Highly Scalable Graph Search for the Graph500 Benchmark

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ABSTRACT
Graph500 is a new benchmark to rank supercomputers with a large-scale graph search problem. We found that the provided reference implementations are not scalable in a large distributed environment. We devised an optimized method based on 2D partitioning and other methods such as communication compression and vertex sorting. Our optimized implementation can handle BFS (Breadth First Search) of a large graph with 5.86 (68.7 billion vertices) and 2.40 (1.1 trillion) edges in 10.58 seconds while using 1366 nodes and 16,392 CPU cores. This performance corresponds to 103.9 GF/s. We also studied the performance characteristics of our optimized implementation and reference implementations on a large distributed memory supercomputer with a Fat-Tree-based Infiniband network.

Categories and Subject Descriptors
G.2.2 [Discrete Mathematics]: Graph Theory – Graph algorithms; D.1.3 [Programming Techniques]: Concurrent Programming – Distributed programming

General Terms
Algorithms, Performance.

Keywords
Graph500, BFS, Supercomputer

1. INTRODUCTION
Large-scale graph analysis is a hot topic for various fields of study, such as social networks, micro-blogs, protein-protein interactions, and the connectivity of the Web. The numbers of vertices in the analyzed graph networks have grown from billions to tens of billions and the edges have grown from tens of billions to hundreds of billions. Since 1994, the best known de facto ranking of the world’s fastest computers is TOP500, which is based on a high performance Linpack benchmark for linear equations. As an alternative to Linpack, Graph500 [1] was recently developed. We conducted a thorough study of the algorithms of the reference implementations and their performance in an earlier paper [19]. Based on that work, we now propose a scalable and high-performance implementation of an optimized Graph500 benchmark for large distributed environments.

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Here are the main contributions of our new work:

1. Optimization of the parallel level-synchronized BFS (Breadth-First Search) method to improve the cache-hit ratio through various optimizations in the 2D partitioning, graph structure and vertex sorting.

2. Optimization of the complete flow of the Graph500 as a lighter-weight benchmark with better graph construction and validation.

3. A new traversal record of 103.9 giga-edges/second with our optimized method on the currently 5th-ranked Tsubame 2.0 supercomputer.

4. A thorough study of the performance characteristics of our optimized method and those of the reference implementations.

Here is the organization of our paper. In Section 2, we give an overview of Graph500 and parallel BFS algorithms. In Section 3, we describe the scalability problems of the reference implementations. We explain the proposed optimized BFS method in Section 4, and the optimized graph construction and validation in Section 5. In Section 6, we describe our performance evaluation and give detailed profiles from our optimized method running on the Tsubame 2.0 supercomputer. We discuss our findings in Section 7, review related work in Section 8, and conclude and consider future work in Section 9.

2. GRAPH500 AND PARALLEL BFS ALGORITHMS
In this section, we give an overview of the Graph500 benchmark [1], the parallel level-synchronized BFS method, and then the mapping of this method for the sparse-matrix vector multiplication.

2.1 Graph500 Benchmark
In contrast to the computation-intensive benchmark used by TOP500, Graph500 is a data-intensive benchmark. It does breadth-first searches in undirected large graphs generated by a scalable data generator based on a Kronecker graph [16]. The benchmark has two kernels: Kernel 1 constructs an undirected graph from the graph generator in a format usable by Kernel 2. The first kernel transforms the edge tuples (pairs of start and end vertices) to efficient data structures with sparse formats, such as CSR (Compressed Sparse Row) or CSC (Compressed Sparse Column). Then Kernel 2 does a breadth-first search of the graph from a randomly chosen source vertex in the graph.

The benchmark uses the elapsed times for both kernels, but the rankings for Graph500 are determined by how large the problem
is and by the throughput in TEPs (Traversed Edges Per Second). This means that the ranking results basically depend on the time consumed by the second kernel.

After both kernels have finished, there is a validation phase to check if the result is correct. When the amount of data is extremely large, it becomes difficult to show that the resulting breadth-first tree matches the reference result. Therefore the validation phase uses 5 validation rules. For example, the first rule is that the BFS graph is a tree and does not contain any cycles.

There are six problem sizes: toy, mini, small, medium, large, and huge. Each problem solves a different size graph defined by a Scale parameter, which is the base 2 logarithm of the number of vertices. For example, the level Scale 26 for toy means 2^26 and corresponds to 10^{10} bytes occupying 17 GB of memory. The six Scale values are 26, 29, 32, 36, 39, and 42 for the six classes. The largest problem, huge (Scale 42), needs to handle around 1.1 PB of memory. As of this writing, SuperCane 38 is the largest that has been solved by a top-ranked supercomputer.

2.2 Level-synchronized BFS

All of the MPI reference implementation algorithms of the Graph500 benchmark use a “level-synchronized breadth-first search”, which means that all of the vertices at a given level of the BFS tree will be processed (potentially in parallel) before any vertices from a lower level in the tree are processed. The details of the level-synchronized BFS are explained in [2][3].

<table>
<thead>
<tr>
<th>Algorithm I: Level-synchronized BFS</th>
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<tr>
<td>1 for all vertices v parallel do</td>
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Algorithm I is the abstract pseudocode for the algorithm that implements level-synchronized BFS. Each processor has two queues, CQ and NQ, and two arrays, PRED for a predecessor array and VISITED to track whether or not each vertex has been visited.

At any given time, CQ (Current Queue) is the set of vertices that must be visited at the current level. At level 1, CQ will contain the neighbors of r, so at level 2, it will contain their pending neighbors (the neighboring vertices that have not been visited at levels 0 or 1). The algorithm also maintains NQ (Next Queue), containing the vertices that should be visited at the next level. After visiting all of the nodes at each level, the queues CQ and NQ are swapped at line 16.

VISITED is a bitmap that represents each vertex with one bit. Each bit of VISITED is 1 if the corresponding vertex has been already visited and 0 if not. PRED has predecessor vertices for each vertex. If an unvisited vertex v is found at line 12, the vertex u is the predecessor vertex of the vertex v at line 14. When we complete BFS, PRED forms a BFS tree, the output of kernel2 in the Graph500 benchmark.

The Graph500 benchmark provides 4 different reference implementations based on this level-synchronized BFS method. Their details and algorithms appear in an earlier paper [19]. However we found out that the fundamental concept of the level synchronized BFS can be viewed as a sparse-matrix vector multiplication. With reference to the detailed algorithmic explanations in [19], we only explain the basic BFS method here.

2.3 Representing Level-Synchronized BFS as Sparse-Matrix Vector Multiplication

The level-synchronized BFS in II-B is analogous to a Sparse-Matrix Vector multiplication (SpMV) [20] which is computed as y = Ax where x and y are vectors and A is a sparse matrix.

A is an adjacency matrix for a graph. Each element of this matrix is 1 if the corresponding edge exists and 0 if not. The vertex x corresponds to CQ (Current Queue) where x(v) = 1 if the vertex v is contained in CQ and x(v) = 0 if not. x(v) means the v-th element of vector x. Then the neighboring vertices of CQ can be represented as the vertex v where y(v) ≥ 1. We get the neighboring vertices from SpMV. Then we can compute the lines 12 to 15 in the Algorithm I.

2.4 Mapping Reference Implementations to SpMV

In a distributed memory environment, the graph data and vertex data must be distributed. There are four MPI-based reference implementation of Graph500: replicated-csr (R-CSR), replicated-csc (R-CSC), simple (SIM), and one_sided (ONE-SIDED).

All four of the reference implementations use 1D partitioning method. The method of vertex distribution is the same in these four implementations. Assume that we have a total of P processors. VISITED, PRED and NQ are simply divided into P blocks and each processor handles one block. There are two partitioning methods for the sparse matrix shown in Figure 1. For the Figure 1, we assume that the edge list of the given vertex is a column of the matrix. With the vertical partitioning (left), each processor has the edges incident to the vertices the processor owns. With the horizontal partitioning (right), each processor has the edges emanating from the vertices the processor owns. Figure 1 also shows how SpMV can be computed in parallel with P processors. The computation of the reference implementations can be abstracted as the computation of SpMV. R-CSR and R-CSC use vertical partitioning for SpMV while SIM and ONE-SIDED use horizontal partitioning for SpMV.

\[
\begin{bmatrix}
A_1 \\
A_2 \\
\vdots \\
A_P
\end{bmatrix} \times x \quad \begin{bmatrix}
A_1 & A_2 & \cdots & A_P
\end{bmatrix} \times x
\]

Figure 1. SpMV with vertical and horizontal partitioning.
3. SCALABILITY ISSUES OF REFERENCE IMPLEMENTATIONS

In this section we give overviews of three reference implementations of Graph500, R-CSR (replicated-csr), R-CSC (replicated-csc), and SIM (simple) and we use experiments and quantitative data to show why none of these algorithms can scale well in large systems.

Before moving to the detailed descriptions of each implementation, we need to cover how CQ (Current Queue) is implemented in the reference implementations. The reference implementations also use bitmaps with one bit for each vertex in CQ and NQ. In these bitmaps, if a certain vertex v is in a queue (CQ or NQ), then the bit that corresponds to that vertex v is 1 and if not, then the bit is 0.

We categorize the reference implementations into replication-based and non-replicated methods, which are described in Sections 3.1 and 3.2. Then in Section 3.3 we present a scalable BFS approach with 2D partitioning.

3.1 Replication-based method

3.1.1 Algorithm Description

For the R-CSR and R-CSC methods that divide an adjacency matrix vertically, CQ is duplicated to all of the processes. Then each processor independently computes NQ for its own portion.

Copying CQ to all of the processors means each processor sends its own portion of NQ to all of the other processors. CQ (and NQ) is represented as a relatively small bitmap. For relatively small problems with limited amounts of distribution, the amounts of communication data are reasonable and this method is effective.

However, since the size of CQ is proportional to the number of vertices in the entire graph, this copying operation leads to a large amount of communication for large problems with large distribution.

3.1.2 Quantitative Evaluation for Scalability

Figure 2 shows the communication data volume for each node with the replication-based implementation and SCALE of 26 for each node as the problem size. This is a weak-scaling version, and computes the theoretical results when using 2 MPI processes per node. In such a weak-scaling setting, the number of vertices increases in proportion to the increasing number of nodes. This result clearly shows that the Replication-based method is not scalable for a large distributed environment.

3.2 Non-replicated Method

3.2.1 Algorithm Description

The simple reference implementation or SIM that divides an adjacent matrix in a horizontal fashion, locates the portion of the CQ by dividing it into \( P \). The NQ queue is already divided into \( P \) blocks, and so each processor can use the NQ queue from the previous level as the CQ of the current level.

Edge lists of the vertices in CQ are merged to form a set \( N \) of edges. The edges of a set \( N \) are the edges emanating from the vertices in CQ and incident to the neighboring vertices. These \( N \) include both the edges to be handled by the local processor and by other processors. The edges incident to the vertex owned by remote processors are transmitted to those processors.

The number of edges to be transmitted to remote processors can be up to the number of edges of the adjacency matrix owned by the sender-side processor. Thus the communication data volume is constant without regard to the number of nodes.

However, the Replication-based method with vertical partitioning transmits CQ as a bitmap. On the other hand in the simple implementation, SIM needs to send edges, pairs of a CQ vertex and neighboring vertex because the predecessor vertex is needed to update PRED, the predecessor array. Therefore, the Replication-based approach with vertical partitioning is better than the simple approach in a small-scale environment with fewer nodes.

3.2.2 Quantitative Evaluation for Scalability

The other two reference implementations with horizontal partitioning are called “simple” and “one-sided”. In these implementations all-to-all communication that sends a different data set to each of the other processors is needed when sending the set of N edges. This all-to-all communication is not scalable for large distributed environments.

Figure 3 shows the communication speed of the all-to-all communication on Tsubame 2.0 when using 4 MPI processes per node. We used MVAPICH2 1.6 as the MPI implementation. The communication was implemented with the MPI_Alltoall function, and we used three different transmission buffer sizes, 64 MB, 256 MB, and 1,024 MB.

The amount of data that each node transmits to other nodes is 4 MB when using the 1,024 MB buffer size, 64 nodes, and 256 MPI processes. The results shown in Figure 3 show that the all-to-all communication is not scalable.

Figure 2. Theoretical message size per node (GB)

Figure 3. Average data rate per node with all-to-all communication
With 512 nodes, the performance is quite slow even with a small buffer size such as 64 MB. Also, even if we use 1,024 MB as the buffer size, the performance is 1/4 of 32 nodes.

Our experimental testbed, Tsubame 2.0, uses a Fat-Tree structure with its Infiniband network. If the theoretical peak performance were achieved, there would be no performance degradation even for all-to-all communication is among all of the nodes. However, the communication latency cannot be ignored in a large system. The actual performance is always less than the theoretical maximum.

3.3 Scalable Approach: 2D Partitioning-Based BFS
To solve the scalability problems described in Section 3.1 and Section 3.2, a scalable distributed parallel breadth-first search algorithm was proposed in [4]. Their scalable approach uses a level-synchronized BFS and 2D partitioning technique to reduces the communication costs since it can handle both vertical and horizontal partitioning, unlike 1D partitioning. Our proposed method optimizes this 2D partitioning technique and also uses some other optimization techniques. Here is a brief overview of the 2D partitioning technique.

Assume that we have a total of \( P \) processors, where the \( P = R \cdot C \) processors are logically deployed in a two-dimensional mesh which has \( R \) rows and \( C \) columns. We use the terms processor-row and processor-column with respect to this processor mesh. Adjacency matrix is divided as shown in Figure 4 and the processor \((i, j)\) is responsible for handling the \( C \) blocks from \( A_{i,j}^{(1)} \) to \( A_{i,j}^{(C)} \). The vertices are divided into \( R \cdot C \) blocks and the processor \((i, j)\) handles the \( k \)-th block, where \( k \) is computed as \((j-1) \cdot R + i\).

![Figure 4. 2D Partitioning Based BFS [4]](image)

Each level of the level-synchronized BFS method with 2D partitioning is done in 2 communication phases called “expand” and “fold”. In the expand phase, every processor copies its CQ to all of the other processors in the same processor-column, similar to vertical 1D partitioning. Then the edge lists of the vertices in CQ are merged to form a set N. In the fold phase, each processor sends the edges of N to the owner of their incident vertices, similar to horizontal 1D partitioning. With the 2D partitioning, these owners are in the same processor-row. 2D partitioning method is equivalent to a method of combining the two types of 1D partitioning. If \( C \) is 1, this corresponds to the vertical 1D partitioning and if \( R \) is 1, it corresponds to the horizontal 1D partitioning.

The advantage of 2D partitioning is to reduce the number of processors that need to communicate. Both types of 1D partitioning require all-to-all communication. However, with the 2D partitioning, each processor only communicates with the processors in the processor-row and the processor-column.

4. U-BFS: SCALABLE BFS METHOD
Our BFS implementation is based on a 2D partitioning method and is highly optimized for a large-scale distributed computing environment by using various optimization techniques. In this section we present these techniques. As described in Section 3.3, in the 2D partitioning, the \( P = R \cdot C \) processors are deployed in a \( R \cdot C \) mesh.

4.1 Overview of Optimized 2D Partitioning-based BFS (U-BFS)
Our U-BFS optimization also uses the 2D partitioning technique. The communication method of the expand phase is the same approach as R-CSC, one of the reference implementations. Each set of vertices in CQ is represented as a bitmap and each processor gathers CQ at each level by using the MPI_AllGather function.

The communication of the fold phase is optimized in our implementation by compression of the data described in Section 4.2. Because the amount of data communicated in each fold phase is much larger than that of the expand phase when we use a naive method, compression of the data is important.

We divide the processing of the fold phase into senders and receivers. The senders send the edges emanating from the vertices in CQ as compressed data. The receivers receive data from the senders, decompress the data to edges, and process edges. Both senders and receivers are handled by multiple parallel threads and the communication can be done asynchronously. Our proposed method has highly efficient processing with this parallel and asynchronous communication.

In addition, our method can efficiently utilize the CPU caches by vertex sorting (Section 4.4), binding the threads to CPU cores (Section 4.3), and then giving higher priority to the receiver threads (Section 4.3) to reduce the cache replacements from thread switching.

The algorithm appears as Algorithm II. CQ and NQ are bitmaps. In Lines 1-2, NQ is initialized and the BFS root is inserted into NQ in line 3. Lines 5-10 are done by all of the processors. Line 6 is the expand communication. In Line 7, Task A and Task B run in parallel. Task A is the receiver processing and Task B is the sender processing.
Algorithm II: Optimized 2D partitioning algorithm

Variables: MAP which is described in the Section 4.4 is for conversion from the sorted number to the original number.

Main
1. for all vertices $u$ in $\text{NQ}$ do
2. $\text{NQ}[u] \leftarrow 0$
3. $\text{NQ}[\text{root}] \leftarrow 1$
4. fork;
5. for level = 1 to $\infty$
6. $\text{CQ} \leftarrow \text{all gather NQ in this processor-column}$
7. parallel $\text{Task A}$ and $\text{Task B}$
8. $\text{Synchronize}$;
9. if $\text{NQ} = \emptyset$ for all processors then
10. | terminate loop;
11. join;

Task A (sender)
1. for each vertex $u$ in $\text{CQ}$ parallel do (contiguous access)
2. if $\text{CQ}[u] = 1$ then
3. | for each vertex $v'$ adjacent to $u$ do
4. | compress the edge $(u, v')$ and send it to the owner of vertex $v'$

Task B (receiver)
1. for each received data parallel do
2. | decompress the data and get the edge $(u, v')$
3. | if $\text{visited}[v'] = 0$ then
4. | $\text{VISITED}[v'] \leftarrow 1$;
5. | $v \leftarrow \text{MAP}[v']$;
6. | $\text{PRED}[v] \leftarrow u$;
7. | $\text{NQ}[v] \leftarrow 1$;

4.2 Optimized Communication with Data Compression

The compression of communication data is greatly important because the bottleneck of the distributed BFS is often the communication. We optimized the communication of the fold phase because the amount of data communicated in the fold phase is much larger than that of the expand phase when we use a naive method. In the fold phase, sender side sends the edges emanating from the vertices in CQ. The edges are represented as a list of a tuple $(u, v)$ where $u$ is the vertex of CQ and $v$ is the neighboring vertex of CQ. We compress vertex $u$ and $v$ with different compression techniques.

Here is our compression method for vertex $u$. Since CQ is a bitmap, by checking CQ in order of bitmap array, the vertices in $u$ will be sorted in ascending-order. Thus the average difference between two vertex $u$ in successive edges will be small. The vertex $u$ is represented by simply encoding the difference from the prior tuple with variable-length quantity which is used by the standard MIDI file format and Google’s protocol buffers etc. We use general variable-length quantity (VLQ) for unsigned integer. An integer value will be one or more bytes. Smaller numbers will be smaller number of bytes. The most significant bit (MSB) of each byte indicates whether another VLQ byte follows. If the MSB is 1, another VLQ byte follows and if 0, this is the last byte of the integer. The rest 7 bits of each byte form an integer. The least significant will be first in a stream. For example, we assume that the CQ bitmap is ‘01100001’ which means three vertices 1, 2, and 7 are in the CQ and we also assume that the numbers of edges emanating from the vertices 1, 2, and 7 are 2, 1, and 3 respectively. The series of vertices $u$ will be ‘1’, ‘1’, ‘2’, ‘7’, ‘7’, ‘7’ because we check the CQ in order. Then we will get the difference from the prior vertex, ‘-’, ‘0’, ‘1’, ‘5’, ‘0’, ‘0’. With VLQ, a smaller number will be encoded into smaller number of bytes. Therefore, we can compress the vertices $u$.

For the compression of vertex $v$, each edge $(u, v)$ will be sent to the owner of vertex $v$. We distribute these vertices with a round robin method. The owner of vertex $v'$ is computed by dividing $v$ by $P$. For example, the owner of vertex $v$ ($= v_{\text{vnode}} \cdot P + v_{\text{proc}}$) is $v_{\text{vnode}}$. Thus by sending $v_{\text{vnode}} = v/P$ instead of $v$, we can reduce the size of the data for vertex $v$.

The data representation of vertex $u$ is compressed by using the differences from the preceding data. However to support multi-threaded processing, we need to introduce a packet, a unit which can be decoded independently. If we send a series of edges in one packet containing (300, 533), (301, 12), (301, 63), (303, 1222) as $(u, v)$, then the data representation of the vertex $u$ without VLQ encoding, but with the difference technique is “300, 1, 0, 2”. With the VLQ encoding, it becomes “0xAC, 0x01, 0x01, 0x02”.

When there are $P$ processors, the data representation of the vertex $v$ is “8, 0, 0, 19”. Without this optimization, the size of each data tuple $(v, u)$ would be 16 bytes, since each vertex is represented as 8 bytes, but with this optimization, the data size is about 5 bytes on average, since the vertex $v$ consumes 4 bytes and the vertex $u$ consumes about 1 byte as long as the difference between the contiguous values is less than 128 with the VLQ encoding.

4.3 Parallelizing the Sender and Receiver Processing

Our optimized implementation uses hybrid parallelism of MPI and OpenMP for communication and computational efficiency. As our multi-threading strategy to use N CPU cores for MPI processes, we run (N-1) threads for the sender-side processing (Task A) and (N-1) threads for the receiver side (Task B), and only 1 thread for communication. In total, (2N-1) threads run in parallel. The communication dataflow in the fold phase appears in

![Figure 5. A dataflow diagram of the fold communication.](image-url)
Figure 5. Each thread at the sender side reads the graph data and CQ, compress edges and then store compressed edge data into the packet buffer prepared for each sender thread.

Once the packet buffer is full, the data is passed to the communication thread and emitted with the asynchronous MPI sender function MPI_Isend. When a communication thread receives the data, it creates a packet index for multi-threaded processing and passes it to the receiver thread. Once the receiver thread receives the packet index and the data, the thread decodes it, updates the VISITED bitmap, and creates NQ.

The receiver thread decodes the incoming edge \((v, u)\) where \(u\) is the vertex of CQ and \(v\) is the neighboring vertex of CQ, determines whether vertex \(v\) was already visited by checking the VISITED bitmap, and if not, updates the VISITED bitmap and NQ. However since the incoming vertices \(v\) are received at random, the random access to the VISITED bitmap can degrade the performance. We optimize updating VISITED bitmap in the next section.

We also optimize for the NUMA architecture. If we have \(N\) CPU cores for MPI processes, then \((2N – 1)\) threads are created. Unless these threads are bound to a CPU, the OS scheduler will allocate more than \(N\) CPU cores. For a multi-socket CPU like that of Tsubame 2.0, if a thread is moved to another CPU core at a different CPU socket, then the cache hit ratio is reduced. Also, for a NUMA architecture, the memory and the CPU where the data processing thread is running should be kept close together.

To reduce cache misses due to thread switches, the threads for the receiver side have higher priority to the threads for the sender side. This thread priority also reduces the amount of buffers required for receiver processing.

### 4.4 Improving Cache Hit Ratio with Vertex Sorting

We explained in the previous section that each receiver thread (Task B) generates random memory access to the VISITED bitmap, which lowers the cache hit ratio. The distribution of the degrees follows Zipf’s law as observed in our earlier work [19], since Graph500 uses a Kronecker graph [16]. The frequency of access to VISITED is determined by the vertex degree (the number of edges of each vertex). If the degree of vertex \(v\) is high, then \(v\)th element of VISITED will be accessed more frequently. The high cost of random memory access can be reduced by sorting the vertices in decreasing order of degrees. We call this optimization Vertex Sorting.

To optimize accessing the VISITED bitmap, we use the sorted order in the row index of the adjacent matrix. We do not use the sorted order in the column index. If we use the sorted order in both row and column index, the conversion of PRED from the sorted order to original order is necessary. This conversion needs all-to-all collective communication which we need to pay expensive cost. Therefore, in the adjacent matrix, the row index and the column index for the same vertex are different.

In Algorithm II, \(v’\) is the sorted number and \(v\) is the original number. The number we get from the adjacent matrix is the sorted one. We change the sorted number to the original one in line 5 of Task B (receiver-side processing) by using MAP. In the 2D partitioning, vertices are divided into \(R \cdot C\) blocks and each block is allocated to a processor. We sort the vertices only in a local block. Therefore, Vertex Sorting does not affect the vertex partitioning and the graph partitioning. The adjacent matrix and the MAP in Algorithm II are created in the construction phase (kernel 1). The method of sorting the vertices in the construction phase is described in Section 5.1.

### 5. GRAPH CONSTRUCTION FOR U-BFS AND OPTIMIZING VALIDATION

In this section we present the method of the construction phase for optimized BFS described in the previous section and the method of optimizing validation.

The current Graph500 benchmark must conduct 64 iterations of the BFS executions and validations. The validation phase dominates the overall time of the Graph500 benchmark, and so it is critically important to accelerate this phase to speed up the entire experiment.

#### 5.1 Graph Construction

In the construction phase, we construct a sparse matrix representation of the graph from the edge list generated in the generation phase. Our optimized algorithm of the construction phase is shown in Algorithm III.

**Algorithm III: Graph Construction**

**Input:** L is a generated edge list.

**Output:** Two arrays of a sparse matrix representation for the graph, P and V. P is a pointer array and V is an index array.

MAP for conversion from the sorted number to the original number.

C and I are arrays that have the same length of \(P\). S is an array whose length is the number of local vertices.

```plaintext
1 for each column \(v\) of local adjacency matrix
2 | \(P[v]\) ← 0
3 | \(C[v]\) ← 0
4 | \(I[v]\) ← 0
5 for each local vertex \(v\)
6 | \(S[v]\) ← 0
7 for each edge \((v0, v1)\) in L
8 | send edge \((v0, v1)\) to its owner
9 | send edge \((v1, v0)\) to its owner
10 for each received edge \((u, v)\)
11 | \(C[v]\) ← \(C[v]\) + 1
12 | send vertex \(v\) to its owner
13 for each received vertex \(v\)
14 | \(S[v]\) ← \(S[v]\) + 1
15 sort \(S\) and create MAP for conversion from sorted number to original number and MAP for inverse conversion
16 for \(v\) is 1 to the number of columns of local adjacency matrix
17 | \(P[v]\) ← \(P[v-1]\) + \(C[v]\)
18 for each edge \((v0, v1)\) in L
19 | send edge \((v0, v1)\) to its owner
20 | send edge \((v1, v0)\) to its owner
21 for each received edge \((u, v)\)
22 | send vertex \(u\) to its owner
23 | receive vertex \(u\) and send MAP[\(u\)] to sender processor
24 | receive MAP[\(u\)] as \(u’\)
25 | \(V[\(P[v]\) + \(I[v]\)] ← u’\)
26 | \(I[v]\) ← \(I[v]\) + 1
```

In our implementation, the matrix is 2D partitioned. Both edges and vertices have their owner processor. In Algorithm III, local
adjacency matrix is a portion of the adjacency matrix allocated to the processor and local vertices are the vertices the processor owns.

5.2 Validation
The validation determines the correctness of the BFS result based on the edge tuples generated in the graph generation phase. By profiling the validation phase of the reference implementation, we found two validation rules in the Graph500 specification dominating the all-to-all communications.

1) Each edge in the input list has vertices with levels that differ by at most one or neither is in the BFS tree.
2) Each node and its parent are joined by an edge in the original graph.

A processor that owns an edge tuple \((v_0, v_1)\) needs to communicate with the owner processor of \(v_0\) and the owner processor of \(v_1\). Implemented in a naive fashion, this requires all-to-all communication involving all of the processors, which is not scalable. We devised an approach that divides the edge tuples with 2D partitioning and allocates them to each processor before the first BFS execution. The number of processors involved in communication is fewer than the original version, making the work scalable.

6. PERFORMANCE EVALUATION
We used Tsubame 2.0, the fifth fastest supercomputer in the TOP500 list of June 2011, to evaluate the scalability of our optimized implementation.

6.1 Overview of the Tsubame 2.0 supercomputer
Tsubame 2.0 is a production supercomputer operated by the Global Scientific Information and Computing Center (GSIC) at the Tokyo Institute of Technology. Tsubame 2.0 has more than 1,400 compute nodes interconnected by high-bandwidth full-bisection-wide Infiniband fat nodes.

Each Tsubame 2.0 node has two Intel Westmere EP 2.93 GHz processors (Xeon X5670, 256-KB L2 cache, 12-MB L3), three NVIDIA Fermi M2050 GPUs, and 50 GB of local memory. The operating system is SUSE Linux Enterprise 11. Each node has a theoretical peak of 1.7 teraflops (TFLOPS). The main system consists of 1,408 computing nodes, and the total peak performance can reach 2.4 PFLOPS. Each of the CPUs in Tsubame 2.0 has six physical cores and supports up to 12 hardware threads with Intel’s hyper-threading technology, thus achieving up to 76 gigaflops (GFLOPS).

The interconnect that links the 1,400 computing nodes with storage is the latest QDR Infiniband (IB) network, which has 40 Gbps of bandwidth per link. Each computing node is connected to two IB links, so the communication bandwidth for the node is about 80 times larger than a fast LAN (1 Gbps). Not only the link speed at the endpoint nodes, but the network topology of the entire system heavily affects the performance for large computations. Tsubame 2.0 uses a full-bisection fat-tree topology, which handles applications that need more bandwidth than provided by such topologies as a torus or mesh.

6.2 Evaluation Method
In the software environment we used gcc 4.3.4 (OpenMP 2.5) and MVAPICH2 version 1.6 with a maximum 512 nodes. Tsubame 2.0 is also characterized as a supercomputer with heterogeneous processors and a large number of GPUs, but we did not use those parts of the system. Each node of Tsubame 2.0 has 12 physical CPU cores and 24 virtual cores with SMT (Simultaneous Multithreading). Our implementation treats 24 cores as a single node and the same number of processors is allocated to each MPI process.

In our experiments, the 2D-partitioning-based processor allocation \(R \cdot C\) was per Table 1. \(R\) and \(C\) were determined with the policy of allocating division numbers as similarly as possible. The number of MPI processes should be a power of two and the value of \(R\) and \(C\) was determined by the MPI processes irrespective of the number of nodes.

The result in this paper is the maximum performance of 16 iteration BFS runs. But the result of the Graph500 list is the median of 64 BFS runs. Therefore, there is a little difference between the performance result in this paper and our official score of the Graph500 list.

| Table 1. The values of \(R\) and \(C\) with the # of MPI processes |
|--------------------------|-----------------|----------------|-----------------|-----------------|-----------------|
| # of MPI processes | R   | C     | R   | C     | R   | C     | R   | C     | R   | C     |
|--------------------------|-----------------|-----------------|-----------------|-----------------|-----------------|
| 1                        | 1           | 1               | 2           | 2               | 4           | 4               | 8           | 8               | 16            | 16            |
| 2                        | 1           | 2               | 2           | 4               | 4           | 8               | 8           | 16               | 16            | 32            |
| 4                        | 8           | 8               | 16          | 16              | 32           | 32              | 32           | 32               | 32            | 64            |

6.3 Performance of U-BFS
Figure 6 shows the performance of U-BFS in a weak-scaling fashion with SCALE 26 per node. We use 2 MPI processes for each node. We get the performance of 99.0 GE/s (TEPS) with 1024 nodes and SCALE 36. We also conducted the experiment with 1366 nodes and our optimized implementation achieved 103.9 GE/s (TEPS) with 1366 nodes (16,392 CPU cores) and SCALE 36.

6.4 Comparison with Reference Implementation
We compared U-BFS with the latest version (2.1.4) of the reference implementations in Figure 7 and Figure 8. This experiment was done in a weak-scaling fashion, so the problem size for each node was held constant, SCALE 24 in Figure 7 and SCALE 26 in Figure 8. The horizontal axis is the number of nodes and the vertical axis is the performance in GE/s.

U-BFS and the two reference implementations, R-CSR and R-CSC use 2 MPI processes for each node. The reference implementation, simple, uses 16 MPI processes for each node, since the implementation does not use multithreading parallelism. As shown in the graph, there were some results that could not be measured due to problems in the reference implementations. When the number of nodes is increased for SIM, the system ran out of memory. With R-CSR and SCALE 32, a validation error occurred and there was a segmentation fault at higher SCALE values. With R-CSC, the construction phase crashes above SCALE 34. Figure 7 and Figure 8 show that U-BFS outperformed R-CSC and SIM for all of the nodes. With numbers of nodes fewer than 32, the R-CSR implementation is best, but our method shows performance advantages with more than 32 nodes. For example, our optimized method is 2.8 times faster than R-CSR with 128 nodes and SCALE 26 for each node. (All of the final problem sizes were SCALE 33.)
Figure 6. Performance of optimized BFS (U-BFS) (Scale 26 per node)

Figure 7. Comparison with Reference implementations (Scale 24 per node)

Figure 8. Comparison with Reference implementations (Scale 26 per node)

Figure 9. Effect of each optimization

Figure 10. Comparing Communication Data Volume

Figure 11. Performance breakdown of the overall execution times

Figure 12. Execution times without communication wait times

Figure 13. Performance comparison with varying number of MPI processes (1, 2, 4, and 8) per node

Figure 14. Performance breakdown with 64 nodes and varying number of MPI processes per node

Figure 15. Performance with varying problem size per node
6.5 Effect of Each Optimization

Figure 9 shows the performance effects of each of our proposed optimization techniques. For the reference data, we also show the performance of SIM. The Fold phase, which is the main computation part of 2D partitioning-based algorithm, uses the same approach as the simple reference implementation. Therefore, our optimization should show similar performance characteristics.

We also prepared a “base” version as a multi-threaded implementation with 2D partitioning. This version parallelizes the senders and receivers, and also uses asynchronous communication. Even without any optimizations, this almost doubles the performance over the simple reference implementation.

Compared with the base version, the communication compression technique (Section 4.2) boosts the performance by 2 to 3 times. We also had a maximum of a 27% performance improvement due to the CPU affinity and the technique (Section 4.3) of giving higher priority to the receiver processing. Another 10% came from the technique of vertex sorting (Section 4.4). However, these optimizations except for the communication compression were not effective with larger numbers of nodes because the network communications became the major bottleneck.

Figure 10 compares the communication data volume of our optimized method and the reference implementation in a weak-scaling setting. The vertical axis is the transmitted data volume (GB) per node involved in one traversal of BFS with the SCALE 26 problem size. This profiling used 2 MPI processes per node.

The data volume of the Replication-based reference implementations including R-CSC and R-CSR is a theoretical value since we could not measure the data with more than 256 nodes because of the limitations of the reference implementations. This graph shows the measured data for the 2D-partitioning-based optimization methods comparing communication compression and no compression.

The communication data volume increases in proportion to the number of nodes for the weak-scaling setting since the Replication-based implementation needs to send CQ to all of the other processes. Meanwhile, the 2D-based partitioning method is scalable since the communication data volume becomes relatively smaller. With the data compression technique used by U-BFS, the communication data volume can be reduced to around one third of the base version. For example, the data volume was reduced from 32.3 GB to 13.1 GB with 512 nodes.

Figure 11 shows the breakdown of the execution time with U-BFS in a weak-scaling setting. The vertical axis is the execution time of BFS with SCALE 26 for a single node that runs 2 MPI processes. Since the communication at the fold phase is asynchronous, the communication waits only occur when the communication cannot keep up with the computation. The synchronization wait is the waiting time when synchronizing with all of the other nodes just before the level-synchronized BFS moves to the next step.

In our profiling results, the communication costs for the Expand phase is relatively small. A large part of the execution time is spent in the computations. However with more than 16 nodes, the communication wait time is increasing, which increases the overall execution time. This means the bottleneck of U-BFS with more than 16 nodes is the communication. More analysis appears in the Discussion section.

With the profiling result, the overall execution time, which excludes the communication wait time of each optimization, is shown in Figure 12. This result shows that the execution time without the communication wait time decreases as expected with our optimization techniques. This is also observed even with relatively large numbers of nodes where communications becomes the bottleneck.

6.6 Performance comparison with varying number of MPI processes

Figure 13 shows the performance characteristics when varying the number of MPI processes per node in a weak-scaling setting. The result does not show great differences between 2 or 4 MPI processes, but the performance is degraded with 1 and 8 MPI processes.

Figure 14 shows that the processing time at the receiver side decreases with larger numbers of nodes. The processing time at the receiver side (which requires the random access to the visited bitmap) can be reduced by increasing the cache hit ratio, since the number of vertices allocated for each MPI process decreases with more MPI processes.

Meanwhile the processing time at the sender side is increasing with larger numbers of MPI processes. This is because the number of CPU cores allocated for each MPI process, N, is decreased since (N-1) threads are running as senders. The communication time is also increased for 1 and 8 MPI processes. For these reasons, performance degradation is seen with 1 and 8 MPI processes per node.

6.7 Performance with varying problem size

per node

Figure 15 shows the performance characteristics as the problem size changes. U-BFS executes 2 MPI processes per node. The maximum problem size U-BFS can compute on the Tsuname 2.0 environment is SCALE 27. The experimental results shown in Figure 15 show the best performance is obtained with SCALE 26 per node when using 128 and 256 nodes. With 512 nodes, the SCALE 27 problem size shows the best performance. Also, we ran the same experiments with less than 64 nodes for reference, although our proposed optimization targets large environments. SCALE 26 per node shows the best performance with 1, 2, 4, 8, 16, 32, and 64 nodes.

6.8 Profiling Execution Time and Communication Data Size at Each Level

Figure 16 compares the reference implementations, R-CSC and R-CS, with U-BFS in terms of the execution time at each level of the level-synchronized BFS method. Note that the reference implementations only use 32 nodes and our optimized method uses 128 nodes, but the problem size per node is the same, so this is a fair comparison.

The Kronecker graph adopted by the Graph500 benchmark is a scale-free graph. With such a graph, the search range becomes greatly expanded once it reaches vertexes with high degrees that have large numbers of edges. Since R-CSR computes the CSR-based algorithm, the execution time at each level depends on the number of unvisited vertexes.

Therefore this method consumes more time in the shallow portions. In contrast, R-CSC and U-BFS use the CSC-based method. Unlike the CSR-based method, the execution time at each level is almost proportional to the number of adjacent vertices of CQ.
7. DISCUSSION

7.1 Performance with Fat-Tree Network Topology

The performance evaluation of our method in a weak scaling setting reveals that the performance per node is degraded by increasing numbers of nodes. According to the profiling result, we also found that the bottleneck is the communication and this performance result came from the increased communication waits.

The network topology of Tsubame 2.0 is a Fat-Tree. Therefore if the theoretical peak performance is achieved, even all-to-all collective communication with all of the nodes should have the same performance as two arbitrary nodes communicating with each other, since the communication path between arbitrary pairs of nodes will not overlap. However in reality, the communication paths interfere when the communication between nodes is simultaneous, which leads to performance degradation. This interfere is described in [22] as a hotspot.

Figure 20 is the average transmission rate per node with U-BFS. This value is measured by simply dividing the entire data transmission rate by the BFS execution time. The result shows that the transmission rate is decreasing with increasing numbers of nodes.

We also measured the effect on other group communication when the groups, each consisting of 32 nodes, did all-to-all collective communication to all of the involved nodes transmitting different data to the other nodes). The result is shown in Figure 21. In the figure, 2 and 4 groups of 32 nodes simultaneously did all-to-all collective communication with 64 nodes and 128 nodes, respectively. No communication is done between any arbitrary nodes. This experimental result reveals and verifies that...
communication by other groups influences the transmission data rate due to the hotspot.

Therefore the performance degradation of our optimized method with increased numbers of nodes is caused by the limitations of the Fat-Tree-based network topology. The overlap problem of this Fat-Tree communication routing has been investigated by various methods, and the unexpected communication degradation shown in Figure 9 might possibly solved by applying some of these approaches.

7.2 Comparing with 3D-torus based systems
Our optimized approach achieves 103.9 GE/s as TEPS with SCALE 36 and 1366 nodes of Tsubame 2.0. When comparing this value with other systems based on the Graph500 benchmark results announced in November 2011, the top-ranked system achieves 254.3 GE/s as TEPS with SCALE 32 and 4,096 nodes of BlueGene/Q. Another leading TEPS score was the system called Hopper that achieves 113.4 GE/s with SCALE 37 and 1,800 nodes.

8. RELATED WORK
Yoo [4] presents a distributed BFS scheme with 2D graph partitioning that scales on the IBM BlueGene/L with 32,768 nodes. Our implementation is based on their distributed BFS method but we optimized the method further. Bader [3] describes the performance of optimized parallel BFS algorithms on multithreaded architectures such as the Cray MTA-2.

Aydin [21] conduct the performance evaluation of a distributed BFS with 1D partitioning and 2D partitioning on the Cray XE6 and the Cray XT4. His work is similar to our work but his method of 2D partitioning is different from Yoo[4]’s method and his method needs additional communication that degrade the performance of BFS. Their achieved score in [21] was only 17.8 GE/s on Hopper with 40,000-cores.

Agarwal [2] proposes an efficient and scalable BFS algorithm for commodity multicore processors such as the 8-core Intel Nehalem EX processor. With the 4-socket Nehalem EX (Xeon 7560, 2.26 GHz, 32 cores, 64 threads with HyperThreading), they ran 2.4 times faster than a Cray XMT with 128 processors when exploring a random graph with 64 million vertices and 512 million edges, and 5 times faster than 256 BlueGene/L processors on a graph with an average degree of 50. The performance impact of their proposed optimization algorithm was tested only on a single node, but it would be worthwhile to extend their proposed algorithm to larger machines with commodity multicore processors, which includes Tsubame 2.0.

Harish [10] devised a method of accelerating single-source shortest path problems with GPGPUs. Their GPGPU-based method solves the breadth-first search problem in approximately 1 second for 10 million vertices of a randomized graph where each vertex has 6 edges on average. However, the paper concluded that the GPGPU-method does not match the CPU-based implementation for scale-free graphs such as the road network of the 9th DIMACS implementation challenge, since the distribution of degrees follows a power law in which some vertices have much higher degrees than others. However since the top-ranked supercomputers in TOP500 have GPGPUs for compute-intensive applications, it would be worthwhile to pursue the optimization of Graph500 by exploiting GPGPUs.

9. CONCLUDING REMARKS AND FUTURE WORK
In this paper we proposed an optimized implementation of the Graph500 benchmark in a large-scale distributed memory environment. The reference code samples provided by the Graph500 site were neither scalable nor optimized for such a large environment. Our optimized implementation is based on the level-synchronized BFS with 2D partitioning and we propose some optimization methods such as communication compression and vertex sorting. Our implementation does 103.9 GE/s as TEPS (Traversals Edges Per Second) with SCALE 36 and 1366 nodes of Tsubame 2.0. This score is 3rd score in the ranking list announced in November 2011. We found the performance of our optimized BFS is limited by the network bandwidth. We also propose approaches for optimizing the validation phase, which can accelerate the overall benchmark. Many of our proposed approaches in this paper can also be effective for other supercomputers such as Cray and BlueGene. For future work we will show the effectiveness of our implementation in other large systems.

10. REFERENCES
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