

Congestion Control Using Predictive Approach in Mobile Ad Hoc Network

Hariom Soni, Pradeep Kumar Mishra

Abstract- The Active Queue Management algorithms stabilize the instantaneous queue length through mapping the congestion measurement into packet drop probability to achieve high throughput and low average delay. Random Early Detection [RED] is widely used AQM mechanism detecting and avoiding the incipient congestion. The detection of congestion is based on priori estimation of congestion and calculation of average queue length and RED queue parameter settings. This paper describes the research work done in recent and new mechanism to estimates the network congestion and avoiding them, so that network performance will improve. Our goal in this paper is to achieve high throughput and low end to end delay in congested network.

Keywords- [RED] AQM RED,

I. INTRODUCTION

Mobile Ad hoc Network (MANET) [1] is self-organizing network of mobile devices which does not rely in any fixed infrastructure. MANET nodes can be personal devices such as laptop, mobile phones and personal digital assistance (PDA's). Nodes in MANET can take part in the communication if they are in the range of network, and can move freely within transmission range of network and nodes Which are outside the transmission range of network cannot take part in communication. The dynamic nature of MANET with limited resources that can vary with time such as battery power, storage space bandwidth makes QoS provisioning, a challenging problem. To prevent congestion, the current internet use end-to-end congestion control [2], in this mechanism end host are responsible for detection of congestion and packet loss is treated as implicit congestion notification signal from routers. After detection of incipient congestion, packet transmission rate is reduced to decrease the congestion level.

One of recent research areas for the network is seeking some mechanism to avoid high packet loss rates. When a packet is dropped before it reaches to its destination, all the resources has been consumed in transit are inevitably wasted. In extreme cases, this situation can lead to congestion collapse in which high degrees of packet delay and loss occurs caused by routers discarding packets due to excessive queue size. Congestion control and queue management in the network has been one of the active areas of research in the past few years. Few enhancements have been made by some researchers to solve the problems of high packet loss rates. Loss rates are especially high during heavy network congestion, when a large number of connections compete for limited network bandwidth.

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Due to an exponential increase in network traffic, many congestion control mechanisms have been proposed, including the deployment of explicit congestion notification (ECN), along with active queue management (AQM) techniques.

Traditional Queue Management such as Drop-tail Queue Management, this mechanism allow packet to enter in queue till the queue is empty and drop the entire incoming packets when queue gets full. In drop tail there is no any such approach for early detection of congestion in network i.e. in congested network packet drop is common problem and re-forwarding of all dropped packets will consume resources such as battery power, transmission link and processing power of nodes. This technique results in some serious drawbacks. Thus, Drop tail queues are not suited to interactive network applications because the drop-tail queues are always full or close to full for long periods of time and packets will continuously dropped when the queue reached its maximum length.

II. RANDOM EARLY DETECTION

Floyds et al proposed Random Early Detection (RED) [3] in 1993. The basic idea of this mechanism is that the router can detect incipient congestion by monitoring the average queue length. Once the congestion is detected, router selects the source terminal to notify the congestion. So the source terminal can reduce the data transmission rate before the queue overflow, and try to alleviate the network congestion. RED [4][5] algorithm consists of two steps: the first step is to calculate the average queue length, and the second step is to calculate the packet drop probability. Packet drop probability is used to decide whether to drop the packet or not, packet drop is treated as the signal of congestion.

```
Initialization
  avg = 0
  count = 1
end
for each packet arrival
  calculate the new average queue size 'avg'
  if the queue is nonempty
    avg = (1 - wq) * avg + wq * q
  else
    m = f(time - q_time)
    avg = (1 - wq)^m * avg
  end
  if minth <= avg < maxth
    increment count
    calculate probability 'pa'
    pb = maxp * (avg - minth) / (maxth - minth)
    pa = pb / (1 - count * pb)
  end
  with probability 'pa'
    mark the arriving packet
    count = 0
  end
else if maxth <= avg
  mark the arriving packet
  count = 0
end
else
  count = -1
end
```

Figure 1: RED queue management algorithm

A. Calculation of the Average Queue Length

RED calculates the average queue length (Avg_q), by using the following formula:

$$Avg_q = (1 - W_q) * Avg_q + q * W_q \dots\dots\dots (1)$$

Here, W_q represents the weighted value, and q represents the actual queue length in the sampling moments.

B. Calculation of the Packets Drop Probability

RED has two thresholds Min_{th} and Max_{th} , which are related with queue length. When the packet reaches the router, RED calculates the average of the queue length Avg_q immediately. Then it determines the packet drop probability based on Avg_q , Min_{th} and Max_{th} . When avg_q is greater than Max_{th} , all packets are discarded, and the packet loss rate is 1. When Avg_q is between Min_{th} and Max_{th} , we have the following Packet Drop Probability (PDP) formula:

$$P_b = Max_p * (Avg_q - Min_{th}) / (Max_{th} - Min_{th}) \dots\dots (2)$$

$$P = P_b / (1 - count * P_b) \dots\dots\dots(3)$$

Packet drop probability is used to decide whether to drop the packet or not, packet drop is treated as the signal of congestion.

III. ARED

Adaptive RED, which is abbreviating for ARED [8], is a kind of improved RED algorithm. ARED presents an automatic configuration mechanism, which can configure the parameters Max_p according to the stream changes. The basic idea of ARED is make RED more or less aggressive based on the observation of the average queue length. If the average queue length remains around minimum threshold (Min_{th}) then early detection is simply too aggressive. On the other side if the average queue length remains around maximum threshold (Max_{th}), then early detection is being too conservative. The algorithm changes the packet dropping probability according to its aggressiveness sensed. Thus it can to maintain the average queue length between Min_{th} and Max_{th} . If the average queue length swings around Min_{th} , it shows the congestion control is too radical, then we decrease Max_p , and let

$$Max_p = Max_p / \alpha \dots\dots\dots(4)$$

IV. EXPLICIT CONGESTION NOTIFICATION

Random Early Detection [6] drops packet early, before queue becomes full to detect the incipient congestion. This mechanism has some drawbacks such as drop some packets are dropped before even there is available queue space. This mechanism sometime unnecessarily drops packets. ECN is an optional feature that is only used when both source and destination also routers in the path support it and both endpoints are willing to use it. ECN permits end-to-end notification of network congestion by marking Congestion Experience [CE] bit in IP header of packet without dropping them. When ECN is successfully negotiated, router which is ECN-aware may set a mark in the IP header to signal incipient congestion. The receiver of the packet echoes the congestion indication to the source, Source upon reception of notification from destination reduces packet transmission rate thus packet loss is reduces and resources are conserved.

V. RELATED WORKS

Several solutions have been proposed in the literature for the Queue Management in Mobile Ad hoc Networks (MANET's). Some of them are as follows:

K. Dinesh Kumar et al propose a predictive queue management strategy named PAQMAN [7] that proactively manages the queue which requires negligible computational overhead and is lightweight. PAQMAN does not require any prior knowledge of the traffic model, this reduces Packet loss, Increases transmission efficiency. The performance has been compared with drop tail and those results show that PAQMAN reduces packet loss ratio while at the same time increasing transmission efficiency.

Zhenyu et al propose an AQM scheme with dynamic reference queue threshold named ARTAQM [8]. Adopting a dynamic reference queue is the prominent feature of ARTAQM. Using an adaptive filtering algorithm NOEKF, the predicted traffic rate can be calculated. By means of measuring PLR and average traffic rate, the estimated average traffic rate in the next time can be deduced. The difference of the estimated average rate and the link capacity is the input of squashing function to adjust the reference queue. Therefore, the relationship between traffic condition and the reference queue length is established. Simulation results are compared with other schemes, ARTAQM offers stable and flexible queue length reduces packet loss ratio and increases link utilization. Simulation results show that ARTAQM outperforms than other schemes in terms of queue stability, less packet loss ratio and higher link utilization.

MRED algorithm [9], it is modified from RED algorithm as its name. In addition to the condition of $Avg_q > Max_{th}$, the primary idea of MRED algorithm takes an extra condition of $q > Max_{th}$ into account to decide if packets are dropped directly. MRED algorithm can provide higher transmission throughput and avoid the sensitivity of RED performance to the parameter setting. And Progressive Random Early Detection (PRED) algorithm improved by the progressive adjustment method is proposed. It regulates the packet dropping probability (PDP) progressively by comparing the instantaneous queue size with the progressive maximum queue threshold parameters. PRED not only adopts the instantaneous queue size to adjust the maximum threshold adaptively, but also regulates the packet dropping probability progressively by comparing the instantaneous queue size with the progressive maximum queue threshold parameters. Hence, the proposed PRED algorithm can keep advantages of conventional RED.

Torres Rob et al presented an innovative TCP [10] flow control method. This algorithm combines RED (Random Early Detection) with TCP window adjustment to improve the network performance. Taking the advantages of RED and window adjustment, the algorithm demonstrates superior stability and fast response with controlled packet dropping rate, and still fully utilizing the network resource. Author presented a novel analytical model based on the discrete Markov process in this research. Analysis and simulation show the effectiveness and robustness of the algorithm. The result of the algorithm shows that while fully utilizing the network resource this scheme achieves increased network stability with desired latency and packet dropping rate.



VI. NETWORK PERFORMANCE PARAMETERS IN MANET

Network performance refers to the service quality of providers to the customer. Performance parameters are used to measure the quality of services of the network. These parameters are given below.

A. Average end to end Delay

The average end-to-end delay [1] of data packets is the interval between the data packet generation time and the time when the last bit arrives at the destination. End-to-end delay generally includes all delays, along the path from source to destination. This includes the transmission delay, processing delay, propagation delay, queuing delay experienced at every node in the route.

B. Network Throughput

Network throughput is the average rate of successful message delivery over a communication channel. Throughput is measured in bits per second (bps or bit/s), and sometimes in data packets per time slot or data packets per second.

C. Packet loss ratio

Packet loss occurs when one or more packets of data traveling across a computer network fail to reach their destination. Packet loss is calculated as total lost packet to the total no of transmitted packets.

VII. PROPOSED SOLUTION

Congestion occurs in the situation when the demand for resources exceeding the available capacity. Congestion control mechanism control congestion either before congestion occur or after congestion actually occurred. Proposed congestion control mechanism is based on controlling congestion before it occurs. This mechanism is based on detection of incipient congestion using Random Early Detection with prediction of congestion level and timing in which it become burst. Prediction will based on following assumption:

- 1) Segmented queue size Q into n stages given to an arbitrary value by request.
- 2) The coefficient X_n ($0 < X_0 < X_1 < \dots < X_{n-1} < X_n = 1$) varies between 0 and 1 based on its position from empty queue to maximum available queue size, segments the maximum queue size Q progressively as the progressive maximum queue threshold parameters $Q * X_{n-1}$.
- 3) The shifting factors S_n ($S_0, S_1, S_2, \dots, S_{n-1}$)

Shifting factor S will be based on calculation of rate of change in congestion level in regular interval and compares with the congestion level with the previous interval.

VIII. EXPECTED OUTCOME

RED queue management algorithm performance is better than traditional queue management algorithm, but its performance in most cases is affected by setting of queue parameters. Proposed algorithm will predict the behavior of network by monitoring the rate of change in congestion level in regular interval and compares with the congestion level with the previous interval. Based on the prediction proper action will be taken to decrease the congestion level, this work will reduce the congestion in Mobile Ad hoc Network [MANET] and will improve performance.

REFERENCES

- [1] Patel j. k., Dubey j., "Mobile Ad Hoc Network Performance Improvement Using Strategical RED", Ninth International Conference on Wireless and Optical Communications Networks WOCN2012, 2012, IEEE.
- [2] R. Braden, D. Clark, J. Crowcroft, B. Davie, S. Deering, D. Estrin, S. Floyd, V. J. G. Minshall, C. Patridge, L. Peterson, K. Ramakrishnan, S. Shenker, J. Wroclawski, and L. Zhang. Rfc-2309 recommendation on queue management and congestion avoidance in the internet. Technical report, IETF, pp 246-350, 1998.
- [3] S. Floyd and V. Jacobson. "Random early detection gateways for congestion avoidance", IEEE/ACM Transactions on Networking, 1(4):397-413, 1993.
- [4] V. Firoiu and M. Borden, "A study of active queue management for congestion control", In Proceedings of the IEEE Infocom, pages 1435-1444, Tel Aviv, Mar 2000.
- [5] Chin-Hui Chien; Wanjiun Liao, "A self-configuring RED gateway for quality of service (QoS) networks, Multimedia and Expo", 2003. ICME '03., Page(s): 1 - 793-6.
- [6] A. Kuzmanovic, A. Mondal, S. Floyd, K. Ramakrishnan. RFC 5562 - "Adding Explicit Congestion Notification Capability to TCP's SYN/ACK Packets", AT&T Labs Research, June 2009.
- [7] K. Dinesh Kumar, I. Ramya & M. Roberts Masillamani, "Queue Management in Mobile Adhoc Networks (Manets)" 2010 IEEE.
- [8] Zhenyu Na and Qing Guo "An Improved AQM Scheme with Adaptive Reference Queue Threshold" 978-1-4577, 2011 IEEE.
- [9] Guan-Yi Su and Chian C. Ho "Random Early Detection Improved by Progressive Adjustment Method" Proceedings of IEEE 2008 6th National Conference on Telecommunication Technologies and IEEE 2008 2nd Malaysia Conference on Photonics, August 2008, Putrajaya, Malaysia. 2008 IEEE.
- [10] Rob Torres, John Border, George Choquette, Jun Xu, and Je-Hong Jong, "Congestion Control using RED and TCP Window Adjustment" 978-1-4673, 2013 IEEE.