92¢ /MFlops/s, Ultra-Large-Scale Neural-Network Training on a PIII Cluster

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Abstract

Artificial neural networks with millions of adjustable parameters and a similar number of training examples are a potential solution for difficult, large-scale pattern recognition problems in areas such as speech and face recognition, classification of large volumes of web data, and finance. The bottleneck is that neural network training involves iterative gradient descent and is extremely computationally intensive. In this paper we present a technique for distributed training of Ultra Large Scale Neural Networks (ULSNN) on Bunyip, a Linux-based cluster of 196 Pentium III processors. To illustrate ULSNN training we describe an experiment in which a neural network with 1.73 million adjustable parameters was trained to recognize machine-printed Japanese characters from a database containing 9 million training patterns. The training runs with an average performance of 163.3 GFlops/s (single precision). With a machine cost of $150,913, this yields a price/performance ratio of 92.4¢ /MFlops/s (single precision). For comparison purposes, training using double precision and the ATLAS DGEMM produces a sustained performance of 70 MFlops/s or $2.16 / MFlop/s (double precision).

1. Introduction

Artificial neural networks are a class of parametric, nonlinear statistical models that have found wide-spread use in many pattern recognition domains, including speech recognition, character recognition, signal processing, medical diagnosis and finance. The typical network in such an application has 100–100,000 adjustable parameters and requires a similar number of training patterns in order to generalize well to unseen test data. Provided sufficient training data is available, the accuracy of the network is limited only by its representational power, which in turn is essentially proportional to the number of adjustable parameters. Thus, in domains where large volumes of data can be collected — such as speech, face and character recognition, and web page classification — improved accuracy can often be obtained by training a much larger network.

In this paper we describe a method for distributed training of Ultra Large Scale Neural Networks (ULSNN), or networks with more than one million adjustable parameters and a similar number of training examples. At its core, the algorithm uses Emmerald, a single-precision (32 bit) general matrix-matrix multiply (SGEMM) based on the Pentium III SIMD Streaming Extensions (SSE), with a peak performance in excess of 1090 MFlops/s on a single 550 MHz Pentium III. The use of single-precision floating point operations is justified by the fact that we have found it sufficient for gradient-based training of ULSNN’s. For medium–large scale neural networks as few as 16 bits precision is sufficient [2].

To illustrate the use of our ULSNN training code, we describe an experiment in which a neural network with 1.73 million adjustable parameters is being trained to recognize machine-printed Japanese characters from a database containing 9 million training patterns. The training is running on Bunyip, a 196 processor, Linux-based Intel Pentium III cluster consisting of 98 dual 550 MHz processor PC’s, each containing 384 MBytes of RAM, 13 GBytes of hard disk and 3x100 Mb/s fast Ethernet cards. All components in Bunyip are “COTS” (Commodity-Off-The-Shelf), and were sourced from a local PC manufacturer (see http://tux.anu.edu.au/Projects/Beowulf/).

Our longest experiment took 56 hours and 52 minutes, requiring a total of 31.2 Peta Flops (1015 single-precision
floating-point operations), with an average performance of 152 GFlops/s (single precision) while under load. With no other user processes running the performance increases to 163.3 GFlops/s which was sustained for a four hour test before returning access to other users. Total memory usage during training was 32.37 GBytes. The total machine cost, including the labor cost in construction, was AUD$253,000, or USD$150,913 at the exchange rate of AUD$1 = .5965¢ USD on the day of the final and largest payment. This gives a final price/performance ratio of USD 92.4¢ /MFlops/s (single precision). For comparison purposes, training using double precision and the ATLAS DGEMM [11] produced a sustained performance of 70 MFlops/s or $2.16 /MFlops/s (double precision).

2. “Bunyip” Hardware Details

The machine used for the experiments in this paper is “Bunyip”, a 98-node, dual Pentium III Beowulf-class system running Linux kernel 2.2.14. Our main design goals for this machine were to maximise CPU and network performance for the given budget of AUD $250,000 (about USD $149,125). Secondary factors to be balanced into the equation were: amount of memory and disk; reliability; and the overall size of the machine. All dollar figures quoted in the remainder of this paper are US dollars.

The Intel Pentium III processors were chosen over Alpha or SPARC processors for price/performance reasons. Dual-CPU systems were preferable as overall cost and size per CPU is lower than single-CPU or quad-CPU systems. Unfortunately, at the time of designing this machine AMD Athlon and Motorola/IBM G4 systems were not available in dual-CPU configurations. We were also keen to use the SSE floating point instructions of the Pentium III range. 550 MHz CPUs were eventually selected as having the best price/performance available in the Pentium III range at that time.

For the networking requirements, we decided to go with a commodity solution rather than a proprietary solution. Gigabit ethernet was considered, but deemed too expensive at around $300 per node for the NIC and around $1800 per node for the switch. Instead, a novel arrangement of multiple 100 Mb/s NICs was selected with each node having three NICs which contributed some $65 per node (plus switch costs – see below).

The configuration for each node is dual Intel Pentium III 550 CPUs on an EPoX KP6-BS motherboard with 384 MBytes RAM, 13 GByte UDMA66 (IDE) hard disk and three DEC Tulip compatible 100 Mb/s network interfaces, one of which has Wake-On-LAN capability and provision for a Boot ROM. The nodes have no removable media, no video capability and no keyboards. Each node cost $1282.

With reference to figure logically the 96 nodes are connected in four groups of 24 nodes arranged as a tetrahedron with a group of nodes at each vertex. Each node in a vertex has its three NICs assigned to one of the three edges emanating from the vertex. Each pair of vertices is connected by a 48-port Hewlett-Packard Procurve 4000 switch (24 ports connecting each way). The switching capacity of the Procurve switches is 3.8 Gb/s. The bi-sectional bandwidth of this arrangement can be determined by looking at the bandwidth between two groups of nodes and the other two groups through 4 switches, giving a total of 15.2 Gb/s. The 48-port switches cost $2386 each.

Two server machines, more or less identical to the nodes, with the addition of CD-ROM drives, video cards and keyboards, are each connected to a Netgear 4-port Gigabit switch which is in turn connected to two of the HP Procurve switches via gigabit links. The two server machines also act as connections to the external network. Two hot-spare nodes were also purchased and are used for development and diagnostic work when not required as replacements for broken nodes.

2.1 Total Cost

All up we spent 98 x $1282 ($125,636) on the computational nodes (including the two hot-sares), $17,594 on the six 48-port and the 4-port gigabit switches (6 x $2386, 2 x $894 (gigabit interfaces) and $1490 for the gigabit switch), $3870 on servers (including gigabit NICs, monitors etc.), $944 for network cables, $179 on electrical work, $238 on power cables and power boards, and $298 on boot EPROMs. The ex-library shelving was loaned to us, but would have cost $354 from a local second-hand furniture...
shop. Although no component was explicitly budgeted for staff time, this amounted to about 3 weeks to assemble and configure the machine which adds approximately $1800 to the overall cost of the machine. All up, the total cost was USD $150,913.

3. Emmerald: A SIMD SGEMM for Intel Pentium III Processors

This section introduces Emmerald, the high performance software kernel of our ULSNN training system. It provides a single-precision, dense, matrix-matrix multiplication routine that uses the single instruction, multiple data (SIMD) features of Intel PIII chips (SIMD Streaming Extensions, or SSE). The SSE provide a set of new floating-point assembler instructions that allow simultaneous operation on four single-precision floating-point numbers. Emmerald outperforms a naive (3-loop) matrix-matrix multiply by 8 times for square matrices of size $64 \times 64$, and a peak of 29 times for matrices of size $672 \times 672$. Emmerald can be downloaded from http://beaker.anu.edu.au/~daa/research.html.

3.1 Single precision general matrix-matrix multiply (SGEMM)

Without resorting to the complexities associated with implementing Strassen’s algorithm on deep-memory hierarchy machines [9, 10], dense matrix-matrix multiplication requires $2MNK$ floating point operations where $A : M \times K$ and $B : K \times N$ define the dimensions of the two matrices. Although this complexity is fixed, skillful use of the memory hierarchy can dramatically reduce overheads not directly associated with floating point operations. Memory hierarchy optimization combined with the use of SSE gives Emmerald its performance advantage.

Emmerald implements the SGEMM interface of Level-3 BLAS, and so may be used to improve the performance of single-precision libraries based on BLAS (such as LAPACK [8]). There have been several recent attempts at automatic optimization of GEMM for deep-memory hierarchy machines, most notable are PHiPAC [6] and the more recent ATLAS [11]. ATLAS in particular achieves performance close to vendor optimized commercial GEMMs. Neither ATLAS nor PHiPAC make use if the SSE instructions on the PIII for their implementation of SGEMM.

3.2 SIMD Parallelisation

A SIMD GEMM must aim to minimize the ratio of memory accesses to floating point operations. We employed two core strategies to achieve this:

- accumulate results in registers for as long as possible to reduce write backs;
- re-use values in registers as much as possible.

In [5] several dot-products were performed in parallel inside the innermost loop of the GEMM. Taking the same approach we found experimentally that 5 dot-products in the inner loop gave the best performance. Figure 2 shows how these 5 dot products utilise SIMD parallelism.

3.3 Optimizations

A number of techniques are used in Emmerald to improve performance. Briefly, they include:

- L1 blocking: Emmerald uses matrix blocking [4, 6, 11] to ensure the inner loop is operating on data in L1 cache. Figure 3 shows the L1 blocking scheme. The block dimensions $m$ and $n$ are determined by the configuration of dot-products in the inner loop (Section 3.2) and $k$ was determined experimentally.
• **Unrolling:** The innermost loop is completely unrolled for all possible lengths of \( k \) in L1 cache blocks, taking care to avoid overflowing the instruction cache.

• **Re-buffering:** Since \( B' \) (Figure 3) is large \((336 \times 5)\) compared to \( A' \) \((1 \times 336)\), we deliberately buffer \( B' \) into L1 cache. While buffering \( B' \) we re-order its elements to enforce optimal memory access patterns. This has the additional benefit of minimising translation look-aside buffer misses [12].

• **Pre-fetching:** Values from \( A' \) are not buffered into L1 cache. We make use of SSE pre-fetch assembler instructions to ensure \( A' \) values will be in L1 cache when needed.

• **L2 Blocking:** Efficient L2 cache blocking ensures that peak rates can be maintained as long as \( A, B \) and \( C \) fit into main memory.

### 3.4 Emmerald Results

The performance of Emmerald was measured by timing matrix multiply calls with size \( M = N = K = 16 \) up to 700. The following steps were taken to ensure a conservative performance estimate:

- wall clock time on an unloaded machine is used rather than CPU time;
- the stride of the matrices, which determines the separation in memory between each row of matrix data, is fixed to 700 rather than the optimal value (the length of the row);
- caches are flushed between calls to \text{sgemm}() .

Timings were performed on a PIII 450MHz running Linux (kernel 2.2.14).

Figure 4 shows Emmerald’s performance compared to ATLAS and a naïve three-loop matrix multiply. The average MFlops/s rate of Emmerald after size 100 is 1.69 times the clock rate of the processor and 2.09 times faster than ATLAS. A peak rate of 890 MFlops/s is achieved when \( m = n = k = \text{stride} = 320 \). This represents 1.98 times the clock rate. On a PIII 550 MHz (the processors in Bunyip) we achieve a peak of 1090 MFlops/s. The largest tested size was \( m = n = k = \text{stride} = 3696 \) which ran at 940 MFlops/s at 550 MHz. For more detail see [1].

### 4. Artificial Neural Networks

A one-hidden-layer artificial neural network maps input vectors \( x = (x_1, \ldots, x_n) \in \mathbb{R}^n \) to output vectors \( y = (y_1, \ldots, y_m) \in \mathbb{R}^m \) according to the formula:

\[
y_i(x) = \sigma \left( \sum_{j=1}^{n_k} w_{ij} h_j(x) \right),
\]

where \( \sigma : \mathbb{R} \to \mathbb{R} \) is some squashing function (we use \( \sigma = \tanh \)). \( w_{ij}^{ho} \) are the adjustable parameters connecting the hidden nodes to the output nodes, and \( h_j(x) \) is the activation of the \( j \)-th hidden node:

\[
h_j(x) = \sigma \left( \sum_{k=1}^{n_t} w_{jk}^{ih} x_k \right).
\]

In the last expression, \( w_{jk}^{ih} \) are the adjustable parameters connecting the input nodes to the nodes in the hidden layer. Given a matrix \( X \) of \( n_p \) training patterns and a matrix \( T \) of desired outputs for the patterns in \( X \),

\[
X = \begin{bmatrix} x_{11} & \cdots & x_{1n_t} \\ \vdots & \ddots & \vdots \\ x_{n_p 1} & \cdots & x_{n_p n_t} \end{bmatrix},
T = \begin{bmatrix} t_{11} & \cdots & t_{1n_o} \\ \vdots & \ddots & \vdots \\ t_{n_p n_o} & \cdots & t_{n_p n_o} \end{bmatrix},
\]

the goal is to find sets of parameters \( w_{ij}^{ho} \) and \( w_{jk}^{ih} \) minimizing the mean-squared error:

\[
E = \sum_{i=1}^{n_p} \sum_{j=1}^{n_o} [y_j(x_i) - t_{ij}]^2,
\]

where \( x_i \) is the \( i \)-th row of the data matrix \( X \). Usually [3] is minimized by some form of gradient descent.
4.2 Computing The Gradient Using Matrix-Matrix Multiply

If we write $Y$ for the matrix of outputs $y_j(x_i)$, $H$ for the matrix of hidden activations $h_j(x_i)$, and $W^{ih}$ and $W^{ho}$ for the parameter matrices $w^{ih}_{kl}$ and $w^{ho}_{ij}$ respectively, then

$$H = \sigma (X \ast W^{T}_{ih})$$
$$Y = \sigma (H \ast W^{T}_{ho})$$

where “$\ast$” denotes ordinary matrix multiplication and $\sigma (A)$ means apply $\sigma$ elementwise to the components of $A$. Defining

$$Y_\Delta = (I - Y \ast\ast Y) \ast\ast (T - Y) ,$$
$$H_\Delta = (I - H \ast\ast H) \ast\ast (Y_\Delta \ast W_{ho}) ,$$

where “$\ast\ast$” denotes elementwise matrix multiplication, we have

$$\nabla_{ih}E = H^T_\Delta \ast X ,$$
$$\nabla_{oh}E = Y^T_\Delta \ast H ,$$

where $\nabla_{ih}E$ is the gradient of $E$ with respect to the parameters $W^{ih}$ and $\nabla_{oh}E$ is the gradient of $E$ with respect to $W^{ho}$ [3].

Thus, computing the gradient of the error for an artificial neural network can be reduced to a series of ordinary matrix multiplications and elementwise matrix multiplications. For large networks and large numbers of training patterns, the bottleneck is the ordinary matrix multiplications, which we implement using Emmerald’s SGEMM routine. In all our experiments we found 32 bits of floating-point precision were enough for training. For neural networks with $\approx 10,000$ parameters, as few as 16 bits are sufficient [2].

Armed with the gradient $\nabla E$, we can adjust the parameters $W$ by a small amount in the negative gradient direction $W := W - \alpha \nabla E$ and hence reduce the error. However, because the gradient computation can be very time-consuming (a total of 52.2 Tera-floating point operations in our largest experiment), it is more efficient to employ some form of line search to locate a local maximum in the direction $\nabla E$. For the experiments reported in the next section we used the Polak-Ribière conjugate-gradient descent method [4, §5.5.2] to choose the search direction, combined with an exponential step-size scheme and quadratic interpolation in order to locate a maximum in the search direction. We were also able to speed the search for a maximum by using gradient information to bracket the maximum, since only the sign of the inner product of the gradient with the search direction is required to locate the maximum in that direction, and the sign can be reliably estimated with far fewer training patterns than is required to estimate the error.

4.3 Training Set Parallelism

Since the error $E$ and gradient $\nabla E$ are additive over the training examples, the simplest way to parallelize the training of a neural network is to partition the training data into disjoint subsets and have each processor compute the error and gradient for its subset. This works particularly well if there are a large number of training patterns so that each processor can work with near-optimal matrix sizes. The communication required is the transmission of the neural network parameters to each slave processor, and the transmission of the error and gradient information back from each slave to a master node which reduces them to a single error or gradient vector.

5. Communication

This section discusses the communication costs associated with distributed NN training, arguing that these costs are non-trivial for ULSNNs. A reduce algorithm optimised for Bunyip’s topology is also discussed.

5.1 Communication Costs

The inter–process communication costs during network training arise from broadcasting the network parameters to all processes and reducing the network error and gradients from each process to the master process. The parameter broadcasting is cheap, since many copies of the same data is sent to all processes. Broadcasts can take advantage of features such as TCP/IP broadcasting. The reduce process is more difficult with each process generating unique vectors which must be collected and summed by the master process. The time taken to reduce data grows with both the number of parameters and the number of processes. The remaining communication consists of start and stop messages which are insignificant compared to the aforementioned costs.

A typical neural network with 100 inputs, 50 hidden layer neurons, and 50 output neurons, requires 7500 parameters, or 30 KBytes of data (single precision), to be sent from every node to the master node. A naive reduction over 194 processes using a 1Gb/s link, such as used in Bunyip, would take 0.05 seconds assuming 100% network utilisation. Our ULSNN with 400 inputs, 480 hidden layer neurons and 3203 output neurons requires 1,729,440 parameters or 6.6 MBytes of data per process which would require 10.1 seconds. There is sufficient memory on each node to occupy both processors for 446 seconds calculating gradients before a reduce operation is required. Consequently the reduce operation would cost at least 2.3% of the avail-
able processing time, more if not enough training data is available or the network size is increased.

This demonstrates that although communication costs for distributed NN training are minimal for commonly implemented network sizes, ULSNN training must optimise inter-process communication to achieve the best performance.

We reduced communication as much as possible by only distributing the neural-network parameters to all the slaves at the very start of training (rather than at each step), and thereafter communicating only the search direction and the amount to step in that direction. One significant reduction operation is required per epoch to send the error gradient vector from each process to the master which then co-ordinates the step size search with the slaves.

All communication was done using the LAM implementation of MPI (http://www.mpi.nd.edu/lam). Communicating parameters or directions to all processors required a 6.6 MBytes broadcast operation from the server to each of the 194 processors in the cluster, while reducing the gradient back to the master required 6.6 MBytes of data to be communicated from each processor back to the server. LAM/MPI contains a library reduce operation which uses a simple $O(\log n)$ algorithm that distributes the load of the reduce over many processes instead of naïvely sending 194 gradient vectors to one node [7]. This results in a reduce operation on Bunyip which takes 8.5 seconds over 8 stages.

### 5.2 Optimising Reductions

There are two problems with existing free implementations of MPI reduce operations. The first is the lack of shared memory protocols on clusters with multi-processor nodes, instead using slow TCP/IP communications between processors on the same motherboard. Secondly, the reduce operation does not take advantage of the topology of the cluster. For example, the best reduce algorithm to use on a ring network might be to send a single vector to each node on the ring in turn, which adds its contribution before passing the vector to the next node. On a star network the best algorithm might be to send each contribution to the central server and sum as they arrive.

To decrease the time taken per reduce, we wrote a customised routine utilising shared memory for intra-node communication and MPI non-blocking calls for inter-node communication. This routine is summarised by Figure 5. It is split into 4 stages, each of which takes advantage of an aspect of Bunyip’s topology shown in Figure 1.

1. Each node contains two processors, both running an instance of the training process. All 97 nodes (including the server), reduce 6.6 MBytes of data though shared memory between processes, taking 0.18 seconds. The time taken to add the two sets of data together is approximately 0.005 seconds.
2. Each node in group A can open a 100 Mb/s connection to any node in group B via switch 0. Thus all 24 nodes in A can reduce to their B counterparts in parallel. This requires 0.66 seconds. The same trick is used for reducing from group C to D. The reduced data now resides only on the B and D nodes. The total bandwidth for all 96 nodes in this stage is 4.03 Gb/s.
3. Each node contains 3x100 Mb/s NICs. This allows a node to receive data from three other nodes simultaneously provided the TCP/IP routing tables are correctly configured. We split the 24 nodes in each group into 6 sets of 4 nodes. The first of each set (see node BA in Figure 5) is designated as the root and the other three nodes send to it via different NICs. This takes 0.9 seconds achieving a bandwidth of 185 Mb/s into each root node, or 2.22 Gb/s across all 12 root nodes.
4. The final step is a standard MPI library reduce from 6 B nodes and 6 D nodes to the master process. This is the slowest step in the process taking 3.16 seconds, including the time spent waiting for the the nodes to synchronize since they do not start reducing simultaneously.

The overall time taken for the optimised reduce to complete is 4.9 seconds. The actual time saved per reduction is 3.6 seconds. The training performance speedup from this saving varies with the duration of the gradient calculation which depends linearly on the number of training patterns. Figure 6 illustrates the expected speedup achieved
by using the optimised reduce instead of the MPI library reduce, against the total number of training patterns used. In practice our peak performance of 163.3 GFlops/s benefits by roughly 1% from the optimised reduce, however the speedups are much more marked for smaller (and more frequently encountered) data sets.

6. Japanese Optical Character Recognition

In this section we describe our distributed application of the matrix-matrix multiply technique of Section 3 used to train an artificial neural network as a classifier for machine-printed Japanese characters.

6.1 The Problem, Data and Network Architecture

Japanese optical character recognition (Japanese OCR) is the process of automatically recognizing machine-printed Japanese documents and converting them to an electronic form. The most difficult aspect of Japanese OCR is correctly classifying individual characters, since there are approximately 4000 characters in common usage.

The training data for our neural network consisted of 168,000 scanned, segmented, hand-truthed images of Japanese characters purchased from the CEDAR group at the University of Buffalo. The characters were scanned from a variety of sources, including books, faxes, newspapers and magazines. Figure 7 gives an idea of the varying quality of the character images.

Each character in the CEDAR database is represented as a binary image of varying resolution. We down-sampled all the images to a $20 \times 20$ grey-scale format. The neural network had 400 input nodes, one for each pixel. The database contained examples of 3203 distinct characters, hence the neural-network had 3203 output nodes. The hidden layer was chosen to have 480 nodes. In total, the network had 1.73 million adjustable parameters.

168,000 training examples are not sufficient to avoid over-fitting in a network containing 1.73 million adjustable parameters, so we generated synthetic data from the original characters by applying random transformations including line thickening and thinning, shifting, blurring and noise addition. The total number of training examples including the artificial ones was 9,264,000 approximately 5.4 per adjustable network parameter. These were distributed uniformly to 193 of the processors in Bunyip. A further 6320 examples of the CEDAR data set were used for testing purposes.

6.2 Training

With reference to equations (1), (2), and (3), the total number of floating point operations required to compute the error $E$ in a neural network is $2 \times n_p \times (n_i + n_o) \times n_h$, which equals 32 Tera floating-point operations for the Japanese OCR experiment. A gradient calculation uses $n_p \times (4 \times n_i \times n_h + 6 \times n_h \times n_o)$, or 92 Tera floating-point operations.

To assist with load balancing, each slave processor stepped through its training patterns 320 at a time. Between each step the master node was polled to determine whether more steps were required. Once 80% of the total training data had been consumed, the master instructed all slaves to halt computation and return their results (either the error or the gradient). In this way the idle time spent waiting for other slaves to finish was reduced to at most the length of time needed by a single processor to process 320 patterns. With 80% of the data, an error calculation required 26 TFlops and a gradient calculation requires 74 TFlops, or 135 GFlops and 383 GFlops per processor respectively.


| Patterns | % Error |
|----------|---------|
| 343800   | 51      |
| 611200   | 46      |
| 1833600  | 33      |

*Table 1.* Generalisation error decreases as the total number of patterns increases.

### 7. Results

This section describes the classification accuracy achieved; then concentrates on the performance scalability over processors before finishing with peak performance results which result in our claim of a price/performance ratio of \(92.4\text{¢} /\text{MFlop/s}\).

#### 7.1 Classification Accuracy

The network’s best classification error on the held-out 6,320 examples is 33%, indicating substantial progress on a difficult problem (an untrained classifier has an error of \(1 - \frac{1}{3200} = 99.97\%). We observed an error rate of 5\% on the 40\% of the data which contained the most examples of individual characters. Continued training after the 33\% error rate was achieved improved the performance on the common characters at the cost of greatly decreased performance on the rare ones. This leads to the conclusion that overfitting is occurring on characters with only one or two examples from the original data set, despite the number of transformations being generated. A more uniform accuracy could be achieved by generating more transforms of rare characters, or preferably, using a greater number of original examples.

A very large amount of data is required for two reasons. The first is to avoid overfitting. Table 1 compares the generalisation accuracy with the total number of training examples used (including transformations of the original 168,000 patterns). Each data point in this graph represents approximately 48 hours training time. Training was halted after 10 epochs result in no classification improvement on the test set.

#### 7.2 Communication Performance

Recalling from Section 5.1 that communication overhead increases with decreasing patterns then the second motivation for large training sets is to reduce such overhead. Figure 8 demonstrates how the performance scales with the number of processors used (including transformations of the original 168,000 patterns). Each data point in this graph represents approximately 48 hours training time. Training was halted after 10 epochs result in no classification improvement on the test set.

For a small number of processors, both networks exhibit linear performance scale up, but we observe that for many processors the larger problem scales better despite the increased number of network parameters. This is due to the communication overhead in the small network increasing dramatically as each processor has less data to process before needing to initiate a reduce. The effect would be clearer for a large network (causing long gradient vectors to be reduced) with few training patterns, however this scenario is not usually encountered due to overfitting. Finally we observe that with a large enough data set to fill the memory of every node, we achieve near linear scaling.

#### 7.3 Price/Performance Ratio

Bunyip was dedicated to running the JOCR problem for four hours with 9,360,000 patterns distributed across 196 processors. Bunyip actually consists of 194 processors, however, we co-opted one of the hot-spare nodes (included in the quoted price) to make up the other two processors. Over this four hour period a total of 2.35 PFlops were performed with an average performance of 163.3 GFlops/s. This performance is sustainable indefinitely provided no other processes use the machine. To calculate the price/performance ratio we use the total cost derived in Section 2.1 of USD$150,913, which yields a ratio of
8. Conclusion

We have shown how a COTS (Commodity-Off-The-Shelf) Linux Pentium III cluster costing under $151,000 can be used to achieve sustained, Ultra-Large-Scale Neural-Network training at a performance in excess of 160 GFlops/s (single precision), for a price/performance ratio of 92.4¢/MFlop/s.

Part of the reason for the strong performance is the use of very large training sets. With the current networking set-up, performance degrades significantly with less data per processor, as communication of gradient information starts to dominate over the computation of the gradient.

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