

Generation of a New Algorithm using Priority Measurement Based Routes

Dr. Naveen Hemrajani¹, Dharm Singh² and Arushi Rawal³

¹ Suresh Gyan Vihar University,

Received: 23 May 2011 Accepted: 19 June 2011 Published: 2 July 2011

Abstract

With the explosive growth in multimedia wireless applications such as video streaming and conferencing, the need for appropriate bandwidth allocation mechanism has also increased as in absence of this a flow may experience considerable performance degradation due to free admission of randomly large numbers of flows. Thus, admission control is used to deal with this problem. The role of admission control algorithms is to make sure that admittance of new flows into the networks doesn't violate service commitments made by network to admitted flows. This paper compares the performance of four measurement based admission control algorithms for controlled load service. We evaluate link utilization and adherence to service commitments achieved by these algorithms. Further a new algorithm is proposed using priority measurement based on the round trip times of the nodes.

Index terms— Admission Control, bounded delay, link utilization, intricate collisions. and conferencing, the need for appropriate bandwidth allocation mechanism has also increased as in absence of this a flow may experience considerable performance degradation due to free admission of randomly large numbers of flows. Thus, admission control is used to deal with this problem. The role of admission control algorithms is to make sure that admittance of new flows into the networks doesn't violate service commitments made by network to admitted flows. This paper compares the performance of four measurement based admission control algorithms for controlled load service. We evaluate link utilization and adherence to service commitments achieved by these algorithms. Further a new algorithm is proposed using priority measurement based on the round trip times of the nodes.

1 INTRODUCTION

he simplicity of access mechanism and less sources made accessing audio and video a simple matter. With incomparable amount of audiovisual information becoming available in digital form, on WWW, broadcast data streams, and in personal and professional databases, there is an explosive growth in multimedia wireless applications and without a proper bandwidth allocation mechanism flow may experience significant decrease during session. This is where admission control comes into play in order to maintain a good quality of existing flow.

There have been many proposals for supporting real time applications in packet networks by providing some requests real-time service, it must characterize its traffic so that the network can make its admission control decision. Typically, sources are described by either peak and average ??Ferrai and Verma, 1990) or a filter like token bucket (Ohnishi et al., 1988); these descriptions provide upper bounds on the traffic that can be generated by the source. The traditional real time service provides a hard or absolute bound on the delay of every packet; in ??Clark et al., 1992) and ??Ferrai and Verma, 1992), this service model is called guaranteed service.

Admission control algorithms (ACA's) for guaranteed service use the a priori characterizations of sources to calculate the worst case behavior of all the existing flows in addition to the incoming one. Network utilization

under this model is usually acceptable when flows are smooth; when flows are bursty, however guaranteed service inevitably results in low utilization (Zhang and Ferrari, 1994; Yaffey et al., 1994).

Admission control is a network Quality of Service (QoS) procedure. Admission control determines how bandwidth and latency are allocated to streams with various requirements [Saito, 1993]. Thus this scheme needs to be implemented between network edges and core to control the traffic entering the network.

An application aiming to use the network to transport traffic with QoS must first request a connection, which involves informing network about the traffic's characteristics and QoS required by application. This information is stored in traffic contract. After judging whether it has enough resources available to accept the connection, the network either accepts or denies the request. This is admission control. It is useful in situations where a certain number of connections may all share a link, while greater number of connections causes significant degradation to the point of making them useless.

The user must be able to get a service whose quality is sufficiently predictable that the application can operate in an acceptable way over duration of time determined by the user" [Braden et al., 1994]. Admission control is the main task that a Bandwidth broker has to perform. Most of the brokers use simple admission control modules, although proposals for more sophisticated controls are also there.

Admission control algorithms ensure that new flow in the network does not violate service commitments made by network to admitted flows. These commitments could be quantitative or qualitative. The main criteria to evaluate the algorithms are to see how it can fulfill its role of ensuring that service commitments are not violated. To ensure complete commitment conformance we can allocate enough resources to meet the worst case requirements of each flow. The main results of the comparisons are summarized below:-1. In the operating region where losses occur under all MBAC's, they can all be induced to give the same loss-load curve by tuning their measurement parameters. 2. All the MBAC's studied perform similarly because they are all based on admission equations of the same form:

2 Global Journal of

$V' < f(\mu) - g(\mu)$

Where, V' is the measured load, μ is the link bandwidth, and $f(-)$ and $g(-)$ are functions of the source's reserved rate and number of admitted sources.

3. For immediate implementation of MBAC for controlled-load services, we recommend the following algorithm: $V' < v\mu - kr$ Where, V' is a utilization factor, μ is the link bandwidth, $k > 0$ a constant, r , the reserved rate of an incoming flow.

The performances of these algorithms, while somewhat insensitive to the form of the admission control equations, appears rather sensitive to changes in the parameters controlling the measurement process. Types of Measurement Based Admission Control algorithms (MBAC):-1. Measured Sum: -It uses measurement to estimate the load of existing traffic. This algorithm admits the new flow if the following test succeeds:

$V' + r < v\mu$ Where v is a user-defined utilization target as explained below, and V' the measured load of existing traffic.

3 Acceptance Region tangent at origin:-It computes

an acceptance region that maximizes the reward of utilization against the penalty of packet loss. Given link bandwidth, switch buffer space, a flow's token bucket filter parameters, the flow's burstness, and desired probability of actual load exceeding bound, one can compute an acceptance region for a specific set of flow types, beyond which no more flow of those particular types should be accepted.

4 Acceptance region tangent at peak:-A new flow is

admitted by the network if the condition stated under satisfies $p(1 - e^{-sp}) + e^{-sp} > \mu$ 4. Hoeffding Bounds (HB):-It computes equivalent bandwidth for a set of flows using the Hoeffding bounds. The equivalent bandwidth of a set of flows is defined in references as the bandwidth $C(\epsilon)$ such that the stationary bandwidth requirement of the set of flows exceeds this value with probability at most ϵ (called as loss rate in this paper).

In an environment where large portion of traffic is best-effort traffic, real time traffic rate exceeding its equivalent bandwidth is not lost but simply encroaches upon best-effort traffic. In reference the measurement based equivalent bandwidth based on Hoeffding bounds (Ch) assuming peak rate (p) policing of n flows is given by: $(Ch)(v, \{p_i\}_{i=1}^n, \epsilon) = v' + \frac{1}{2} (\ln(1/\epsilon) \sum_{i=1}^n p_i^2) / 2$ Where, V' is the measured average arrival rate of existing traffic, ϵ is the probability that arrival rate exceeds the link capacity. It indicates that the measured average arrival rate may be approximated by measured average load.

5 III.

6 NEW PROPOSED ALGORITHM

In this priority based algorithm, priority is decided using the ROUND TRIP TIME (RTT) of various nodes. Node with highest priority i.e. least RTT is allowed to admit flow first. To calculate RTT a source node sends a ping packet to a receiver and track the sending time. The receiver sends a packet back and the sender calculates RTT

100 from tracked sending time. The changes in certain files are? packet.h ? tcl/lib/nsdefault.tcl ? tcl/lib/nspacket.tcl
101 ? ping.tcl
102 procedure recv is defined which calculate the RTT of each node. Now in a network a new flow is admitted, first
103 priority of the nodes is checked for the conflicting flows and node having the highest priority i.e. the minimum
104 RTT of the conflicting flow is admitted to the network given that the bandwidth is available to admit the flow.
105 To deal with the problem of starvation in the network time stamping can be used. In this if a unique time stamp
106 is attached to each flow. And if the flow is rejected then its time stamp is increased by 1. So all the flow can be
107 admitted and no flow can go with starvation. Here, epsilon is the probability that arrival rate exceeds the link
108 capacity and is set to 0.7 S is the Sampling period which defines the sensitivity of exponential averageing; w is
109 the weigh function. Here, S is the Sampling period s is the period of point sampling which cannot be greater
110 than 1

111 7 V. CONCLUSION

112 Graphs were drawn comparing actual and estimated bandwidth utilization. The red line show estimated
utilization and green shows the actual ^{1 2}



1

Figure 1: Flowchart Algorithm:- 1 .

113

¹© 2011 Global Journals Inc. (US)

²August

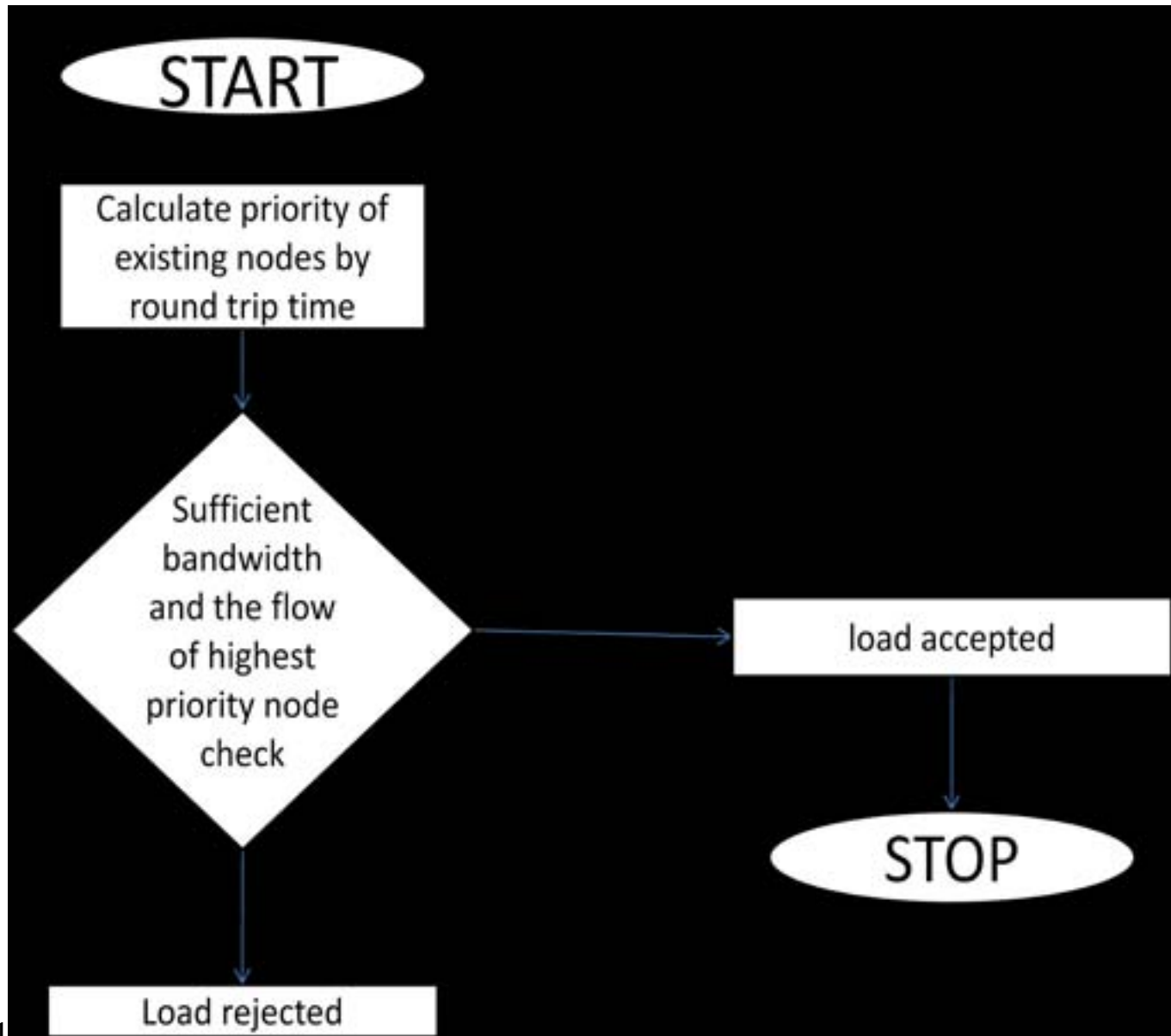
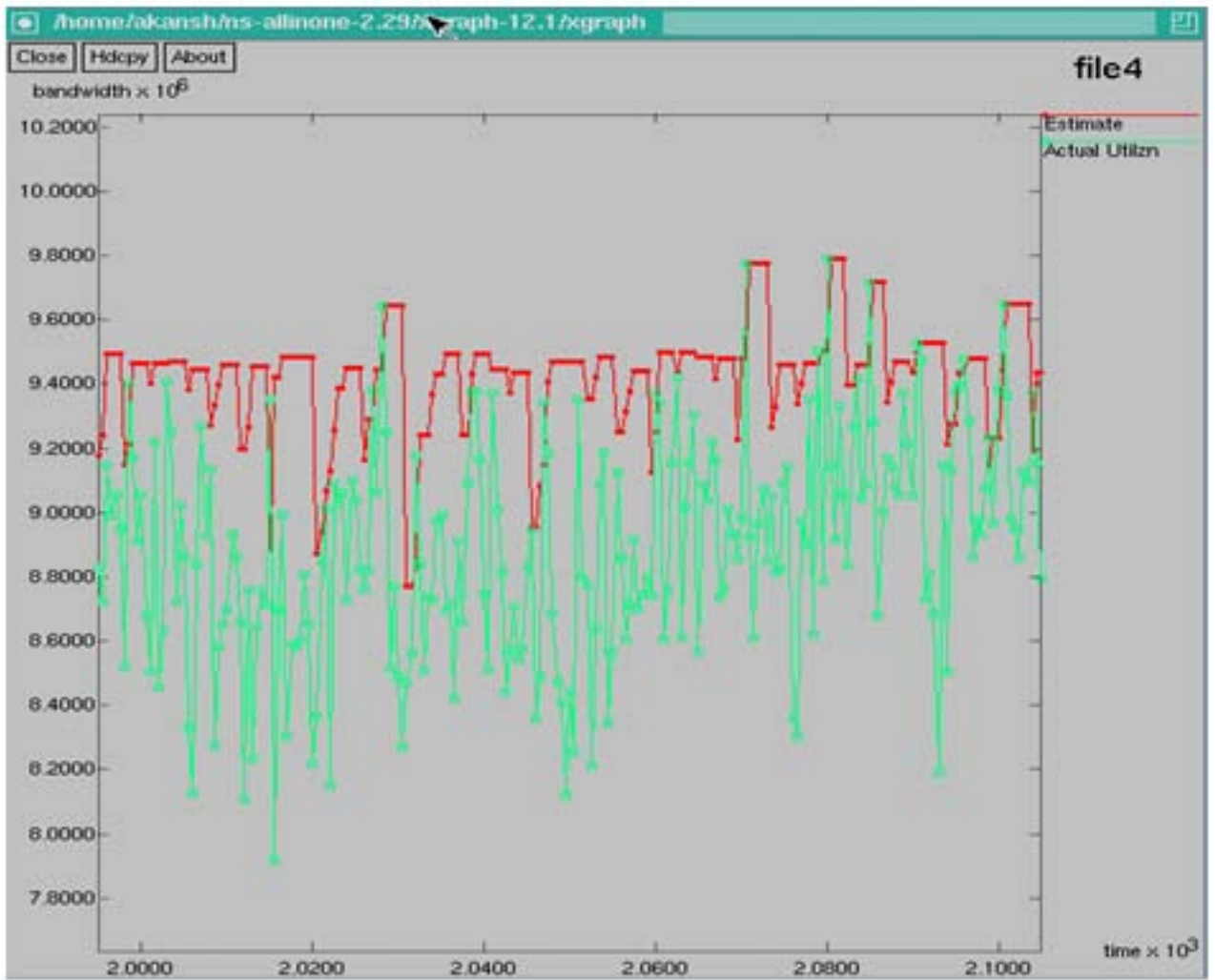


Figure 2: Fig 1 .



13

Figure 3: Fig 1 . 3 :

amount of network resources required to a set of flows.

2. Measurement based approach; it relies on the measurement of actual traffic load in making admission decisions. Admission Control types:-

1. Distributed approach a. Measurement based Admission Control method. 2. Centralized approach
- a. Flow reservation and admission control.
- b. Call admission control for contention access mechanism.
- c. An admission control strategy for differentiated services.

The performance of four admission control algorithms -one parameter based and three measurements (one parameter based, and equivalent bandwidth) -for controlled load service is compared.

?

bas(e)bandwidth,

Figure 4:

114 .1 II. ROLE OF ADMISSION CONTROL

115 There are two basic approaches to admission control: 1. Parameter based approach; it computes the The file
116 ping.tcl is executed which contain the procedure to calculate the round trip time of each node. The four node
117 architecture is created in the file and the ping packets are sent to each node one by one. The utilization. All the
118 four graphs were prepared to justify the algorithms prepared and compared them for the different video packets.
119 The utilization of bandwidth and the packet drops were checked and compared.

120 The performance for each of the admission control algorithms above is calculated by the measuring the actual
121 link utilization and the drop rate. x-graph plots a snap shot of actual and estimated bandwidth utilized in the
122 period [2000, 2100] seconds at the end of the simulation. Also if you set the trace_flow flag to 1, the output
123 would indicate times at which flows come in and leave.

124 The HB algorithm gives the best utilization for smaller packet size and the drop rate of packet is minimum
125 for ACTO algorithm.

126 For the transmission of packet of size 1250 Bytes the ACTO algorithm give the best bandwidth utilization
127 .The ACTP algorithm gives near about the same utilization. But the packet drop rate in MS and HB algorithm
128 is near to zero.

129 .2 August

130 So finally we conclude that the algorithm HB is best for smaller packets but as the packet size increases and the
131 algorithms ACTO and ACTP gives the best bandwidth utilization.

132 [J. Select. Area. Commun ()] , *J. Select. Area. Commun* 1990. 8 (3) p. .

133 [Ferrai and Verma] *A scheme for real time channel establishment in wide area networks*, D Ferrai , D C Verma
134 . IEEE.

135 [Ohnishi et al. (1988)] 'Flow control schemes and delay/loss tradeoff in ATM networks'. H Ohnishi , T Okada ,
136 K Noguchi . *IEEE J Select Areas Commun* Dec. 1988. 6 (9) p. .

137 [Zhang and Ferrari ()] 'Improving utilization for deterministic service in multimedia communication, presented
138 at the IEEE Int'. H Zhang , D Ferrari . *Conf. Multimedia Computing and Systems*, 1994.

139 [Braden et al. (1994)] *Integrated Services in the Internet Architecture: an Overview*, R Braden , D Clark , S
140 Shenker . RFC 1633. June 1994.

141 [Duffy et al. ()] 'Statistical analysis of CCSNSS7 traffic data from working CCS subnetworks'. D E A A Duffy ,
142 M McIntosh , W Rosenten , Willinger . *IEEE J Select. Areas Commn* 1994. 12 (3) p. .

143 [Clark et al.] 'Supporting real time applications in an integrated services packet network; architecture and
144 mechanism'. D D S J Clark , L Shenker , Zhang . *Proc. ACM SIGCOMM' 92*, (ACM SIGCOMM' 92)
145 p. .

146 [Saito] *Teletraffic Technologies in ATM Networks*. Artech House, Hiroshi Saito .