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Dynamically forecasting network performance using the Network Weather Service *

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The *Network Weather Service* is a generalizable and extensible facility designed to provide dynamic resource performance forecasts in metacomputing environments. In this paper, we outline its design and detail the predictive performance of the forecasts it generates. While the forecasting methods are general, we focus on their ability to predict the TCP/IP end-to-end throughput and latency that is attainable by an application using systems located at different sites. Such network forecasts are needed both to support scheduling (Berman et al., 1996) and, by the metacomputing software infrastructure, to develop quality-of-service guarantees (DeFanti et al., to appear; Grimshaw et al., 1994).

1. Introduction

As network technology advances, the resulting improvements in interprocess communication speeds make it possible to use interconnected but separate computer systems as a high-performance computational platform or *metacomputer*. Effective application scheduling (particularly of distributed parallel applications) is fundamental if such metacomputers are to be used successfully. Since the resources composing a metacomputer are shared, contention causes their load and availability to vary over time. As a result, the performance that each resource can deliver to an application also varies with time.

In this paper, we describe a distributed service that dynamically forecasts the performance various networked resources can deliver to an application. The service operates a distributed set of sensors from which it gathers readings of the instantaneous conditions. It then uses numerical models to generate forecasts of what the conditions will be for a given time frame. We think of this functionality as being analogous to weather forecasting, and as such, term the service the *Network Weather Service* (NWS).

We have developed the NWS for use by schedulers in a networked computational environment. The AppLeS scheduling methodology [1,3] makes extensive use of its facilities and we are currently implementing versions for Legion [17,25] and Globus/Nexus [9,15]. Initial scheduling results using the NWS are promising [4]. In this paper, we focus on the problem of network performance forecasting within the context of scheduling, and the predictive methodologies that we have chosen to explore initially. We have developed a prototype of the NWS that forecasts network performance (latency and bandwidth) and available CPU percentage for each machine that it monitors.

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The forecasting methods we have implemented fall into three categories:

- *mean-based* methods that use some estimate of the sample mean as a forecast,
- *median-based* methods that use a median estimator, and
- *autoregressive* methods.

To gauge the effectiveness of each method, we report both the mean square prediction error, and the mean percentage prediction error generated by each method as accuracy measures. While mean-based predictive methods generally yield lower mean square error measures, and median-based methods are better in terms of mean percentage error, the best forecasting technique for each setting is difficult to predict. The system, therefore, tracks the accuracy (using prediction error as an accuracy measure) of all predictors, and uses the one exhibiting the lowest cumulative error measure at any given moment to generate a forecast. In this way, the NWS automatically identifies the best forecasting technique for any given resource.

In the next section (section 2), we briefly describe the structure and implementation of this prototype. Section 3 describes the sensory mechanisms and section 4 describes the forecasting methods we have currently implemented. In section 5 we compare NWS measurements with their corresponding forecasts for several different network environments. We conclude in section 6 with an evaluation of the results, and a description of our future research.

2. Structure and implementation

To serve as a viable tool for scheduling, the Network Weather Service must

- *sense* resource performance throughout the system,
- *forecast* the future performance of each resource, and
- *disseminate* the forecast information to all interested client schedulers.

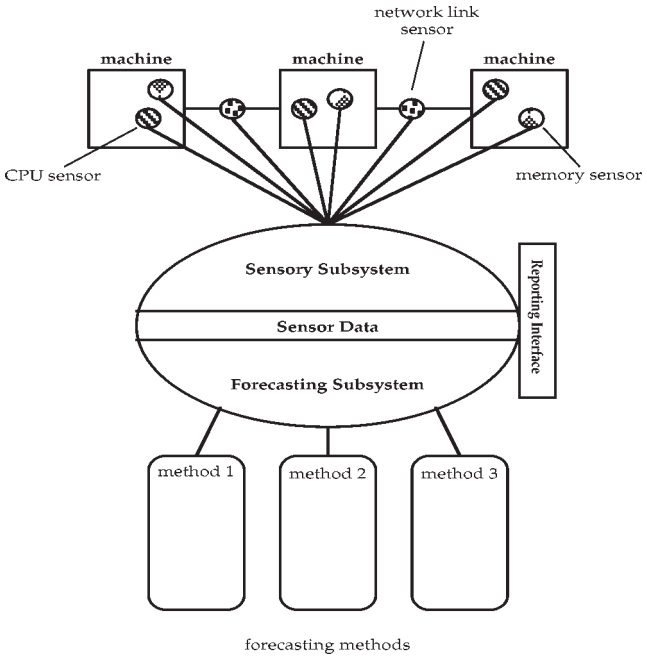


Figure 1. The structure of the Network Weather Service.

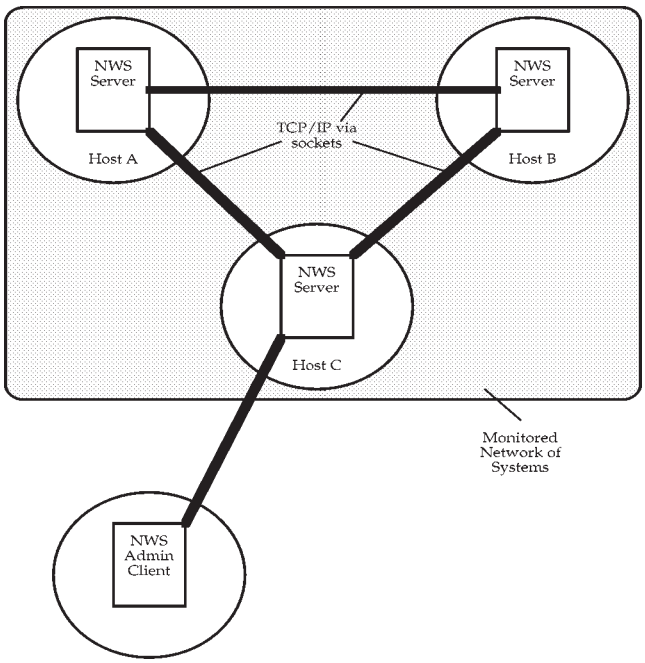


Figure 2. NWS servers running on three monitored hosts.

Unlike the real-world weather service, however, the operation of the NWS can significantly change the conditions which it is attempting to forecast. Moreover, we do not assume that computational or network resources within the system will be devoted exclusively to the NWS as such resources would constitute possible failure points and performance bottlenecks. Our view, instead, is that the NWS must limit its intrusiveness so that its resource consumption does not adversely impact the performance of the applications it is designed to serve. The need to limit the intrusiveness of the NWS influences both the implementation of the overall system and the forecasting techniques we have chosen. Since the problems of non-intrusive resource monitoring [8,21,28] and load forecasting [2,7,18,23,26] both pose open research questions, we have separated the sensory and forecasting functions of the NWS. The resulting modular design is intended to provide a general facility in which a variety of different monitoring and forecasting techniques can be employed easily. Figure 1 depicts the architecture of the system.

Sensory data is compiled into a logically central (although physically distributed) database to serve as inputs to a collection of forecasting models. Each sensor periodically takes a performance measurement from the resource it is monitoring and stores it with a time stamp in the database. The resulting collection of measurements (ordered by time stamp) form a time series describing the behavior of the resource from which they were taken. That is, each resource is characterized by its own time series. Based on this information history and any *a priori* knowledge of a resource's expected performance response, the NWS forecasting subsystem generates a prediction of what the performance will be for each resource during a given time frame. Once generated, the current forecast data, along with quality measures describing its accuracy (i.e., mean square prediction error, mean percentage prediction error, etc.) are published according to the specifications of an independent reporting interface. Each of the sensory, forecasting, and reporting functions is implemented as a separate subsystem providing a generalizable facility that is easily extended and modified.

3. Sensing

NWS sensors report the observed performance that a resource is able to deliver at the time a measurement is taken. A network link sensor, for example, reports periodic measurements of latency and bandwidth across a particular link. Measurements are taken as close to the application level as possible, since the goal is to forecast the performance an application can actually obtain from each resource. The units of measurement that each sensor uses depend, in general, on the demands of the forecasting models[1].

Each machine to be monitored must execute a copy of the NWS server. Figure 2 depicts three monitored hosts, A, B and C within the monitored region (shown shaded in the figure). These hosts each run a copy of the NWS server, each of which is capable of establishing and maintaining a connection (TCP/IP in the current implementation) between itself and the others. An administrative client utility that controls the system may execute on any machine inside or

[1] We have based an initial sensory subsystem implementation on the TCP/IP socket functionality provided by the *netperf* network performance utility, although we have modified the code substantially. Netperf proved to be a robust and powerful substrate for our purposes, and we encourage those interested to visit the netperf World Wide Web site for further information, see [29].

outside the monitored network, requiring connectivity to at least one of the NWS servers.

Currently, each server maintains a network performance sensor and a CPU availability sensor. All servers in the system share a common list of hosts being monitored, and the TCP port number to which each server is attached. Periodically, each server chooses a host from the list and conducts a communication "experiment" with that host. During an experiment, the round-trip time of a single-word packet is measured. The resulting value is divided by two to yield an approximation of the latency or start-up overhead associated with a communication. Immediately after the latency has been estimated, the initiating server sends a predetermined (and parameterizable) quantity of data and times the transfer. Throughput is then calculated as the data size divided by the transfer time.

$$\textit{throughput} = \textit{data size}/\textit{data transfer time}. \tag{1}$$

The resulting measure includes the overhead necessary to initiate a TCP/IP communication stream, which can be significant. To calculate the effective throughput rate, the latency is subtracted from the time recorded for the data transfer, and the result is used as the actual time to transfer the data.

$$\textit{effective\_throughput} = \textit{data size}/(\textit{data transfer time} - \textit{latency}). \tag{2}$$

After an experiment is complete, the latency, throughput, and effective throughput are recorded in an internal database.

Each server also periodically records the local CPU availability using the Unix utilities *vmstat* or *uptime*. If vmstat is available, the sensor parses its output to determine the percentage of time the system is idle, the time spent (by the machine) executing in system space, the percentage of time spent executing user processes, and the number of running processes. Using these measures, it estimates the percentage of time the system is willing to devote to a single application. If *vmstat* is not available, then the CPU monitor uses uptime in the manner described by the local Unix utility reference manual.

3.1. Periodicity

The periodicity with which the NWS sensors take measurements is synchronized by an external administrative client process. Notice that if communication experiments are conducted strictly under the control each server's local clock, two or more servers might choose to test the same link simultaneously. Moreover, the NWS servers work with a logical, application-level picture of the network's topology – they do not consider which links might share a common medium requiring exclusive access (such as ethernet), and which ones might be able to handle simultaneous communication without interference. The servers, therefore, pass around a token which contains the right to conduct a single experiment. When holding the token, each server is free to choose the experiment that it wishes to conduct. The token assures that at most one server may be conducting a communication experiment at any given time. A central utility operated by the NWS administrator controls the rate at which the token may be passed from server to server. By controlling that rate, the administrator can control the overall periodicity of the communication measurements.

We note that this method of controlling intrusiveness does not scale well. Since one token is active within the system at a time, and the token must be handled by each system, the number of systems being monitored imposes a maximum frequency with which measurements can be taken. For small collections of machines it is adequate, but for larger-scale systems such as the Distributed Object Computational Testbed (DOCT) [11] or the I-way [10,24] we will need another mechanism.

3.2. Storage requirements

Storage intrusiveness is also an issue. Some of the forecasting methods discussed in section 4 require a history of measurements. In order to bound the storage requirements of the system, we limit the number of measurements each server is allowed to maintain to a fixed, but parameterized quantity. Further, each server is responsible for maintaining a history of the measurements it takes. History data can be collected from the servers by a utility (i.e., for human display purposes) but the data remains distributed until it is demanded.

4. Forecasting

The NWS operates a set of forecasting methods that it can invoke dynamically, passing as parameters the performance measurements it has taken from each resource. In this section we describe the methods we have included in the current implementation. After each new measurement is taken, it is passed to all of the methods, and a new forecast is generated. That is, for each forecasting method f at measurement time t,

$$\textit{prediction}_f(t) = \textit{METHOD}_f\big(\textit{value}(t), \textit{history}_f(t)\big), \tag{3}$$

where

$\textit{value}(t) =$ the measured value at time t,
$\textit{prediction}_f(t) =$ the predicted value made by method f for measurement $\textit{value}(t+1)$,
$\textit{history}_f(t) =$ a finite history of measurements, forecasts, and residuals generated previously to time t using method f,
$\textit{METHOD}_f =$ inforecasting method f.

The values supplied by the sensory subsystem are treated as a time series by the forecasting methods, and each method

maintains a history of previous activity and accuracy information. In particular,

$$err_f(t) = value(t) - prediction_f(t-1) \tag{4}$$

is the error residual associated with a measurement and a prediction of that measurement generated by method f.

The current implementation generates a forecast every time a measurement is taken. Since each method is evaluated whenever a new datum is available, we restrict ourselves to those methods we can implement with limited computational complexity. However, an alternative implementation we are considering generates forecasts only when they are requested by a client; this approach may make more computationally complex methods feasible.

4.1. Mean-based methods

One class of predictors that we have investigated uses arithmetic averaging (as an estimate of the mean value) over some portion of the measurement history to predict the value of the next measurement. The running average, defined as

$$RUN\_AVG(t) = \frac{1}{t+1}\sum_{i=0}^{t} value(i), \tag{5}$$

uses the average of the measurement taken at time t with all previous measurements as a predictor of the measurement to be taken at $t+1$.

Since the running average considers the entire history of measurements when making each forecast, the weight given to each measurement decreases linearly with time. If the most recent values better predict the next measurement, then an average taken over a fixed-length history (thereby fixing the weight given to each measurement) will be a better predictor. The fixed-length or "sliding window" average is calculated as

$$SW\_AVG(t, K) = \frac{1}{K+1}\sum_{i=t-K}^{t} value(i), \tag{6}$$

where $K \geqslant 0$ is an integer specifying the number of samples to consider in the window. Note that for $K = 0$, *SW_AVG* uses the last measurement only as a predictor. That is,

$$LAST(t) = SW\_AVG(t, 0). \tag{7}$$

Recent work by Harchol-Balter and Downey [20] indicates that this is a useful predictor for CPU resources, hence we include it as a separate method.

The choice of K for *SW_AVG* may be difficult to determine *a priori* for each resource, and in fact, may vary over time. To set K dynamically so that it adapts to the time series, we employ a gradient-descent strategy. Let $K(t)$ be the value of K at time t, and

$$err_i(t) = \big(value(t) - SW\_AVG\big(t, K(t)+i\big)\big)^2.$$

Then we define

$$ADAPT\_AVG(t) = \begin{cases} SW\_AVG(t, K(t)-1) \\ \quad \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_{-1}(t), \\ SW\_AVG(t, K(t)) \\ \quad \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_0(t), \\ SW\_AVG(t, K(t)+1) \\ \quad \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_1(t) \end{cases} \tag{8}$$

and

$$K(t+1) = \begin{cases} K(t)-1 & \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_{-1}(t), \\ K(t) & \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_0(t), \\ K(t)+1 & \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_1(t). \end{cases} \tag{9}$$

The value of K is adjusted at each time step in the direction that yields the lowest error. We use a measure of the square error in $err_i(t)$ arbitrarily. It is also possible to use a measure of the absolute percentage error, but our initial experiments indicate that the results are similar. Note that the value of $K(t)$ must be carried as part of the history for *ADAPT_AVG*, and that $K(0)$ is set to some reasonable starting value. We also arbitrarily restrict K to be between a predetermined maximum and minimum. Setting a maximum threshold limits the computational complexity of the predictor; the minimum value prevents it from becoming "stuck" in a local minimum. In the experiments presented in the next section, we set $5 \leqslant K \leqslant 50$.

Stochastic gradient or recursive prediction error estimators are powerful predictive techniques with recursive formulations [27]. For example, modern implementations of the TCP/IP protocol include a dynamic predictor of end-to-end round-trip time based on stochastic gradient filter [30]. We follow the exposition of the technique provided in [22] which includes a description of a very efficient implementation for the Unix kernel. We define

$$GRAD(t, g) = (1-g)\cdot GRAD(t-1, g) + g \cdot value(t) \tag{10}$$

for a *gain* ($0 < g < 1$). The choice of g controls the accuracy with which *GRAD* estimates the mean value of the time series and the lag time until it converges to a stable estimate. *GRAD* oscillates randomly about the true average with a standard deviation $\sigma_{GRAD} = g \cdot \sigma_{value(t)}$. Hence a larger value of g will yield a more widely varying estimate. However, *GRAD* converges exponentially with time constant $1/g$ to the true mean. If the time series is not stationary, *GRAD* must reconverge as the mean moves. The convergence rate must be faster than the drift in the mean or the predictor will fail to converge. Empirically, a value of 0.05 works well, although we expect to study the problem of finding an appropriate g further. We have experimented with techniques to dynamically adapt g on the fly, but have yet to identify an effective method for the resources we currently monitor with the NWS.

4.2. Median-based methods

The median value can also serve as a useful predictor, particularly if the measurement sequence contains randomly-occurring, asymmetric outliers. Our presentation of these techniques follows the exposition in [12,19]. The median over a sliding window of fixed length whose leading edge is the most recent measurement is used as the forecast for the next measurement. That is, we define

$Sort_K$ = the sorted sequence of the K most recent measurement values,
$Sort_K(j)$ = the jth value in the sorted sequence,

$$MEDIAN(t,K) = \begin{cases} Sort_K((K+1)/2) \\ \quad \text{if } K \text{ is odd,} \\ (Sort_K(K/2) + Sort_K(K/2+1))/2 \\ \quad \text{if } K \text{ is even.} \end{cases} \tag{11}$$

As with *SW_AVG* the choice of K may be difficult to determine. We, therefore, include an adaptive median filter that is analogous to *ADAPT_AVG*:

$$ADAPT\_MED(t) = \begin{cases} MEDIAN(t, K(t)-1) \\ \quad \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_{-1}(t), \\ MEDIAN(t, K(t)) \\ \quad \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_0(t), \\ MEDIAN(t, K(t)+1) \\ \quad \text{if } \min\limits_{i=-1,0,1} err_i(t) = err_1(t), \end{cases} \tag{12}$$

where $K(t)$ is the value at time t and

$$err_i(t) = \big(value(t) - MEDIAN\big(t, K(t)+i\big)\big)^2.$$

$K(t+1)$ is then determined by equation (9).

Median filters are attractive because they will reject the effects of sharply outlying data points or "impulses" from the forecasts they produce. They lack some of the smoothing power of the averaging based methods, however, resulting in forecasts with a considerable amount of jitter [19]. It is possible to combine the positive advantages of both classes of methods in the form of an α-trimmed mean filter that averages the central $K - 2 * \alpha * K$ values within a sliding window of size K for $(0 < \alpha < 0.5)$. We define

$$T = \lfloor \alpha \cdot K \rfloor$$

for window size K, and the trimmed mean to be

$$TRIM\_MEAN(t,K,T) = \sum_{j=T+1}^{K-T+1} \frac{1}{K-2\cdot T} Sort_K(j). \tag{13}$$

It is possible to consider gradient adaptation of α in the same manner that we adapt K for *ADAPT_AVG* and *ADAPT_MED* but the relationship between α and K is not obvious.

4.3. Autoregressive models

Recent work [2,18] has shown that aggregate internet packet traffic can be effectively modeled by autoregressive, integrated, moving average (ARIMA) models. Fitting these models to a specific time series requires the solution to a system of potentially non-linear simultaneous equations, making them difficult to use in a dynamic setting. However, fitting a purely autoregressive (AR) model requires only the solution to a strictly linear system of equations that can be solved recursively via the Levinson Recursion [19]. The general form of a pth order autoregressive model is

$$AR(t,p) = \sum_{i=0}^{p} a_i \cdot value(t-i). \tag{14}$$

If the time series is stationary, then the sequence $\{a_i\}$ that minimizes the overall error can be determined by the solution to the linear system

$$\sum_{i=0}^{N} a_i \cdot r_{i,j} = 0, \quad j = 1,2,\ldots,N, \tag{15}$$

where $r_{i,j}$ is the autocorrelation function for the series of N measurements taken. The Levinson Recursion requires a set of partial correlation (PARCOR) coefficients which can also be derived recursively. Burg [5] and more recently Haddad and Parsons [19] describe a recursive algorithm for calculating both the PARCOR and autoregression coefficients from which we derive our current implementation. We omit the details of the algorithm here due to space constraints, but our implementation follows [19] closely. The algorithm takes time O($p \cdot N$) for N measurements, which becomes prohibitive when N is the length of the entire time series. We, therefore, calculate the $\{a_i\}$ coefficient sequence over a sliding window of the K most recent measurements, rather than the entire series of size N. That is, after each measurement is taken, we recompute the autoregressive coefficients $\{a_i\}$ using only the previous K measurements as an approximation of the complete time series.

The choice of parameters p and K are determined by the computational complexity the NWS is willing to tolerate. Making K as large as possible (as close to the size of the history as possible) will yield the best fit, but making K too large causes the execution cost of each forecast to be prohibitive. The value of p should be set according to the decay of the autocorrelation function $r_{i,j}$, the values for which are not computed explicitly by the method[2]. Since the autocorrelations can be computationally expensive to compute, we choose arbitrary fixed values of $p = 15$ and

[2] The autoregressive model is applicable if the decay in the autocorrelation function is exponential and the value of p is set to the duration of the decay [16]. Our current implementation of the NWS does not attempt to determine the suitability of *AR* for a particular resource. Instead, it assumes that the autoregressive model is applicable, and tracks the prediction error, using *AR* only if the error is lower than other competing predictors (see section 4.4).

$K = 60$. In future implementations, we plan to derive p algorithmically based on estimates of the autocorrelation values, and K based on p and a maximum computational complexity threshold.

4.4. Dynamic predictor selection

Choosing the correct predictive method for each resource that the NWS monitors is difficult. Further, it may be that a particular resource conforms to the assumptions of one method for a period of time, and then changes its behavior so that it is best modeled by a different method. Rather than attempting to choose the correct method *a priori*, our initial implementation maintains *all* of the predictive methods simultaneously, for each resource. Then it uses the error measure calculated in equation (4) to produce an overall fitness metric for each method. The method exhibiting the best overall predictive performance at any time t is used to generate the forecast of the measurement at time $t + 1$. In the initial implementation of the NWS, we use the mean square prediction error

$$MSE_f(t) = \frac{1}{t+1} \sum_{i=0}^{t} \left(err_f(i)\right)^2 \tag{16}$$

and the mean percentage prediction error

$$MPE_f(t) = \frac{1}{t+1} \sum_{i=0}^{t} \left|err_f(i)\right| / value(i) \tag{17}$$

as fitness metrics for each method f at time t. We then define

$$MIN\_MSE(t) = predictor_f(t)$$
if $MSE_f(t)$ is the minimum over all methods at time t, (18)

and

$$MIN\_MPE(t) = predictor_f(t)$$
if $MPE_f(t)$ is the minimum over all methods at time t. (19)

That is, at time t, the method yielding the lowest mean square prediction error is used as a forecast of the next measurement by *MIN_MSE*. Similarly, the forecasting method at time t yielding the lowest overall mean percentage prediction error becomes the *MIN_MPE* forecast of the next measurement. In a scheduling context, it is unclear which fitness metric – mean square error or mean percentage error – will ultimately yield a better schedule. Indeed, the fitness of each forecasting technique may be application-specific. Therefore, the current system maintains and reports both mean square error and mean percentage error, allowing a specific scheduler to choose either.

Table 1 summarizes the predictive methods we have implemented and the fixed values we have chosen for any parameters required by each method.

Table 1
Summary of forecasting methods.

Predictor	Description	Parameters
RUN_AVG	running average	
SW_AVG	sliding window average	$K = 20$
LAST	last measurement	
ADAPT_AVG	adaptive window average	max = 50, min = 5
MEDIAN	median filter	$K = 20$
ADAPT_MED	adaptive window median	max = 50, min = 5
TRIM_MEAN	α-trimmed mean	$\alpha = 0.1$
GRAD	stochastic gradient	$g = 0.05$
AR	autoregression	$K = 60$, $p = 15$
MIN_MSE	adaptive minimum mse	
MIN_MPE	adaptive minimum mpe	

Table 2
Host locations, types, and operating systems.

Location	Host type	Operating system
UCSD Parallel Comp. Lab*	Sparc10	SunOS 4.1.3
UCSD Parallel Comp. Lab	Sparc5	SunOS 4.1.3
San Diego Supercomputer Center	Alpha	DEC OSF/1 V3.0
California Institute of Technology	Hypersparc	SunOS 5.4
University of Oregon	Power challenge	SGI Irix 6.1
National Center for Supercomputer App.	Power challenge	SGI Irix 6.2

5. Forecasting network performance

In this section, we present measurements and corresponding forecasts of latency and throughput for network connections between machines. During the experimental period, the NWS also monitored and predicted CPU availability using the same forecasting methods. Not surprisingly, network performance proved to be the more difficult of the two to predict as the CPU measurements were slowly varying by comparison. We, therefore, use the network performance data to illustrate the forecasting functionality of the NWS.

We monitored the TCP/IP connectivity between the hosts shown in table 2 using the prototype NWS over a 24 hour period, and dynamically forecast the latency and throughput between each pair of hosts. We chose this collection of systems so that we could study the quality of the existing internet connectivity with respect to geographic proximity. In particular, we were interested in identifying representative examples of connectivity for different plausible metacomputing settings. To do so, we report data on the connectivity between the Sparc10 located in the Parallel Computation Lab (PCL) at UCSD (marked with an "*" in table 2) and the other five systems.

Each of these parings is intended to serve as a representative example. The two PCL machines are connected to the same Ethernet segment representing a *intra-lab* connection. The PCL and the San Diego Supercomputer Center (SDSC) are located approximately one-quarter mile apart on the UCSD campus representing the connectivity in a *campus-wide* setting. Caltech is located in Pasadena Cal-

ifornia, approximately 120 miles north of San Diego representing *intra-state* connectivity. The connection between UCSD and the University of Oregon, located in Eugene, Oregon, represents *inter-state* connectivity, and the connection to the National Center for Supercomputing Applications (NCSA) in Urbana, Illinois, represents *transcontinental* connectivity.

Due to space constraints, we only show data for the Sparc10 sending to each of the other systems. In general, the connectivity between two systems is not symmetric with respect to sending and receiving performance. We believe that this asymmetry results from differences in the operating system implementations on the various machines, rather than any inherent network characteristics, and we are working to verify this conjecture. Also we are interested in studying the network performance available to a non-privileged user process executing on each machine, so we use the standard Internet interface in each experiment. There are special networking facilities that exist between some of the sites in the study (i.e., the vBNS [31]) but processes must have equally special access rights to use them.

All of the data were collected between 6:00 PM on Wednesday, September 18, 1996, and 6:00 PM the following day. The NWS was initiated at the beginning of the experimental period so that the forecasters would have access to no previous information (i.e., all start-up and calibration effects would be visible). Measurements were taken at roughly 30 second intervals, and a latency measurement immediately preceded each throughput measurement. The throughput was measured using a 64 K byte data transfer with 4 K socket buffers at both the sending and receiving ends. We report all throughput measurements in units of megabits per second (mbits/s), and latency in milliseconds (ms).

5.1. PCL throughput

In table 3 we summarize the overall accuracy of each forecasting method when forecasting throughput in the PCL over the 24 hour measurement period. The stochastic gradient (*GRAD*) predictor generates the lowest mean square prediction error, and the trimmed mean (*TRIM_MEAN*) is best in terms of lowest mean percentage prediction error (shown boldface in the table). *MIN_MSE* shows the ability of the NWS to determine the best predictive method (in terms of mean square error) dynamically without knowing *a priori* that *GRAD* would perform best. Similarly, *MIN_MPE* shows the NWS's tracking of mean percentage error. Both of these methods yield error rates that are relatively close to the respective minima, although for this series, all of the forecasting methods except *LAST* and *AR* perform reasonably well.

Table 3
Forecasting method error statsitics for PCL throughput measurements.

Predictor	MSE	MPE
RUN_AVG	0.5274	0.0927
SW_AVG	0.5041	0.0902
LAST	0.7892	0.1066
ADAPT_AVG	0.5214	0.0925
MEDIAN	0.5386	0.0901
ADAPT_MED	0.5337	0.0896
TRIM_MEAN	0.5130	**0.0893**
GRAD	**0.4903**	0.0895
AR	0.7139	0.0992
MIN_MSE	0.5136	0.0901
MIN_MPE	0.5417	0.0906

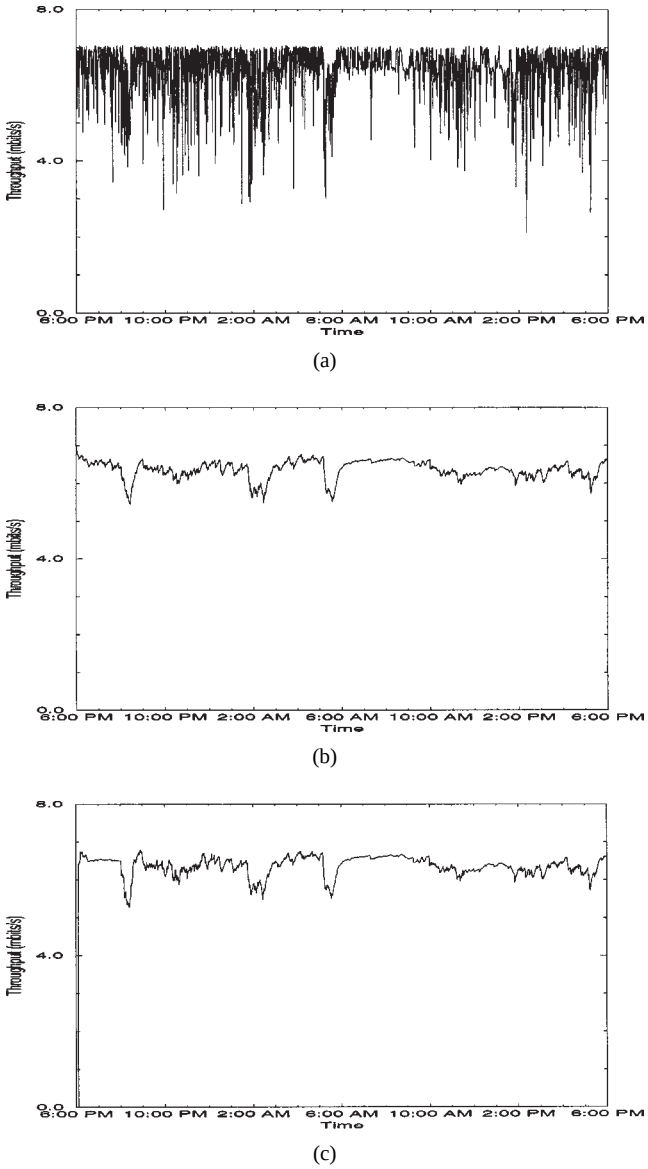


Figure 3. (a) PCL throughput time series, (b) *GRAD* predictions, and (c) *MIN_MSE* predictions.

In figure 3(a) we show the time series of throughput measurements for the intra-PCL connection. The PCL Ethernet segment we monitored is isolated from general Internet traffic by a gateway. Even though it is used only by PCL machines, it still displays considerable performance variation. Predictions made by *GRAD* are shown in figure 3(b) and by *MIN_MSE* in figure 3(c). Except during

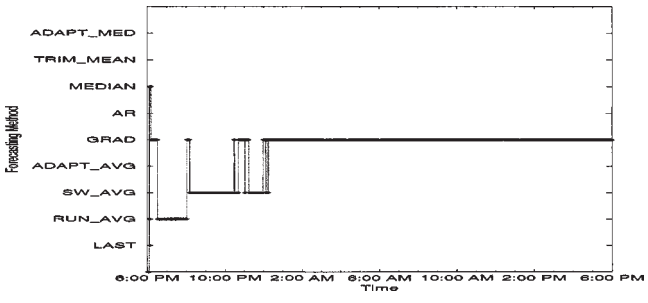


Figure 4. Predictor selection for *MIN_MSE*.

the initial part of the experiment, the prediction curves are identical. That is, even though we did not know ahead of time that *GRAD* would be most accurate, the *MIN_MSE* predictive method automatically identifies it as having the minimum mean square prediction error. The reason that they do not have identical mean square error statistics is that *MIN_MSE* requires some time to recognize *GRAD* as the best predictor.

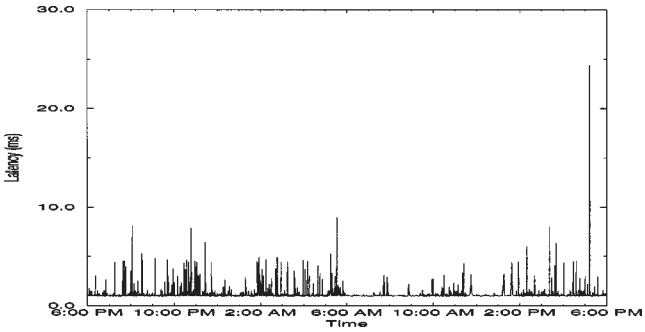
Figure 4 shows predictor selection as a function of time for *MIN_MSE*. The heavy horizontal lines indicate which predictor *MIN_MSE* used at any given point in time. The dotted vertical lines show when the predictor switched from one method to another. Notice that after an initial start-up period (lasting from 6:00 PM until approximately 1:00 AM), *MIN_MSE* uses the values generated by *GRAD* for the remainder of the experiment. Since the NWS is intended to be a continuously available service, such a lengthy calibration or start-up period does not pose a serious problem. The results are similar for *MIN_MPE*; during start-up it switches several times before identifying *TRIM_MEAN* as the most accurate predictor in terms of mean percentage error.

5.2. PCL latency

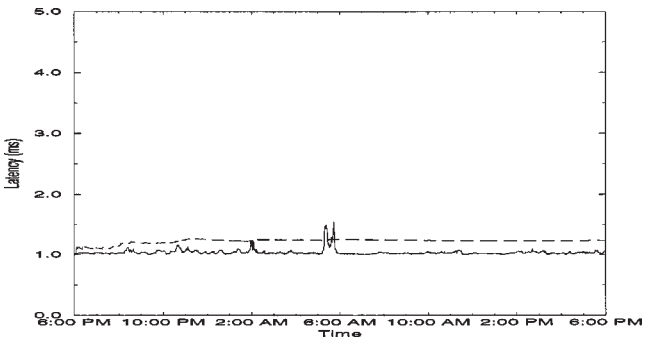
The accuracy of the forecasting methods when predicting latency in the PCL is shown in table 4. For the PCL latency measurements, the median-based forecasters generate predictions having the lowest mean percentage error, and very nearly the lowest mean square error. *RUN_AVG* is slightly better than *MEDIAN* in terms of mean square error, but generates a little over twice the mean percentage

Table 4
Forecasting method error statsitics for PCL latency measurements.

Predictor	MSE	MPE
RUN_AVG	**0.7149**	0.1984
SW_AVG	0.7439	0.2103
LAST	1.427	0.2544
ADAPT_AVG	0.7351	0.1523
MEDIAN	0.7469	**0.0963**
ADAPT_MED	0.7453	0.1012
TRIM_MEAN	0.7317	0.1172
GRAD	0.7261	0.2067
AR	0.9327	0.2991
MIN_MSE	0.7197	0.1961
MIN_MPE	0.7477	0.0968



(a)



(b)

Figure 5. (a) Latency time series for PCL, (b) *RUN_AVG* (dotted) versus *MEDIAN* (solid).

error (19.8% versus 9.6%, respectively). Figure 5(a) shows the time series of latency measurements and in figure 5(b) we compare the predictions generated by *RUN_AVG* (dotted line) to those generated by *MEDIAN* (solid line).

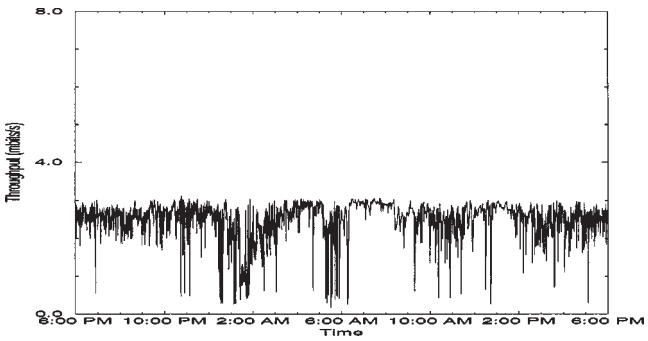
In contrast with the throughput time series (figure 3a), the latency measurements show intermittent outliers departing from an almost uniform value of about 1 ms (figure 5(a)). These outliers do not generally form a trackable trend (their duration is short) and they differ from the stable value by an order of magnitude. As such, a median-based forecasting method will reject them in favor of the uniform tendency. Since the outliers all constitute longer latencies (there are no measurements shorter than 1 ms), a mean-based method will be drawn in the direction of the outliers. Figure 5(b) depicts this relationship. The solid line shows the forecasts generated by *MEDIAN* and the dotted line above it in the figure are the forecasts made by *RUN_AVG*. Note that we have changed the scale of the graph to make the difference easier to discern. Since the *RUN_AVG* forecasts are closer to the outliers when they occur, *RUN_AVG* yields a lower square error measure and, consequently, a lower overall mean square error. However, *RUN_AVG* consistently differs from the uniform 1 ms value, so its cumulative mean percentage error is higher. Since the error terms are not squared when calculating mean percentage error, *MEDIAN* does not accumulate as much error when it encounters an outlier.

Note that *LAST* exhibits poor forecasting performance for this time series because the presence of an outlier does not indicate that another outlier will follow. Note also that

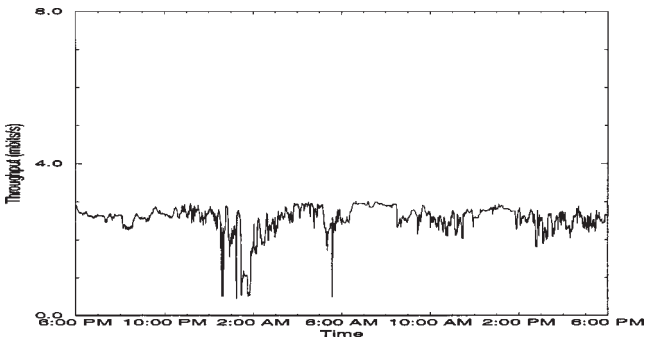
MIN_MSE and *MIN_MPE* correctly identify and track the best predictor according mean square error and mean percentage error, respectively.

5.3. SDSC throughput

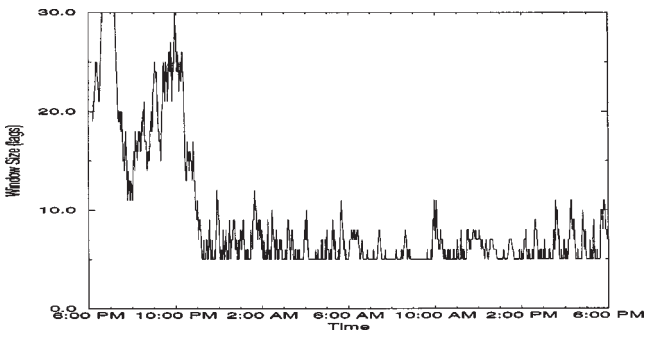
Table 5 shows the accuracy for the throughput forecasts between the UCSD PCL and SDSC. Again, as is true for the PCL throughput measurements, *GRAD* yields the lowest mean square error measure. The best mean percentage error, however, comes from *ADAPT_MED*. We show the throughput time series, the predictions made by *ADAPT_MED*, and a trace of the window size over time in figures 6(a), 6(b) and 6(c), respectively. *ADAPT_MED* starts with an initial window size of 20, but the window eventually settles be between 5 (a preset minimum value) and 10.



(a)



(b)



(c)

Figure 6. (a) PCL-to-SDSC throughput time series, (b) *ADAPT_MED* predictions, and (c) *ADAPT_MED* window size.

Table 5
Forecasting method error statsitics for PCL-to-SDSC throughput measurements.

Predictor	MSE	MPE
RUN_AVG	0.2578	0.2673
SW_AVG	0.1943	0.2162
LAST	0.2872	0.2188
ADAPT_AVG	0.2201	0.2410
MEDIAN	0.2080	0.2221
ADAPT_MED	0.1974	**0.2042**
TRIM_MEAN	0.1968	0.2168
GRAD	**0.1886**	0.2172
AR	0.2269	0.2101
MIN_MSE	0.1919	0.2177
MIN_MPE	0.2017	0.2069

Table 6
Forecasting method error statsitics for PCL-to-SDSC latency measurements.

Predictor	MSE	MPE
RUN_AVG	80.20	0.2215
SW_AVG	82.86	0.2307
LAST	154.72	0.2622
ADAPT_AVG	80.16	0.1539
MEDIAN	79.96	**0.1114**
ADAPT_MED	80.26	0.1125
TRIM_MEAN	**79.28**	0.1246
GRAD	81.05	0.2209
AR	88.67	0.2768
MIN_MSE	79.41	0.1278
MIN_MPE	80.23	0.1121

5.4. SDSC latency

Median-based methods (similar to the results for the PCL) perform best as forecasters of latency between the PCL ad SDSC. Table 6 shows the accuracy of the predictors. The median-based methodologies display lower mean square *and* mean percentage error measures, unlike in the PCL where *RUN_AVG* yields the lowest mean square error. Specifically, *TRIM_MEAN* achieves a slightly smaller mean square error than all of the other methods (except *LAST* and *AR* which are significantly less accurate). In terms of mean percentage error, *MEDIAN* performs, again, only slightly better than *TRIM_MEAN* and *ADAPT_MED*. These three median-based predictors are, in general, twice as accurate the mean-based ones with regard to mean percentage error.

Similar to the PCL latency measurements, the latency time series (figure 7) is characterized by unpredictable outlying deviations from a relatively fixed uniform tendency. The deviations are generally one order of magnitude greater than the typical value of around 3 ms, with a few spikes (those that extend beyond 30 ms) that are as much as two-orders of magnitude greater. Since the deviations are typically short (one sample, in most cases), the median-based methods perform better by rejecting them as unpredictable noise.

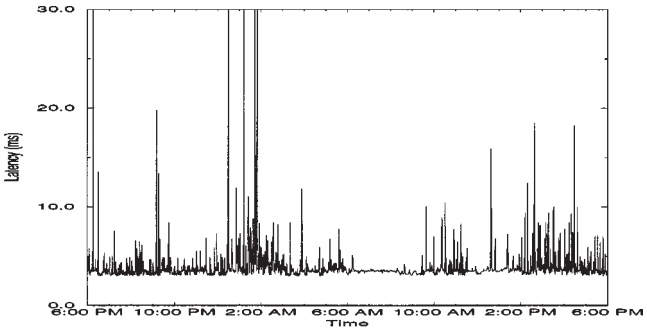


Figure 7. PCL-to-SDSC latency measurements.

5.5. Summary and analysis of network forecasting performance

We show the forecasting accuracy for the PCL-to-Caltech connectivity in tables 7 and 8, for the PCL-to-Oregon connectivity in tables 9 and 10, and for the PCL-to-NCSA in tables 11 and 12 (and we include the measurement data as an appendix).

Again, we show the minimum mean square error and mean percentage error values in each table using boldface type. Tables 13 and 14 summarize the performance of the best predictors for each network setting. Notice that *GRAD* is the overall best predictor of throughput if mean square error is used as an accuracy measure. It fails to yield the lowest error for only the PCL-NCSA connection (table 11), but for that connection it is ranked fifth, (not includ-

Table 7
PCL-to-Caltech forecaster error for throughput.

Predictor	MSE	MPE
RUN_AVG	0.1434	1.506
SW_AVG	0.0815	0.9187
LAST	0.1368	0.9955
ADAPT_AVG	0.0912	1.018
MEDIAN	0.0936	**0.7397**
ADAPT_MED	0.0928	0.7699
TRIM_MEAN	0.0828	0.8107
GRAD	**0.0808**	0.9422
AR	0.1068	1.062
MIN_MSE	0.0812	0.9321
MIN_MPE	0.0927	0.7426

Table 8
PCL-to-Caltech forecaster error for latency.

Predictor	MSE	MPE
RUN_AVG	**7398**	1.841
SW_AVG	7679	1.666
LAST	13880	1.668
ADAPT_AVG	7434	0.6922
MEDIAN	7626	**0.2185**
ADAPT_MED	7637	0.2517
TRIM_MEAN	7592	0.3962
GRAD	7462	1.628
AR	9778	2.275
MIN_MSE	7471	1.537
MIN_MPE	7626	0.2193

Table 9
PCL-to-Oregon forecaster error for throughput.

Predictor	MSE	MPE
RUN_AVG	0.0427	2.409
SW_AVG	**0.0111**	0.4234
LAST	0.0188	0.4868
ADAPT_AVG	0.0132	0.6198
MEDIAN	0.0123	0.3970
ADAPT_MED	0.0127	**0.3955**
TRIM_MEAN	0.0113	0.4065
GRAD	**0.0111**	0.4708
AR	0.0132	0.4529
MIN_MSE	0.0111	0.4480
MIN_MPE	0.0129	0.4015

Table 10
PCL-to-Oregon forecaster error for latency.

Predictor	MSE	MPE
RUN_AVG	181400	0.5380
SW_AVG	187700	0.9994
LAST	362100	1.056
ADAPT_AVG	180700	0.4229
MEDIAN	181400	**0.1099**
ADAPT_MED	181700	0.1304
TRIM_MEAN	**180600**	0.1937
GRAD	183200	0.9817
AR	202900	1.118
MIN_MSE	182200	0.4380
MIN_MPE	181500	0.1114

Table 11
PCL-to-NCSA forecaster error for throughput.

Predictor	MSE	MPE
RUN_AVG	0.0104	2.461
SW_AVG	0.0022	0.4956
LAST	**0.0015**	**0.3287**
ADAPT_AVG	0.0041	0.9790
MEDIAN	0.0027	0.4908
ADAPT_MED	0.0016	0.3808
TRIM_MEAN	0.0023	0.4920
GRAD	0.0025	0.6557
AR	0.0016	0.3604
MIN_MSE	0.0015	0.3429
MIN_MPE	0.0015	0.3295

Table 12
PCL-to-NCSA forecaster error for latency.

Predictor	MSE	MPE
RUN_AVG	60420	0.7080
SW_AVG	59450	0.6764
LAST	115000	0.7181
ADAPT_AVG	60260	0.3830
MEDIAN	61560	**0.0701**
ADAPT_MED	61580	0.0727
TRIM_MEAN	59730	0.2048
GRAD	**58890**	0.6719
AR	72570	0.9422
MIN_MSE	60190	0.6937
MIN_MPE	62370	0.0827

Table 13
Summary of best forecasters for throughput.

Network connection	Minimum MSE	Minimum MPE
PCL	*GRAD*	*TRIM_MEAN*
PCL-SDSC	*GRAD*	*ADAPT_MED*
PCL-Caltech	*GRAD*	*MEDIAN*
PCL-Oregon	*GRAD*	*ADAPT_MED*
PCL-NCSA	*LAST*	*LAST*

Table 14
Summary of best forecasters for latency.

Network connection	Minimum MSE	Minimum MPE
PCL	*GRAD*	*MEDIAN*
PCL-SDSC	*TRIM_MEAN*	*MEDIAN*
PCL-Caltech	*RUN_AVG*	*MEDIAN*
PCL-Oregon	*TRIM_MEAN*	*MEDIAN*
PCL-NCSA	*GRAD*	*MEDIAN*

ing *MIN_MSE* and *MIN_MPE* in the ranking). In general, though, the mean-based predictors tend to outperform the median-based ones, for throughput time series in this study, if mean square error is used to measure prediction accuracy. Analogously, *MEDIAN* is the most accurate predictor of latency, in terms of mean percentage error, for each set of latency measurements. *MIN_MSE* and *MIN_MPE* correctly track the leading predictor in each case without knowing ahead of time which will be most successful.

Notice also that *LAST* is not a good predictor of network performance (particularly of latency) *except* for the cross-country Internet throughput measurements. In that experiment, however, it performs best. We believe that this result supports those reported in [2] which demonstrate the ability of autoregressive models to correctly reflect aggregate traffic patterns in certain wide-area network environments. In particular, the authors analyze packet data taken from the gateway between SDSC and the NSFNET backbone. The PCL-to-NCSA TCP connection we monitored traverses this gateway. Since we are also measuring the effects of protocol and buffer processing on each end of a connection, we expected aggregate packet behavior to dominate in those settings where network paths include many heavily congested gateways. For the PCL-to-NCSA throughput measurements, indeed, *AR* performs only slightly worse than *LAST* as a predictor. The performance of *AR* is competitive with the other forecasters, in terms of mean square error, for the PCL-to-Oregon throughput series as well.

We summarize these results by noting that:

- if mean square error is the accuracy measure used to judge the fitness of a forecasting method, a *stochastic gradient* predictor is a good choice for forecasting throughput over most Internet connections;
- if mean percentage error is used as an accuracy measure, a *sliding-window median* is a good choice of forecasting method for latency, given current Internet technology;
- the best predictor of each performance characteristic (latency and throughput in this study) is, in general, not obvious and varies from resource to resource;
- the dynamically-selecting predictive methods successfully track the best predictor in each case yielding forecast error rates close to the minimum.

6. Conclusions and future work

To predict the performance of resources in a metacomputing environment, we have developed the Network Weather Service. It operates an arbitrary set of performance sensors, and dynamically generates forecasts from the periodic readings it takes. Determining the most appropriate forecasting method for each resource *a priori* is difficult. Indeed, in the absence of a perfect generating model, the best forecasting method for any particular resource may change over time.

In this work, we illustrate the end-to-end TCP/IP throughput and latency performance an application can obtain between the UCSD Parallel Computation Lab, and a variety of geographically dispersed computing sites. The NWS is able to make dynamic short-term forecasts for both of these communication characteristics, although the accuracy of the forecasts varies from site to site. More importantly, the system can correctly identify the best method "on the fly" based on a running tabulation of prediction error. Since we have designed the system to be extensible, we can incorporate a multitude of techniques from which it can choose the best for any given resource and any given time.

Our work with the NWS is very much in its formative stages. We plan to investigate how the system can incorporate modeling techniques which require a computationally-intensive "fitting" phase. The ARIMA models described in [2], the self-similarity analysis outlined in [26], and the semi-nonparametric techniques discussed in [13,14], all provide immediately promising avenues of investigation. We would like to discern the relationship between the computational complexity devoted to making a forecast its accuracy. We also plan to integrate other sensory mechanisms such as those described in [6], and to investigate how groups of forecasts may be composed to yield higher-level performance characteristics.

As of this writing, second generation implementations of the NWS are underway for the Globus/Nexus[9,15] and Legion [17,25] metacomputing systems. These versions will be initially deployed as part of the GUSTO (Globus Ubiquitous Testbed) [9] and DOCT (Distributed Object Computational Testbed) [11] metacomputing testbeds. We plan to use these implementations both to investigate metacomputing scheduling via AppLeS [1,3] and the development of general quality-of-service mechanisms.

Acknowledgements

The NWS is part of the AppLeS project, and as such, owes much of its existence to Francine Berman and the members of the AppLeS Corps at UCSD whom we thank for pearls of wisdom too plentiful to enumerate. We would also like to thank Chandra Krintz and Jenny Schopf at UCSD and Allen Downey at UCB and SDSC for their comments, suggestions, and criticisms, all of which were invaluable. Lastly, we thank Reagan Moore at SDSC, Carl Kesselman at Caltech, Allen Malony and Sameer Suresh Shende at the University of Oregon, and Jeff Terstriep and Randy Sharpe at NCSA for their indispensable insights, use of their facilities, and interminable debugging help.

Appendix

In this appendix, we include the throughput and latency measurement data for TCP/IP communication streams between a Sun Sparc 10 in the UCSD PCL and Sun Sparc 5 in the PCL (figures 8 and 9), a DEC Alpha at the San Diego Supercomputer Center (figures 10 and 11), a Sun Hypersparc located at Caltech (figures 12 and 13), an SGI PowerChallenge located at the University of Oregon (figures 14 and 15), and an SGI PowerChallenge located at

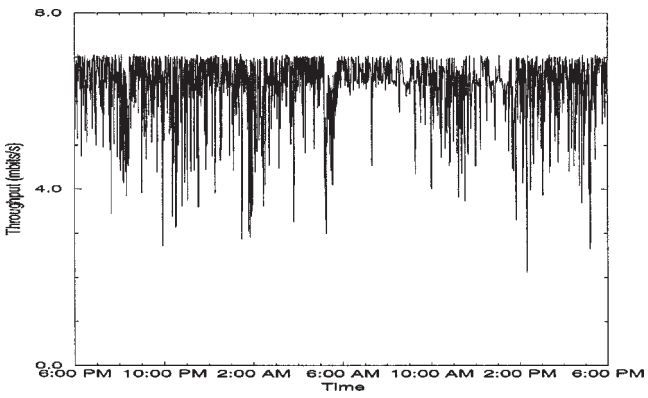


Figure 8. Intra-PLC throughput measurements.

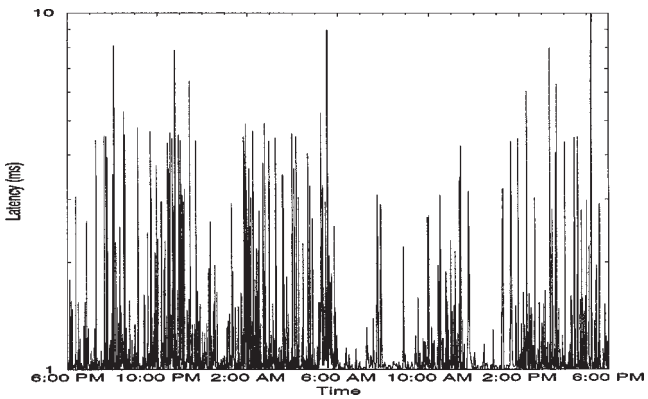


Figure 9. Intra-PLC latency measurements.

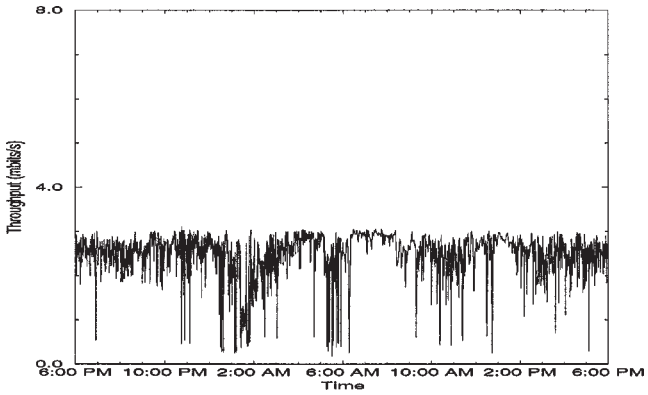


Figure 10. PCL-to-SDCS throughput measurements.

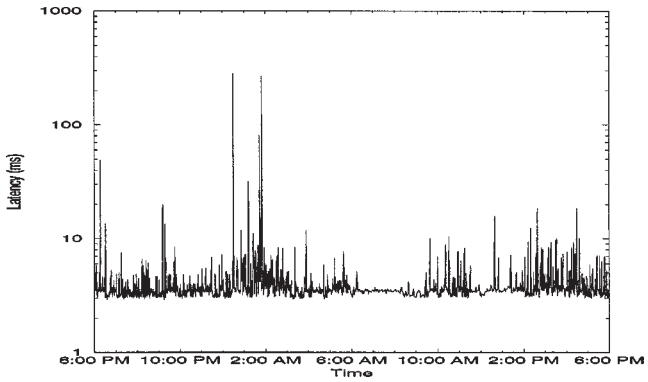


Figure 11. PCL-to-SDCS latency measurements.

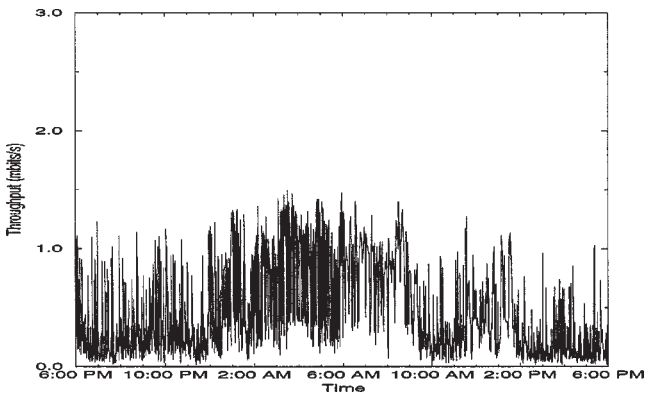


Figure 12. PCL-to-Caltech throughput measurements.

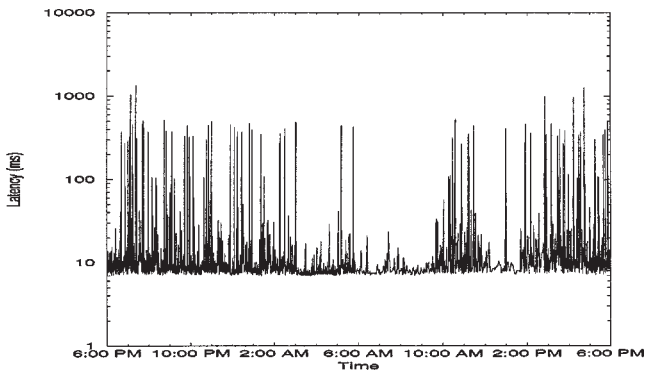


Figure 13. PCL-to-Caltech latency measurements.

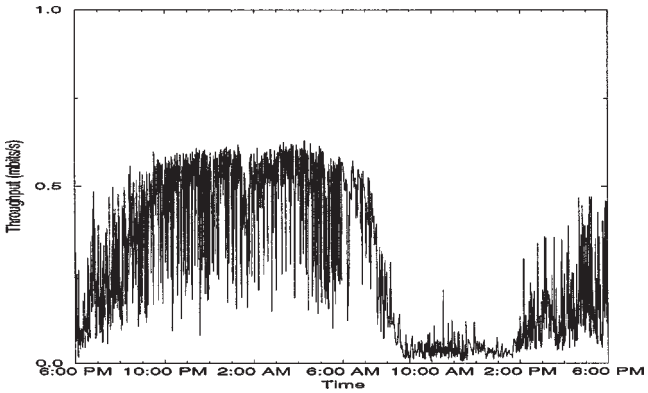


Figure 14. PCL-to-Oregon throughput measurements.

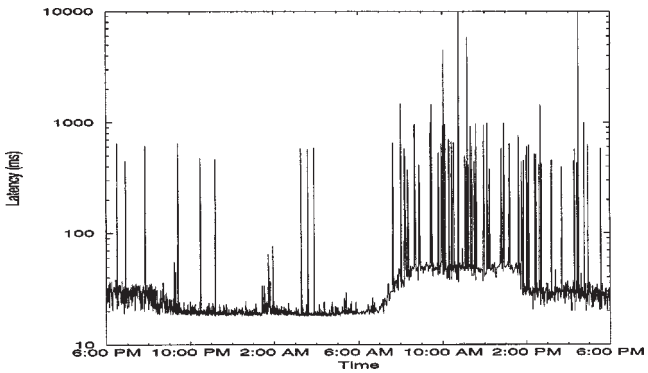


Figure 15. PCL-to-Oregon latency measurements.

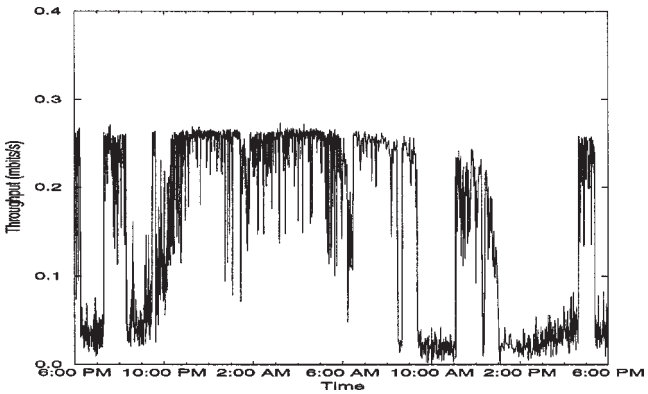


Figure 16. PCL-to-NCSA throughput measurements.

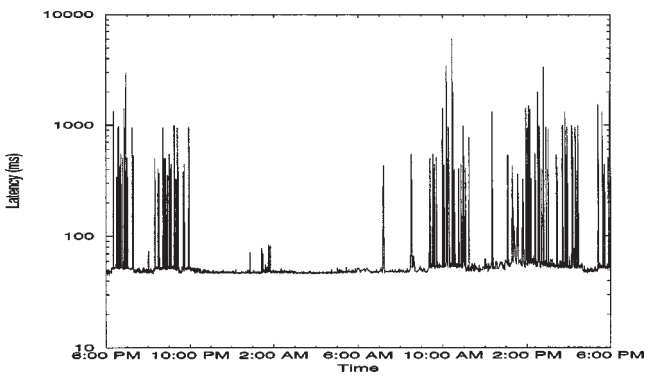


Figure 17. PCL-to-NCSA latency measurements.

the National Center for Supercomputing Applications (figures 16 and 17). Note that scale of the magnitude differs in each graph, and that the latency magnitudes shown in these figures are plotted on a log scale.

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