PERFORMANCE CONSIDERATIONS ON ADMISSION CONTROL FOR MULTIMEDIA SERVICES

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Abstract: Admission control represents a convenient mechanism to provide high-quality communication by ensuring resource availability. This paper gives an overview on different measurement-based admission control algorithms suitable to be applied in multimedia service environments. A new estimator used for the measurement process is introduced, which dynamically changes the time window used for measurements. The performance metrics of interest within the performance analysis are made up of average utilization, packet loss and percentage of admitted flows.

1 INTRODUCTION

Network management techniques have been of interest to the networking research community for a long time. They involve both data control and maintaining a general controlled state throughout the network by providing QoS guarantees. The provision of QoS-controlled service requires the coordinated use of admission control, traffic access control, packet scheduling, and buffer management. Other techniques include flow and congestion control and QoS routing.

Within service management, admission control provides a mean in fulfilling the contracted service level agreement (SLA) between the user and the network provider. As a preventive congestion avoidance mechanism, it attempts to make best use of the finite link capacity across a network by admitting a new flow of data into a network without impacting the guarantees of existing data flows. It can be realized in several ways. Generally we differentiate between distributed and centralized approaches. The admission decision of distributed approaches can be based on particular parameters of signalling messages (e.g. peak rate or mean rate) or on measurements.

A particular method for realizing an admission control decision is by performing measurements upon arrival of a new admission request. These schemes usually comprise of two phases - the measurement phase performed by the estimator and the admission phase performed by the policy rules. Several approaches have been investigated to date and were reported in some valuable contributions such as (Jamin et al., 1997), (Breslau et al., 2000), (Kelly, 2000), (Casetti et al., 1996).

This paper provides a performance analysis on measurement-based admission control for multimedia services. In Section 2 an overview of several estimators and admission policy schemes are presented. Section 3 introduces a dynamic estimator, as well as its performance analysis. Finally, Section 4 gives some concluding remarks.

2 MEASUREMENT-BASED ADMISSION CONTROL

There are several admission control approaches in the literature, and no standardized method for the use on a particular network. Data packet measurement-based admission control (also referred to as passive MBAC) is a group of admission control algorithms, which usually measures the actual traffic load and performs the AC function using the estimation value based on the current measured traffic volume. For this purpose the authors of (Jamin et al., 1997) have evaluated different AC algorithms and compared their performance. Further, (Casetti et al., 1996) describes an adaptive admission control algorithm based on measurements, which is the base for our observations.

A measurement-based admission control algorithm can be divided into a measurement-based estimator and an admission policy component. The pol-
2.1 Policy Algorithms

The measured sum algorithm (MS) uses measurement to estimate the load of existing traffic. Let $\mu$ be the link bandwidth, $\alpha$ the new flow requesting admission, and $r^\alpha$ the rate requested by flow $\alpha$. The new flow is admitted if the following test succeeds:

$$\hat{v} + r^\alpha < c \mu$$  \hspace{1cm} (1)

where $c$ is a user-defined utilization target and $0 < c < 1$. The measured load of existing traffic is denoted with $\hat{v}$. Upon admission of a new flow, the load estimate is increased using:

$$\hat{v}' = \hat{v} + r^\alpha$$  \hspace{1cm} (2)

A measurement-based approach is doomed to fail at very high utilization when delay violations become exceedingly large. It is thus necessary to identify a utilization target and require that the algorithm strives to keep link utilization below this level.

The acceptance region algorithms compute an acceptance region that maximizes the reward of utilizing against the penalty of packet loss. These algorithms are based on Chernoff bounds. Given link bandwidth, switch buffer space, a flow’s token bucket filter parameters, the flow’s burstiness, and desired probability of actual load exceeding bound, an acceptance region can be computed for a specific set of flow types, beyond which no more flow of those particular types should be accepted.

Based on different combinations of measured and declared parameters, four related techniques based upon Chernoff bounds are presented in (Gibbens and F.P.Kelly, 1997). The availability and ease of measurement extractions (e.g., per-flow vs. aggregate) and the need for a priori traffic declarations (e.g., average rate as well as peak rate) will each affect the relative practicability of the four approaches, namely: tangent at peak (ACTP), tangent at arbitrary location, tangent at slope one, tangent at origin (ACTO). Table 1 illustrates basic features of these four algorithms.

For a better overview let us illustrate the computation of the effective bandwidth requirement of the traffic aggregate (all classes added together) for the tangent at slope one algorithm:

$$\hat{v} = X + \frac{C^{K-1}}{4} \sum_{k=0}^{K} p_k^2 n_k$$  \hspace{1cm} (3)

where, $\hat{v}$ is the estimate for traffic load, $K$ is the number of different flow types, $n_k$ is the number of individual flows of a particular type, $p_k$ represents the peak-rate for a particular flow type, $X$ is the measured aggregate utilization, and $C$ is a scaling factor. For admission decision this estimated aggregate load should be smaller or equal to link capacity $\mu$.

The Hoeffding bounds scheme is in fact the computation base for the equivalent bandwidth algorithm (Guerin et al., 1991). It sets a probability threshold on the sum of the source transmission rates.

$$C(\varepsilon) = r_S + \sqrt{\frac{\ln(1/\varepsilon) \sum_k p_k^2}{2}}$$  \hspace{1cm} (4)

where, $C(\varepsilon)$ represents the equivalent bandwidth, $r_S$ is the average aggregate arrival rate, $p_k$ is the peak rate, $\varepsilon$ is the target loss rate and $k$ is the number of flows. When a new flow $\alpha$ requests admission, the admission control check is then based on this criterion:

$$C(\varepsilon) + p_\alpha \leq \mu$$  \hspace{1cm} (5)

2.2 Estimators

In order to be able to maintain a level of service or guarantee of QoS, the algorithm must have available an estimate of current resource requirements, typically bandwidth requirements. Bandwidth estimation may be based upon predictive traffic models, measurements, or a combination of both. Those based on measurements are of interest for this study.

As shown in Figure 1 an average load is computed for every sampling period $S$ with the time window estimator, where $S$ represents an integer number of stochastic packet transmission times. At the end of a measurement window $T$, which is an integer number of sampling periods $S$, the highest average encountered within the window is used as the load estimate for next window $T$. Additionally, whenever a new flow is admitted to the network, the estimate is increased according to the advertised flow information (e.g., peak rate of the requesting flow), and the window is restarted. The estimate is also increased.

<table>
<thead>
<tr>
<th>Acceptance region scheme</th>
<th>Measurement</th>
<th>Per-class declaration</th>
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<tbody>
<tr>
<td>Tangent at peak</td>
<td>Per-class measurements</td>
<td>Peak rate</td>
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<tr>
<td>Tangent at arbitrary location</td>
<td>Per-class measurements</td>
<td>Peak rate, Average rate</td>
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<tr>
<td>Tangent at slope one</td>
<td>Aggregate (line) measurements</td>
<td>Peak rate</td>
</tr>
<tr>
<td>Tangent at origin</td>
<td>Aggregate (line) measurements</td>
<td>Peak rate</td>
</tr>
</tbody>
</table>

Table 1: Characteristics of acceptance region schemes.
immediately if a newly measured average is higher than the current estimate.

As the name suggests, the point samples measurement mechanism used with the acceptance region algorithm takes an average load sample every sampling period $S$.

For exponential averaging, an estimate of the average arrival rate can be used instead of instantaneous bandwidth to compute admission decision with the Hoeffding bounds approach. The average arrival rate $r_\alpha$ is measured once every sampling period $S$. The average arrival rate in then computed using an infinite impulse response function with weight $w$ (e.g., 0.002):

$$\hat{\nu}' = (1 - w)\hat{\nu}' + wr_\alpha$$

where, $r_\alpha$ is the average arrival rate, $\hat{\nu}$ represents the measured load of existing traffic, $S$ is the sampling period, and $w$ is the weight.

### 2.3 Performance Comparison

Table 2 provides the number of transmitted vs. dropped packets for different estimator - policy pairs. This comparison as well as the analysis on output utilization from (Statović-Halimi, 2008a), prove the measurement sum algorithm to reveal the best performance at least cost. The conclusion on least cost is based on the simplicity of this algorithm in comparison to the other three evaluated ones. This algorithm is used together with the time-window estimator.

<table>
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<tr>
<th>Admission control algorithms</th>
<th>Transmitted packets</th>
<th>Dropped packets</th>
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</thead>
<tbody>
<tr>
<td>ACTP with PointSample</td>
<td>26217923</td>
<td>150</td>
</tr>
<tr>
<td>ACTO with PointSample</td>
<td>25906454</td>
<td>35</td>
</tr>
<tr>
<td>HB with ExpAvg</td>
<td>26993894</td>
<td>2038</td>
</tr>
<tr>
<td>MS with TimeWindow</td>
<td>26127938</td>
<td>120</td>
</tr>
</tbody>
</table>

### 3 A DYNAMIC TIME-WINDOW ESTIMATOR

The idea of our estimator is the dynamic adjustment of the time window size to the changing traffic requirements. The main principle of this algorithm is based on the attempt of avoiding the use fixed-length measurement windows, as traffic characteristics are usually unknown or can vary. By means of enlarging or shrinking the measurement window, the adaptation to the changing traffic conditions can be provided, so as to obtain a more or less conservative admission process (Statović-Halimi, 2008b).

The algorithm comprises two phases. The first phase represents the measurement procedure. In order to increase link utilization, the length of the measurement window is here continually shrunk with a factor $f_{\text{small}}$ until the amount of traffic generated by accepted requirements reaches a trigger value, which is smaller than the output link capacity. As a reaction, the algorithm then enlarges the measurement window with a factor $f_{\text{large}}$ until the measured rate drops below the trigger, at which point the window can be shrunk again. This process changes according to the variable traffic conditions. The second phase of the algorithm the outputs of the first phase in order to adjust the trigger value according to traffic fluctuations.

#### 3.1 Simulation Results

For evaluation purposes network simulations are performed with ns-2, using three source models. According to conclusions of (Statović-Halimi, 2008a), our dynamic estimator is used in combination with the measured sum algorithm.

![Figure 2: Dependency of utilization from time window.](image-url)

The simulation environment comprises a two-node topology, and the total number of flows is 9986. The voice over IP traffic uses an exponential ON/OFF
source with transmission peak rate of 64kbit/s, packet size of 200 and idle time of 325ms. Video traffic is simulated by an exponential ON/OFF source with an inter arrival time of 0.1s, exponential holding time of 100s, and video holding time 300s. Background traffic is also applied to the network.

4 CONCLUSIONS

This paper introduces a dynamic time window estimator, which is based on load measurements and adjustment of the time window size. In general, a smaller measurement window \( T \) yields a higher utilization at higher loss rate and a larger \( T \) keeps more reliable loss rates at the expense of utilization level. The source burstiness also gives differences in this context. Extreme traffic fluctuation is more difficult to handle under tight guarantees, so the tradeoff between accuracy and automate configuration is a relevant factor when deciding upon the implemented approach.

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REFERENCES


