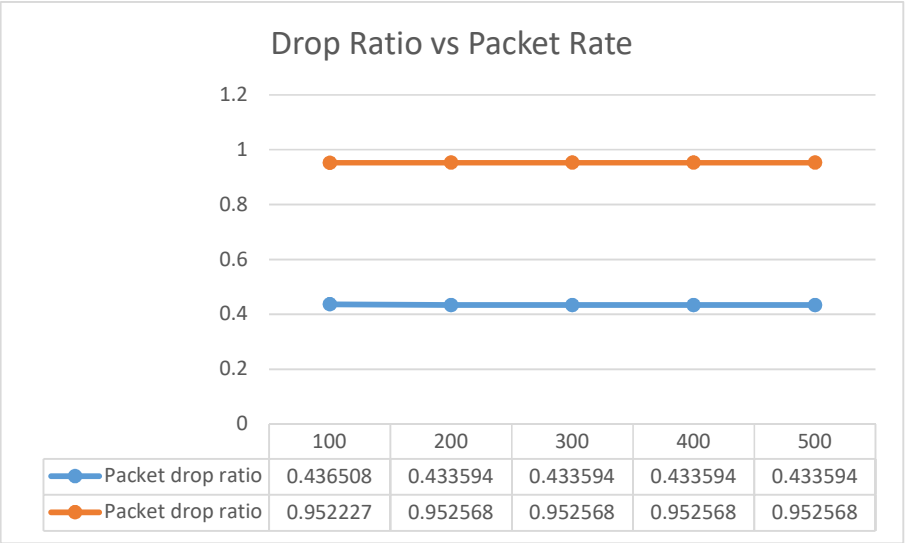
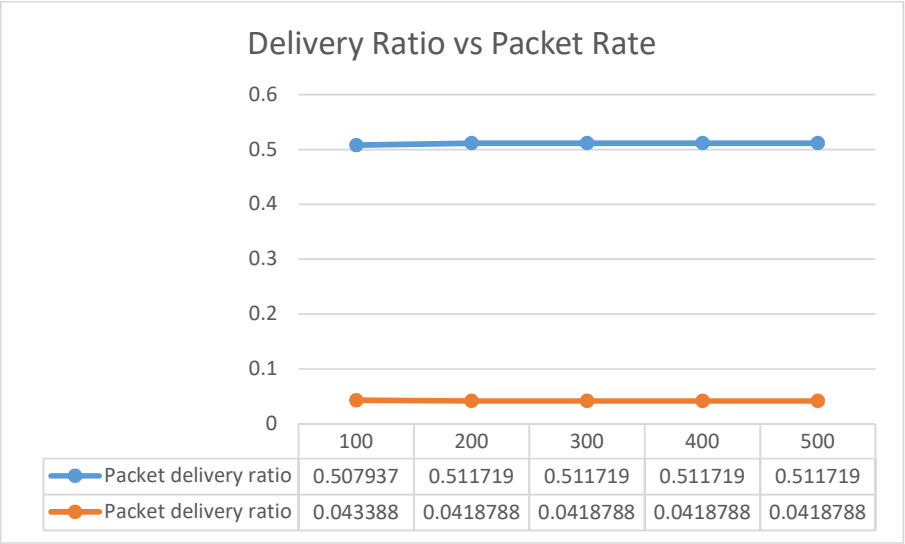


Energy Consumption vs Speed



Varying packet rate:





# Summary:

## Explanation of Results found in 802.11 static network

### Varying Number of Nodes

- Throughput: Decreases as increase of node congesting the network flow more
- End-to-End Delay: Increases because of more congested network
- Delivery Ratio: Decreases with increase of more nodes congested in same network
- Drop Ratio: Starts to increase a little bit because of network congestion.
- Energy Consumption: Increases because of more congested network.

### Varying Number of Flows

- Throughput: Increases due to more flow in the network
- End-to-End Delay: Increases due to more flow in the network
- Delivery Ratio: Shows a trend of decrease with increasing flows
- Drop Ratio: Increases because of network congestion.
- Energy Consumption: Increases because of more congested network.

### Varying Speed of Nodes

- Throughput: Increases due to more movement of the nodes.
- End-to-End Delay: Decreases due to more movement of the nodes.
- Delivery Ratio: Shows a trend of decrease.
- Drop Ratio: Increases because of randomness in a large network.
- Energy Consumption: Remains almost same.

### Varying Number of Packets/Second

- Throughput: Increases and then becomes flat.
- End-to-End Delay: Increases and then becomes flat.
- Delivery Ratio: Increases and then becomes flat.
- Drop Ratio: Decreases and then becomes flat.
- Energy Consumption: Increases and then becomes flat.

## **Explanation of Results found in 802.15.4 static network**

### **Varying Number of Nodes**

- Throughput: Decreases as increase of node congesting the network flow more
- End-to-End Delay: Decreases because of more nodes
- Delivery Ratio: Decreases with increase of more nodes congested in same network
- Drop Ratio: Starts to increase a little bit because of network congestion.
- Energy Consumption: Increases because of more congested network.

### **Varying Number of Flows**

- Throughput: Increases due to more flow in the network
- End-to-End Delay: Increases due to more flow in the network
- Delivery Ratio: Shows a trend of decrease with increasing flows
- Drop Ratio: Increases because of network congestion.
- Energy Consumption: Increases because of more congested network.

### **Varying Speed of Nodes**

- Throughput: Increases and then decreases due to speed increment.
- End-to-End Delay: Increases and then decreases due to more movement of the nodes.
- Delivery Ratio: Shows a trend of increase.
- Drop Ratio: Decreases because of randomness in a large network.
- Energy Consumption: Increases slightly.

### **Varying Number of Packets/Second**

- Throughput: Increases and then becomes flat.
- End-to-End Delay: Increases and then becomes flat.
- Delivery Ratio: Increases and then becomes flat.
- Drop Ratio: Decreases and then becomes flat.
- Energy Consumption: Increases and then becomes flat.

### **Explanation of Performance Metrics Result in RTT/RTO modification algorithm:**

From the mean retransmission graph in the paper, it's seen that modified algorithm provides a lower mean retransmission than default Jacobson's implementation. It was expected because a lower mean indicates that retransmission timer follows a narrow waiting time and a higher mean indicates to a longer waiting time.

Another metric, known as the mean of  $RTO\_N/RTT\_L$  where  $RTO\_N$  is the newly calculated RTO and  $RTT\_L$  is the last RTT sample prior to the RTO update. We then calculated P as the mean value of this

fraction over an entire simulation run. This mean is found to be lower in modified algorithm than in default implementation which is expected too. A retransmission timer with lower mean of this ratios is expected to perform better.

In graphs we noticed that throughput has been increased after this modification that means per second number of received bits has been increased. That means a smaller RTO has occurred and packet drop is early detected. And for this reason throughput has been increased after modification.