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A Parallel Scheme Using the Divideand-Conquer Method

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Abstract. A parallel scheme using the divide-and-conquer method is developed. This partitions the input set of a problem into subsets, computes a partial result from each subset, and finally employs a merging function to obtain the final answer. Based on a linear recursive program as a tool for formalism, a precise characterization for problems to be parallelized by the divide-and-conquer method is obtained. The performance of the parallel scheme is analyzed, and a necessary and sufficient condition to achieve linear speedup is obtained. The parallel scheme is generalized to include parameters, and a real application, the *fuzzy* join problem, is discussed in detail using the generalized scheme.

1. Introduction

The *Divide-and-Conquer* method is a common approach in designing efficient algorithms [3, 4, 8]. Intuitively, in this approach, a problem is decomposed into several subproblems of smaller sizes, each of which is similar or identical to the original problem; a partial solution is produced for each subproblem; then, the solution to the original problem is obtained by merging the partial solutions to the subproblems. An example is sorting, where the problem of sorting a set of elements can be decomposed into several subproblems, each involving the sorting of a disjoint subset of the original set, and a sorted list for the original set is obtained by combining the sorted lists for those subsets.

The *Divide-and-Conquer* method is a natural approach in parallel computing [11, 25, 37]. It is especially desirable in data intensive applications, e.g., in database systems and information retrieval systems. In *LDL* [28], which is a programming language for deductive databases, the *divide-and-conquer* method has been recognized as an important construct for potential parallel processing. In [17], it is pointed out that parallelism offers much better opportunities for speedup and scaleup than pipelined parallelism, and it

is recommended to use the divide-and-conquer method for exploiting partitioned parallelism. In [30], a new data model SVP is presented to capture parallelism in bulk data processing, both pipelined parallelism and partitioned parallelism, and divide-and-conquer mappings are formalized in the form of transducers. However, two issues have not been covered in these articles: one is the characterization of problems that can be parallelized by the divide-and-conquer method; another is the performance analysis for such parallel processing.

We focus on the two issues and apply the method to the fuzzy join problem. A linear recursive program is used to formalize problems to be parallelized. This program represents operations on sets and can be shown to include many problems in the database area and in other disciplines. We present a PArallel Divide-And-Conquer Scheme, or PADAC Scheme in short, for the linear recursive program. By proving a necessary and sufficient condition, we obtain a precise characterization for a problem to be parallelized by the divide-andconquer method.

We analyze the performance of the PADAC Scheme. Linear speedup is the best a parallel algorithm can achieve when compared to a fastest sequential algorithm. We give a necessary and sufficient condition when the PADAC Scheme can achieve linear speedup. The result is a little surprising. Assume that a problem has time complexity $O(n^{c_1} \log^{c_2} n)$. Then linear speedup is possible only when $c_1 = 1$. For a problem of higher order complexity, i.e., when $c_1 > 1$, linear speedup cannot be achieved.

A generalization of the linear recursive program and the corresponding PADAC Scheme is also obtained. In this way, the divide-and-conquer method is applied to operations on sets with parameters, and data duplication is allowed among different processors. Some examples have shown the advantage of the generalized PADAC Scheme. We will discuss the application of this generalized PADAC Scheme to the fuzzy join problem: linear speedup is demonstrated.

The rest of the paper is organized as follows. The PADAC Scheme is formally defined in Section 2, and the characterization is discussed in Section 3. The performance of the PADAC Scheme is analyzed in Section 4. In Section 5, the generalization of the PADAC Scheme is discussed. In Section 6, we apply the generalized PADAC Scheme to solve the fuzzy join problem.

2. The PADAC scheme

2.1. A linear recursive program

We assume that a given problem can be considered as an operation on an input set S to produce an answer W. Operations on sets are very common in database systems and other disciplines. For example, every database query language has some aggregate functions, which produce some values from a large number of tuples [27, 35]. Many other problems such as computing transitive closure and processing a join can also be considered as operations on sets. In logic programming, a predicate can be interpreted as a procedure [26]. In LDL [28], many operations on sets such as sorting are defined by logic programs.

THE DIVIDE-AND-CONQUER METHOD

 r_2

We assume that a given operation on sets can be evaluated by the following linear recursive program.

Program 1 $r_1 = p(\{x\}, W) : - sgl(x, W).$ p(S, W) : - sel(x, S),dif(S, x, SS), p(SS, WS),sgl(x, u),mrg(u, WS, W).

The predicate sgl(x, u) represents a procedure that produces a result u from a single element x (sgl stands for single); the predicate sel(x, S) represents a procedure that selects an element x from a given set S (sel stands for select); the predicate dif(S, x, SS) represents a procedure that computes the subset SS of S by removing the element x from S (dif stands for difference); the predicate mrg(u, WS, W) represents a procedure that merges u and WS to produce W(mrg stands for merge). In the following, we will use predicate and procedure interchangeably.

The predicate p(S, W) represents the operation on sets to be evaluated, where S and W represent the input set and the answer, respectively. Rule r_1 says that when the input set is singleton, i.e., when $S = \{x\}$, the answer W is obtained by sgl(x, W). Rule r_2 says that, in the general case, an element x is chosen from the input set S by sel(x, S) a result u is produced from the selected element x by sgl(x, u), the operation p is carried out recursively on the remaining subset SS to produce a partial result WS by p(SS, WS), and the final answer W is produced from u and WS through mrg(u, WS, W).

We point out here that Program 1 is actually a program scheme. For an operation on sets, after the specifications of those predicates are determined and the corresponding codes are written, Program 1 becomes a concrete program to evaluate the operation. Our intention is to construct a systematic approach for evaluating operations on sets in parallel using the Divide-and-Conquer Method. This would be especially beneficial for ordinary programmers who are accustomed to sequential programs but would like to have their programs parallelized.

Many operations involving sets can be defined in the form of Program 1. A query to a database can also be considered as an operation on a set (all tuples in the database involved). We will discuss two examples in Section 3.2: sorting and computing the transitive closure. More examples can be found in [40] including computing the similarity values between documents in information retrieval [32], the same generation problem, and the UP-FLAT-DOWN problem in deductive database [5, 23, 35].

To describe the execution semantics of Program 1, we introduce the following notations. Let D and R be two domains. We assume that the *input set* S is a finite subset of D, and the answer W is an element of R. The elements of D and R can be either simple values, e.g., numbers, or complex values, e.g., tuples, records, and even sets. For example, in sorting a set of records (with respect to a numerical field), the domain D is the set of all records of the record type, an input set S is a finite set of records, an answer W is a sorted list

406

of records in S, and the domain R is the set of all sorted lists of records. The predicate sgl(x, u) represents a mapping B which transforms a single element x in the input domain D to u = B(x) in the output range R; the predicate mrg(u, WS, W) represents a merging function f which combines the partial results u and WS to form the answer W, and is defined as from $R \times R$ to R.

During an execution of Program 1, rule r_2 is called recursively. Each time it is called, an element x is chosen by the predicate sel(x, S). Let x_i be the element chosen for the *i*th call, and x_n the last element left. When only x_n is left, rule r_1 is applied, and an element $u_n \in R$ is produced by the predicate sgl(x, u). Then, u_n is returned to the last call to r_2 . An element u_{n-1} is produced from x_{n-1} , $f(u_{n-1}, u_n)$ is computed by mrg(u, WS, W) and returned to the previous call to rule r_2 . This process continues until the final answer is computed.

A list $VL = [x_1, x_2, ..., x_n]$ with all elements taken from domain *D* is a valid list for the set $\{x_i : 1 \le i \le n\}$ with respect to the predicate sel(x, S) if $sel(x_i, S_i)$ is true for each *i*, where $S_i = \{x_k : i \le k \le n\}$. The predicate sel(x, S) is used only when *S* has two or more elements. For convenience, we assume that $sel(x, \{x\})$ is always true for any *x*. Conceptually, during an execution of Program 1, a valid list is generated first from the input set through the predicate sel(x, S).

Let L be a list with all elements taken from domain R and f a binary function on R. We define **f-back(L)** with respect to f as the result by applying f backward on L. That is

f-back $([u_1]) = u_1$, and f-back $([u_1, u_2, ..., u_n]) = f(u_1, f$ -back $([u_2, ..., u_n]))$, when n > 1.

For example, f-back $([u_1, u_2, u_3, u_4]) = f(u_1, f(u_2, f(u_3, u_4)))$. Assume that, in an execution of Program 1, a valid list $VL = [x_1, x_2, \ldots, x_n]$ is generated from S. Let $u_i = B(x_i)$ for each i, and $VLB = [u_1, u_2, \ldots, u_n]$. The element u_n is returned from rule r_i ; from the last call to the recursive rule r_2 , $f(u_{n-1}, u_n) = f$ -back $([u_{n-1}, u_n])$ is returned; and so on. The final answer W is f-back(VLB). The execution of the linear recursive program is sequential, and the execution semantics can be described as follows:

A valid list VL = [x₁, x₂,..., x_n] is generated from the input set S through the predicate sel(x, S);
VL is converted into VLB = [u₁, u₂,..., u_n] by the mapping B;

The final answer W = f-back(VLB).

2.2. The PADAC scheme

Our PADAC Scheme is developed to evaluate Program 1 on a shared-nothing architecture with K processors for some constant $K \ge 2$. That is, each processor has its own main memory and secondary memory, and communication is accomplished in some way, say, by a bus or via a network. The number of processors or sites, K, is usually limited in practice, especially in comparison with the input data size. Most research that is devoted to developing parallel methods in database environment is carried out under such an assumption [1, 9, 13, 17, 21, 34, 38, 39].

THE DIVIDE-AND-CONQUER METHOD

The PADAC Scheme has three phases: dividing phase, processing phase, and merging phase. In the dividing phase, we require that *S* be partitioned into *K* non-empty subsets *S_i*. So, no element appears in multiple subsets. We assume that some rule has been specified such that for any set $S \subseteq D$ a *partial order* \prec can be established on *S*. The partial order \prec is required to be defined on the given set *S*, not necessarily on the entire domain *D*. For example, in sorting a set of numbers, the rule may be the comparison of numbers and the partial order is \prec , the arithmetic predicate *less than*. That is, $x_i \prec x_j$ if and only if $x_i < x_j$. On the other hand, in sorting a set of records with respect to a numerical field nf, the rule may be the comparison of the values of nf and $x_i \prec x_j$ if and only if $nf_i < nf_j$. In both cases, the partial order \prec can be defined on the entire input domain. In computing the transitive closure of a directed acyclic graph *G*, the rule may be specified according to reachability, and the partial order \prec can be defined as follows. For any pair of edges (x_1, y_1) and (x_2, y_2) , $(x_1, y_1) \prec (x_2, y_2)$ if and only if there is a path from y_1 to x_2 in the graph *G*. In this case, the partial order is associated with the input set, and not defined on the background domain.

409

An element x of S is a *minimal* (or *maximal*) element of S, if S has no other element y such that $y \prec x$ (or $x \prec y$); two different elements x and y are incomparable if neither $x \prec y$ nor $y \prec x$ is true; for any finite set S, there is a minimal (and a maximal) element; a subset $S_i \subset S$ is called an (proper) *initial segment* of S, if $S_i \neq \emptyset$, $S - S_i \neq \emptyset$, and for any $x \in S_i$, any $y \in S - S_i$, $y \prec x$ is not true. A partial order in which any two different elements are not ordered is called the *trivial partial order*. That is, any element is a minimal (and maximal) element, any two different elements are incomparable, and any proper subset is an initial segment.

The partition of the input set S in our PADAC Scheme is done according to the partial order \prec defined on S. The condition enforced is

(*) For any x ∈ S_i and any y ∈ S_j, y ≺ x is not true if i < j, i.e., no element from a later subset can be *smaller than* any element in any preceding subset.

Such a partition is called a **non-decreasing** partition. A *non-increasing* partition could be defined similarly. But, to simplify the discussion, we use *non-decreasing* partitions only.

We assume that all non-decreasing partitions for a given S may be generated when different partitions are possible. That is, we only require condition (*) be enforced, but do not specify a fixed way in partitioning the input set. In most cases, S should be partitioned equally such that each processor processes about the same amount of data.

In the processing phase, we require that the original Program 1 be evaluated at each site with one subset S_i as input to produce a partial answer W_i . This is carried out independently without communicating with any other processors. So, the original problem is decomposed into K identical problems. In an implementation, any other program for the same problem can be employed to compute W_i at each site. That is, Program 1 is used only for the purpose of formalization and need not be used for execution.

In the merging phase, we require that the merging function f represented by mrg(u, WS, W) in Program 1 be applied to those W_i 's to form the final answer. The merging phase can also be carried out in parallel. There are different ways to apply f to those W_i 's. To address this issue clearly, we introduce the notation of f-terms in the following. For

a given list $[u_1, u_2, ..., u_n]$, a **sublist** is a list of the form $[u_i, u_{i+1}, u_{i+2}, ..., u_k]$, where $1 \le i \le k \le n$. That is, a sublist contains some consecutive elements of the original list. Assume that f is a function from $R \times R$ to R, and each u_i is in R. An **f-term on a list** is defined inductively (but informally in logic) as follows:

f-term $([u_1]) = u_1$, and f-term $([u_1, u_2, ..., u_n]) = f(f$ -term $([u_1, ..., u_k]), f$ -term $([u_{k+1}, ..., u_n])),$ where $1 \le k \le n$.

When n > 1, there are more than one way to form an *f*-term by choosing different values for k in the above expression. In any case, an *f*-term on a list represents an element in R. To compute an *f*-term on a list L with n > 1 elements, L is partitioned into two sublists L_1 and L_2 , and the result is

 $f-term(L) = f(f-term(L_1), f-term(L_2)).$

The *f*-back(*L*) defined in Section 2.1 is an *f*-term on *L* by dividing *L* in a special way. Although different *f*-terms on a list can be formed by dividing the list differently, the order of the elements in the list is not changed, and *f* is applied n - 1 times on a list of *n* elements. For example, $f(f(u_1, u_2), f(u_3, u_4))$ and $f(u_1, f(u_2, f(u_3, u_4)))$ are two different *f*-terms on the list $[u_1, u_2, u_3, u_4]$. To compute $f(u_1, f(u_2, f(u_3, u_4)))$, which is the *f*-back on the list, the function *f* is applied three times, and the computation is sequential. For $f(f(u_1, u_2), f(u_3, u_4))$, *f* is applied also three times, but the two *f*-terms $f(u_1, u_2)$ and $f(u_3, u_4)$ can be computed in parallel.

Different *f*-terms on the list $[W_1, W_2, \ldots, W_K]$ represent different ways to combine those partial answers. Assume K is a power of 2, one way to combine those W_i 's in parallel is to combine $f(W_i, W_{i+1})$ for all odd *i*'s first, and then combine those merged results in a similar way. However, we do not specify a chosen way to combine those W_i 's, and only require that any two *f*-terms on a list be the same. That is, the final answer W is an *f*-term on the list $[W_1, W_2, \ldots, W_K]$, but how *f* is applied to two adjacent *f*-terms (with the order of W_i 's in the list unchanged) will not affect the final answer. This implies parallel execution in the merging phase and gives some flexibility in implementation. Our PADAC Scheme is represented by the following Algorithm 1, which reflects all these criteria on a parallel evaluation of Program 1.

Algorithm 1

410

PHASE 1: S is partitioned non-decreasingly into K non-empty subsets S_i ;

PHASE 2: Each processor executes independently Program 1 with S_i as the input to produce W_i ;

PHASE 3: An *f*-term on the list $[W_1, W_2, \ldots, W_K]$ is computed as the final answer.

During an execution of Algorithm 1, S is non-decreasingly partitioned into K subsets S_i . Then, K processors are working in parallel: each processor evaluates Program 1, produces a valid list PL_i from S_i , converts it to another list PLB_i by applying the mapping B to all elements of PL_i and obtains $W_i = f$ -back(PLB_i). The final answer PW (the answer

THE DIVIDE-AND-CONQUER METHOD

generated in parallel) is an *f*-term on the list $[W_1, W_2, ..., W_K]$. Let *PL* (List generated in Parallel) be the concatenation of the *K* valid lists *PL_i* and *PLB* the concatenation of those *K* lists *PLB_i*. Then, *PW* is an *f*-term on *PLB*, since each $W_i = f$ -back(*PLB_i*) is an *f*-term on *PLB_i*. The execution semantics of the parallel Algorithm 1 can be described as follows:

The input S is non-decreasingly partitioned into K non-empty subsets S_i ;

A valid list PL_i is generated from S_i and converted into PLB_i for each *i*, or a list *PL*, which is the concatenation of these PL_i 's, is generated from *S* and converted into *PLB*, which is the concatenation of these PLB_i 's, in parallel;

The family of W_i 's, where $W_i = f - back(PLB_i)$, is computed in parallel;

The final answer PW is an f-term on $[W_1, W_2, \ldots, W_K]$, which is also computed in parallel.

3. A necessary and sufficient condition

In this section, by proving a necessary and sufficient condition, we will give a precise characterization for a problem to be parallelized by the *divide-and-conquer* method.

3.1. The characterization

We now define the concept that the parallel Algorithm 1 is computationally equivalent to the sequential Program 1. In practical situations, multiple answers are possible for an operation on a given input set. For example, an *EMPLOYEE* relation is to be sorted on the attribute *AGE*, and two tuples t_1 and t_2 represent two employees with the same age. After the relation is sorted, we may exchange the positions of the two tuples t_1 and t_2 and still get a valid answer. Let sgl(x, S) select a tuple with the smallest value on *AGE*. Then the predicate sgl(x, S) has a choice of multiple elements, e.g., when $S = \{t_1, t_2\}$. That is, multiple valid lists for one input set are possible. On the other hand, both the mapping *B* and the merging function *f* are usually well defined, i.e., $B(x_1) = B(x_2)$ when $x_1 = x_2$, and $f(u_1, u_2) = f(v_1, v_2)$ when $u_1 = v_1$ and $u_2 = v_2$. Then, an answer is determined by a valid list from *S*, i.e., one *VL* gives one answer, and multiple answers are possible only when multiple valid lists exist. Thus, different executions of the sequential method can produce nondeterministically a set of valid lists, and the corresponding set of answers. We require different executions of the parallel method to produce nondeterministically the same set of valid lists and the same set of answers as the sequential method.

Definition. For a finite subset S of D, let AVL(S) be the set of all possible valid lists from S generated by Program 1, and APL(S, K), the set of all possible *PL*'s from S generated by Algorithm 1 with $K \ge 2$ processors. Algorithm 1 is said to be **Computationally equivalent** to Program 1 if

(1) For any input S and any $K \ge 2$, AVL(S) = APL(S, K); and

(2) For any VL ∈ AVL(S), when it is generated by both Program 1 and Algorithm 1, Algorithm 1 gives the same answer as Program 1.

Recall that there is a partial order \prec defined on S; Also the input set S is very large and it is partitioned into K non-empty subsets non-decreasingly with respect to \prec in PHASE 1 of The PADAC Scheme, and all non-decreasing partitions for S may be generated in PHASE 1 of Algorithm 1.

Lemma 3.1. The following two statements are equivalent:

- 1. For any finite set $S \subset D$, and any $x \in S$, sel(x, S) is true if and only if x is a minimal element of S.
- 2. For any finite set $S \subset D$, |S| > 1, any $x \in S$, sel(x, S) is true if and only if there is an initial segment S_i of S such that $x \in S_i$, and sel (x, S_i) is true.

Proof: Statement $1 \Rightarrow$ Statement 2

412

Suppose sel(x, S) is true, and |S| > 1. We now prove there is an initial segment S_i of S such that $x \in S_i$ and $sel(x, S_i)$ is true. By Statement 1, x is a minimal element of S, i.e., for any element y of S, $y \prec x$ is not true. Then, $\{x\}$ is an initial segment of S containing x. Let S_i be any initial segment of S containing x. S_i is a subset of S, and x is a minimal element of S. So, x is a minimal element of S_i . By Statement 1, $sel(x, S_i)$ is true.

Suppose there is an initial segment S_i of S such that $x \in S_i$ and $sel(x, S_i)$ is true. We now prove that sel(x, S) is true. By Statement 1, x is a minimal element of S_i , i.e., for any element $y \in S_i$, $y \prec x$ is not true. For any $y \in S - S_i$, $y \prec x$ is not true either, since $x \in S_i$ and S_i is an initial segment of S. So, x is a minimal element of S, and, by Statement 1, sel(x, S) is true.

Statement 2 ⇒ Statement 1

Suppose sel(x, S) is true. We now prove by induction on *n*, the number of elements of *S*, that *x* is a minimal element of *S*. When n = 1, $S = \{x\}$, and *x* is a minimal element of *S*. When n > 1, by Statement 2, there is an initial segment S_i of *S* such that $x \in S_i$ and $sel(x, S_i)$ is true. By the definition of an initial segment, for any $y \in S - S_i$, $y \prec x$ is not true. Since $S - S_i \neq \emptyset$, we have $|S_i| < |S|$. By induction hypothesis, *x* is a minimal element of *S*.

Suppose x is a minimal element of S, we now prove sel(x, S) is true. Since $\{x\}$ is an initial segment of S, and $sel(x, \{x\})$ is true, sel(x, S) is true by Statement 2.

Lemma 3.2. The following two statements are equivalent:

1. For any finite set $S \subset D$, and any $x \in S$, sel(x, S) is true if and only if x is a minimal element of S.

2. AVL(S) = APL(S, K) for any S and any $K \ge 2$.

Proof: Statement $1 \Rightarrow$ Statement 2

Let S_i , i = 1, ..., K, be a non-decreasing partition of S, where each S_i has $n_i > 0$ elements. Suppose that $PL_i = [x_{i,1}, ..., x_{i,n_i}]$ is a valid list from S_i . Let PL be the concatenation of these PL_i 's, that is,

 $PL = [x_{1,1}, \ldots, x_{1,n_1}, x_{2,1}, \ldots, x_{2,n_2}, x_{K,1}, \ldots, x_{K,n_K}].$

Then, $PL \in APL(S, K)$. We now show that $PL \in AVL(S)$, and hence $APL(S, K) \subset AVL(S)$. For any $1 \le i \le K$, and any $1 \le j \le n_i$, let $S_{i,j} = \{x_{i,k} : j \le k \le n_i\}$. Since PL_K is a valid

THE DIVIDE-AND-CONQUER METHOD

list for S_K , $sel(x_{K,j}, S_{K,j})$ is true for all $j, 1 \le j \le n_K$. If $sel(x_{i,j}, S_{i,j} \cup (\bigcup_{i=i+1}^K S_i))$ is true for all $1 \le i < K$, and $1 \le j \le n_i$, then PL is a valid list for S and is in AVL(S). It is easy to show that $S_{i,j}$ is an initial segment of $S_{i,j} \cup (\bigcup_{i=i+1}^K S_i)$, since $S_i, i = 1, ..., K$, is a non-decreasing partition, and $S_{i,j}$ is a subset of S_i . PL_i is a valid list for S_i , so, $sel(x_{i,j}, S_{i,j})$ is true. By Statement 1 and Lemma 3.1, $sel(x_{i,j}, S_{i,j} \cup (\bigcup_{i=i+1}^K S_i))$ is true. That is, $PL \in AVL(S)$, and $APL(S, K) \subset AVL(S)$.

For any $VL \in AVL(S)$, we now show that $VL \in APL(S, K)$, and then $AVL(S) \subset APL(S, K)$. Let $PL = [x_1, x_2, ..., x_n]$ be a valid list in AVL(S). By the definition of valid lists and Statement 1, x_i is a minimal element of the set $\{x_j : i \leq j \leq n\}$. That is, $x_k \prec x_t$ is not true when k > t. VL can be partitioned into K non-empty sublists $VL_i, i = 1, ..., K$ (since S can be partitioned into K non-empty subsets). Let S_i be the set containing all elements of VL_i . Since $x_k \prec x_t$ is not true when $k > t, S_1, S_2, ..., S_K$ is a non-decreasing partition of S, and VL_i is a valid list for S_i by Statement 1. The concatenation of these VL_i 's is VL, i.e., VL can be generated by Algorithm 1 in parallel. So, $VL \in APL(S, K)$. Thus, $AVL(S) \subset APL(S, K)$, and hence AVL(S) = APL(S, K).

Statement 2 \Rightarrow Statement 1

We assume Statement 1 is false, and prove Statement 2 must be false too. By Lemma 3.1, we have the following cases.

Case 1. For some $S \subset D$, there is an element $x \in S$, such that sel(x, S) is true, but for any initial segment S_i containing x, $sel(x, S_i)$ is not true.

Suppose S_j , j = 1, ..., K is a non-decreasing partition of S. No matter $x \in S_1$ or not, $sel(x, S_1)$ is not true. That is, no valid list PL_1 of S_1 will have x as the first element, and no $PL \in APL(S, K)$ will have x as the first element. However, since sel(x, S) is true, there is a valid list in AVL(S) with x as the first element. Then, $APL(S, K) \not\subset AVL(S)$ and $APL(S, K) \neq AVL(S)$.

Case 2. For some $S \subset D$, there is an element $x \in S$, such that sel(x, S) is not true, but $sel(x, S_i)$ is true for an initial segment S_i of S.

Suppose S is partitioned into S_i and $S - S_i$ for K = 2. This is a non-decreasing partition of S. Since $sel(x, S_i)$ is true, there is a valid list for S_i has x as the first element, and there is a $PL \in APL(S, K)$ with x as the first element. However, sel(x, S) is not true, and no valid list in AVL(S) will have x as the first element. That is, $APL(S, K) \not\subset AVL(S)$ and $APL(S, K) \neq AVL(S)$.

Assume that $VL = [x_1, x_2, ..., x_n]$ is a valid list for S, and $u_i = B(x_i)$ for each i. Let

 $VLB_1 = [u_1, \dots, u_{k_1}], VLB_2 = [u_{k_1+1}, \dots, u_{k_2}], \text{ and } VLB_3 = [u_{k_2+1}, \dots, u_n],$

where $1 \le k_1 < k_2 < n$. Let f be a function from $R \times R$ to R, and $t_j = f$ -back(VLB_j), j = 1, 2, 3. The function f is called an **associative function w.r.t. valid lists**, if for any valid list VL, and any t_1 , t_2 and t_3 defined above, $f(f(t_1, t_2), t_3) = f(t_1, f(t_2, t_3))$. If f(f(u, v), w) = f(u, f(v, w)) for any u, v, w in R, f is called an **associative function on R**. It is clear from the definitions that an associative on R, but associative w.r.t. valid lists. A function which is not associative on R, but associative w.r.t. valid lists will be given in Section 3.2.2 when we discuss the transitive closure problem.

Recall that a list $VL = [x_1, x_2, \dots, x_n]$ is a valid list for a set $\{x_i : 1 \le i \le n\}$ if $sel(x_i S_i)$ is true for each i, where $S_i = \{x_j : i \le j \le n\}$. For a sublist $VL = [x_i, x_{i+1}, \dots, x_k]$, where $1 \le i \le k \le n$, there is a subset $\{x_j : i \le j \le k\}$. The sublist may or may not be a valid list for the subset depending on the predicate sel(x, S).

Lemma 3.3. Assume that any sublist of a valid list is a valid list for the corresponding subset. Let $VL = [x_1, x_2, ..., x_n]$ be a valid list, $u_i = B(x_i)$ for each i, and VLB = $[u_1, u_2, \ldots, u_n]$. If f is associative w.r.t. valid lists, then all f-terms on VLB are equal.

Proof: The proof is by induction on n, and we assume $n \ge 3$. When n = 3, VL = $[x_1, x_2, x_3], VLB = [u_1, u_2, u_3].$ There are only two *f*-terms on VLB, i.e., $T_1 = f(f(u_1, u_2), u_3)$. (u_1, u_2) and $T_2 = f(u_1, f(u_2, u_3))$. Let $t_j = f$ -back $(u_j) = u_j$, j = 1, 2, 3. Then, $T_1 = f(f(u_1, u_2), t_2)$, t_3) and $T_2 = f(t_1, f(t_2, t_3))$. Since f is associative w.r.t. valid lists, we have $T_1 = T_2$ by the definition.

Assume that the statement is true for any valid list with no more than n elements, where $n \geq 3$. Let $VL = [x_1, x_2, \dots, x_n, x_{n+1}]$ be a valid list of n+1 elements, and VLB = $[u_1, u_2, \dots, u_n, u_{n+1}]$, where $u_i = B(x_i)$. Suppose T is an *f*-term on VLB. We now prove T = f-back(VLB). By definition, $T = f(T_1, T_2)$, where T_1 is an f-term on $[u_1, \ldots, u_k]$, T_2 is an f-term on $VLB = [u_{k+1}, \ldots, u_{n+1}]$ and $1 \le k \le n$.

If k = 1, then $T_1 = u_1$, and T_2 is an *f*-term on the last *n* elements of VLB. Since a sublist of VL is still valid, $T_2 = f$ -back($[u_{k+1}, \ldots, u_{n+1}]$) by induction hypothesis. Then,

 $T = f(T_1, T_2) = f(u_1, f - back([u_{k+1}, \dots, u_{n+1}])) = f - back(VLB).$

If k > 1, then both T_1 and T_2 are *f*-terms on lists with no more than *n* elements. Since a sublist of VL is still a valid list, we have the following by induction hypothesis:

 $T_1 = f - back([u_1, \dots, u_k]) = f(u_1, f - back([u_2, \dots, u_k]));$

 $T_2 = f - back([u_{k+1}, \ldots, u_{n+1}]).$

Let $VLB_1 = [u_1], VLB_2 = [u_2, \dots, u_k], VLB_3 = [u_{k+1}, \dots, u_{n+1}], \text{ and } t_j = f\text{-back}$ (VLB_i) , j = 1, 2, 3. Then, $T_1 = f(t_1, t_2)$ and $T_2 = t_3$. Since f is associative w.r.t. valid lists, and any sublist of a valid list is still valid, we have the following by induction hypothesis:

$$T = f(T_1, T_2) = f(f(t_1, t_2), t_3) = f(t_1, f(t_2, t_3))$$

= $f(u_1, f\text{-back}([u_2, \dots, u_{n+1}])) = f\text{-back}(VLB).$

Theorem 3.1. Algorithm 1 is computationally equivalent to Program 1 if and only if **P1.** There exists a partial order \prec on S for any finite subset S of D such that for any $x \in S$, sel(x, S) is true if and only if x is a minimal element of S with respect to \prec .

P2. The merging function f is associative w.r.t. valid lists.

Proof: SUFFICIENT.

From P1 and Lemma 3.2, for a given S, we have AVL(S) = APL(S, K) for any $K \ge 2$. Suppose that $VL \in AVL(S)$, and generated by both Program 1 and Algorithm 1. Let the corresponding answer be W and PW respectively. We now prove that W = PW. It is easy to

verify that when P1 holds, any sublist of a valid list is also a valid list for the corresponding

subset. Let VLB be the list obtained from VL by applying B to all elements of VL. Then, as mentioned before, W = f-back(VLB), and PW is an f-term on VLB. Since VL is valid, from P2 and Lemma 3.3, PW = W.

NECESSITY.

Case 1. P1 is not true.

From Lemma 3.2, when P1 is not true, $AVL(S) \neq APL(S, K)$ for some $K \ge 2$, and then Algorithm 1 is not computationally equivalent to Program 1.

Case 2. P1 is true, but P2 is false.

THE DIVIDE-AND-CONOUER METHOD

Since P1 is true, we have AVL(S) = APL(S, K) for any $K \ge 2$. Let $VL = [x_1, x_2, \dots, x_n]$ be a valid list, and partitioned into $VL_1 = [x_1, \ldots, x_{k_1}], VL_2 = [u_{k_1+1}, \ldots, u_{k_2}], VL_3 =$ $[u_{k_2+1}, \ldots, u_n]$, where $1 \le k_1 < k_2 < n$. Let $VLB_1 = [u_1, \ldots, u_{k_1}]$, $VLB_2 = [u_{k_1+1}, \ldots, u_{k_1}]$ u_{k_2}], $VLB_3 = [u_{k_2+1}, \dots, u_n]$, where $u_1 = B(x_i)$. Let $t_j = f$ -back (VLB_j) , j = 1, 2, 3. Assume $f(t_1, f(t_2, t_3)) \neq f(f(t_1, t_2), t_3)$. Since VL is a valid list, by P1, x_i is a minimal element of the set $\{x_j : i \leq j \leq n\}$. So, $x_j \prec x_i$ is not true if j > i. Let S_j be the set containing all elements of VL_j , j = 1, 2, 3. Then, S_1 , S_2 and S_3 is a non-decreasing partition, and VL_j is a valid list from S_j . Thus, for the set S containing all elements in VL, the partition S_1 , S_2 and S_3 may be produced by Algorithm 1 for K = 3. When VL is generated this way, PW will be either $f(t_1, f(t_2, t_3))$ or $f(f(t_1, t_2), t_3)$. But the two f-terms are not equal. That is, there is an execution of Algorithm 1, the list VL is generated, but the answer PW is not the same as f-back(VLB), the answer given by Program 1 when VL is generated. Thus, Algorithm 1 is not computationally equivalent to Program 1.

Corollary 3.1. Algorithm 1 is computationally equivalent to Program 1 if P1 and

P2'. The merging function f is associative on R.

Proof: This is because an associative function is associative w.r.t. valid lists.

The property P1 guarantees AVL(S) = APL(S, K), while P2 or P2' ensures the same answer from a valid list. In P1, the statement that x is a minimal element can be replaced by that x is a maximal element. Then, the parallel algorithm will be modified accordingly. For the sake of simplicity, we use minimal only.

3.2. Examples

Many examples are given in [40] to illustrate the generality of the PADAC Scheme. Among them are computing the similarity values between a given query and a set of documents that have terms in common with the given query in information retrieval [32], sorting, computing the transitive closure, the same generation problem and the UP-FLAT-DOWN problem in deductive database [5, 23, 35]. In this subsection, we discuss how to apply the PADAC Scheme to parallelize the sorting problem and the transitive closure problem. The performance issue will be discussed in the next section.

414

415

3.2.1. Sorting. We consider sorting based on compare-exchange operations. Assume a set of numbers are to be sorted into non-decreasing order. The sorting problem can be formalized into Program 1 in two ways by defining the predicate sel(x, S) and hence the predicate mrg(u, WS, W) in different ways. In both cases, the predicate sgl(x, u) maps an element x into a singleton list u = [x].

Using the trivial order. When the predicate sel(x, S) arbitrarily picks an element from S, the predicate mrg(u, WS, W) represents a function which combines two sorted lists into one. In the sequential execution, u = [x], and the merging function just inserts x into a proper position in WS to get W. When those predicates are defined as above, Program 1 completes the sorting task and represents the sequential algorithm of sorting by insertion. The partial order \prec can be defined as the trivial partial order, i.e., any two different elements are not ordered and any element is a minimal element. So, the predicate sel(x, S) selects a minimal element x from S. It is clear that the merging function is associative. By Corollary 3.1, the sorting problem can be solved by the PADAC Scheme. In the dividing phase, the input set is arbitrarily partitioned into K subset of about the same size, since the trivial partial order is used. After each processor finishes sorting a subset, these sorted lists are merged together to form a sorted list for the input set. This parallel algorithm for sorting is a parallel version of merge-sorting.

Using the partial order <. When the the predicate sel(x, S) selects the smallest number x from S, the predicate mrg(u, WS, W) represents the concatenation function, that is, it appends two lists into one. In this case, Program 1 represents the sequential algorithm of sorting by selecting. The partial order \prec is the arithmetic comparison predicate <. Then, the predicate sel(x, S) selects a minimal element x from S with respect to \prec . The merging function is clearly associative. Thus, by Corollary 3.1, the sorting problem can be solved by the PADAC Scheme using a different partial order. The dividing of the input set is not trivial any more. Suppose K - 1 different numbers x_k are chosen and sorted. Then the dividing of S can be done by comparing each element in S with these K - 1 chosen numbers by binary search. After each subset is sorted, the concatenation of these sorted subsets is the sorted set. This parallel algorithm for sorting is a parallel version of *Quicksorting*.

3.2.2. The transitive closure. A binary relation *E* represents a directed graph G = (V, E), where the set of vertices *V* contains all elements in *E* and the set of edges is the relation *E*. Let *A* be the transitive closure of *E* (or *G*). Then a tuple (x, y) is in *A* if and only if there is a path from *x* to *y* in *G* [35]. Computing the transitive closure of *G* clearly represents an operation from one set to produce another set. The base relation *E* is the input set and the relation *A* is the answer.

In the following we assume G is acyclic, i.e., G has no directed cycles. Some algorithms for computing the transitive closure of acyclic graphs have been proposed [2, 14, 22, 41].

Using an order based on the reachability. Assume G is acyclic, a partial order \prec can be defined on E as follows:

THE DIVIDE-AND-CONQUER METHOD

417

For any pair of edges (x_1, y_1) and (x_2, y_2) , $(x_1, y_1) \prec (x_2, y_2)$ if and only if there is a path from y_1 to x_2 including the case $y_1 = x_2$ (a path of length 0).

Then, an edge (x, y) is a minimal element iff x has no in-coming edges. We assume that sel((x, y), E) takes a minimal element from E. When $E = \{(x, y)\}$, the transitive closure of E, A, is the same as E, i.e., A = E. So, the predicate sgl((x, y), u) maps a single tuple (x, y) to a singleton set $\{(x, y)\}$. The merging function f_1 is defined as

$$f_1(A_1, A_2) = A_1 \cup A_2 \cup (A_1 \# A_2)$$

where # represents the composition operation (e.g., $\{(x, y)\}$ # $\{(y, z)\} = \{(x, z)\}$).

Let (x_0, y_0) be a minimal edge, and AS be the transitive closure of ES, the subset of E without (x_0, y_0) . For any tuple (x, y) in A, the transitive closure of E, there is a path from x to y in G. Let (x, z) be the first edge on the path. If $(x, z) \neq (x_0, y_0)$, i.e., either $x \neq x_0$ or $z \neq y_0$, then all edges in the path are in ES and (x, y) is in AS, since (x_0, y_0) is a minimal edge and x_0 has no in-coming edges. If $x = x_0$ and $z = y_0$, i.e., (x_0, y_0) is the first edge in the path, then either $y_0 = y$, or there is a path from y_0 to y with all edges in ES, and (y_0, y) is in AS. As a result. (x, y) is in $\{(x_0, y_0)\}$ #AS. Thus, the transitive closure of E can be computed as $A = f_1(\{(x_0, y_0)\}, AS\}$.

Then, Program 1 computes the transitive closure of E. The operation # is associative [35], and the set union and join clearly satisfy the distributive law, that is,

$$A_1 \# (A_2 \cup A_3) = (A_1 \# A_2) \cup (A_1 \# A_3)$$

and

 $(A_1 \cup A_2) # A_3 = (A_1 # A_3) \cup (A_2 # A_3).$

Now we show that f_1 is associative, i.e., for any three binary relations A_1 , A_2 and A_3 , $f_1(A_1, f_1(A_2, A_3)) = f_1(f_1(A_1, A_2), A_3)$.

```
\begin{split} f_1(A_1, f_1(A_2, A_3)) &= A_1 \cup f_1(A_2, A_3) \cup (A_1 \# f_1(A_2, A_3)) \\ &= A_1 \cup (A_2 \cup A_3 \cup (A_2 \# A_3)) \cup (A_1 \# (A_2 \cup A_3 \cup (A_2 \# A_3))) \\ &= A_1 \cup A_2 \cup A_3 \cup (A_2 \# A_3) \cup (A_1 \# A_2) \cup (A_1 \# A_3) \cup (A_1 \# A_2 \# A_3) \\ f_1(f_1(A_1, A_2), A_3) &= f_1(A_1, A_2) \cup A_3 \cup (f_1(A_1, A_2) \# A_3) \\ &= (A_1 \cup A_2 \cup (A_1 \# A_2)) \cup A_3 \cup ((A_1 \cup A_2 \cup (A_1 \# A_2)) \# A_3) \\ &= A_1 \cup A_2 \cup (A_1 \# A_2) \cup A_3 \cup (A_1 \# A_3) \cup (A_2 \# A_3) \cup (A_1 \# A_2 \# A_3) \\ &= f_1(A_1, f_1(A_2, A_3)) \end{split}
```

Since both P1 and P2' are satisfied, by Corollary 3.1, the transitive closure of E can be computed in parallel by Algorithm 1.

In PHASE 1, *E* is partitioned non-decreasingly with respect to \prec into E_i , i = 1, 2, ..., K. The in-degree $d_{in}(x)$ of node *x* is maintained for each *x*, and an edge (x, y) is a minimal

edge if and only if $d_{in}(x) = 0$. When such a minimal edge is selected, the value of $d_{in}(y)$ should be decremented by 1. As a result, all out-going edges from y become minimal edges if $d_{in}(y) = 0$ after the decrementing. Suppose K divides |E|. Then, the first |E|/K edges taken are in E_1 , the second |E|/K edges taken are in E_2 , and so on.

418

In PHASE 2, each processor evaluates independently Program 1 with one subset E_i as the input set to produce A_i , the transitive closure of E_i . (As mentioned before, any program for transitive closure can be executed.)

In PHASE 3, the function f_1 is applied to the list $[A_1, A_2, ..., A_K]$ to compute A. Notice that f_1 will be applied K - 1 times and only one join is required each time. This is a benefit from the non-decreasing partition using the partial order \prec , since join is a very expensive, if not the most expensive, operation in database systems.

A merging function not associative but associative w.r.t. valid lists. If the trivial order is used, then any tuple in E is a minimal edge. The predicate sgl((x, y), u) represents the same mapping as mentioned earlier and maps a single tuple (x, y) to a singleton set $\{(x, y)\}$. Let an edge (x_0, y_0) be taken from E, and the transitive closure of the remaining subset be AS. As mentioned before, a tuple (x, y) is in the transitive closure if and only if there is a path from x to y in G. Assume $(x, y) \in A$, but it is not (x_0, y_0) and not in AS either. Then, (x, y_0) is in A iff there is a path from x to y in G containing edge (x_0, y_0) . Thus (x, y) is in $\{(x_0, y_0)\}$ #AS if (x_0, y_0) is the first edge on the path (so $x = x_0$); (x, y) is in AS# $\{(x_0, y_0)\}$ #AS if (x_0, y_0) is the last edge on the path. Let a merging function f_2 be defined as follows:

$f_2(A_1, A_2) = A_1 \cup A_2 \cup A_1 # A_2 \cup A_2 # A_1 \cup A_2 # A_1 # A_2.$

Then, A can be computed by $f_2(\{(x_0, y_0)\}, AS)$. But function f_2 is not associative w.r.t. valid lists. For example, let $E = \{(1, 2), (2, 3), (3, 4), (4, 5), (5, 6)\}$. The list $\{(1, 2), (4, 5), (2, 3), (5, 6), (3, 4)\}$ is a valid list with respect to the trivial order. Let $VL_1 = [(1, 2), (4, 5)]$, $VL_2 = [(2, 3), (5, 6)]$ and $VL_3 = [(3, 4)]$. Then, $VLB_1 = [\{(1, 2)\}, (\{4, 5)\}\}$, $VLB_2 = [\{(2, 3)\}, (5, 6)\}$ and $VLB_3 = [\{(3, 4)\}]$. Let A_i be the transitive closure for the subgraph represented by VL_i , i.e., $A_i = f$ -back(VLB_i) for i = 1, 2, 3. It can be verified that the tuples (2, 5) and (2, 6) are in $f_2(A_1, f_2(A_2, A_3))$, but not in $f_2(f_2(A_1, A_2), A_3)$. So, f_2 is not associative with respective to valid lists when the trivial order is used. By Theorem 3.1, the transitive closure can not be parallelized by the PADAC Scheme with f_2 and the trivial order.

Since f_2 is not associative with respective to valid lists when the trivial order is used, it is not an associative function. However, it is associative with respect to valid lists when the partial order \prec as defined earlier is used. Suppose that $VL = [e_1, e_2, \ldots, e_n]$ is a valid list for E according to \prec , where each e_i is an edge. Let VL be partitioned into two sublists VL_1 and VL_2 . Let E_i be the corresponding set of edges in VL_i , and A_i the transitive closure of E_i for i = 1, 2. Then E_1 and E_2 form a non-decreasing partition of E with respect to \prec , and there is no path in G with an edge in E_2 followed by an edge in E_1 . So, the expression $A_2#A_1 \cup A_2#A_1#A_2$ will produce nothing, and f_2 becomes f_1 which has been identified as an associative function with respect to valid lists when the partial order \prec is used.

THE DIVIDE-AND-CONQUER METHOD

4. Performance analysis

In this section we analyze the performance of the PADAC Scheme. In particular, we discuss the speedup attained by the PADAC Scheme. The speedup of a parallel algorithm is the ratio of the response time of a fastest sequential algorithm to that of the parallel algorithm. That is,

$$Speedup = \frac{RT_s(n)}{RT_p(n, K)}$$

where $RT_s(n)$ is the response time of a fastest sequential algorithm on an input of size *n* and $RT_p(n, K)$ is the response time of the parallel algorithm on *K* processors on an input of size *n*. (The subscripts *s* and *p* stand for *sequential* and *parallel* respectively.)

When Speedup is of the same order as K, that is, when Speedup = $C \times K$ for some constant $0 < C \le 1$, we say that the parallel algorithm achieves linear speedup [25]. Linear speedup is the best a parallel algorithm can achieve when compared to a fastest sequential algorithm. This is because that we can use a single processor to simulate a parallel algorithm on K processors and hence have a sequential algorithm of time $K \times RT_p(n, K)$. In other words, if a super-linear speedup is obtained, then we will have a sequential algorithm of time less than $RT_s(n)$. Notice that, $RT_s(n)$ should be the response time of a fastest sequential algorithm. It is often misleading and of limited value to compare a parallel algorithm to a specific sequential algorithm (instead of a fastest sequential algorithm). For more details, see, e.g., [25].

The divide-and-conquer method has been applied, either explicitly or implicitly, to parallelize many problems. Some experiments showed very slow or even zero speedup [31, 33], and no theoretical foundation has been given to explain the results. In this section, we will give a necessary and sufficient condition as to when the PADAC Scheme can achieve linear speedup. Our analysis shows that linear speedup is not possible unless the problem is of time complexity $O(n \log^c n)$ for a constant $c \ge 0$. Our theory explains well those previous experimental results.

4.1. Notations and assumptions

We discuss the time complexity and focus on the leading term of each expression to simplify the analysis. Some Greek letters are used to represent the orders of magnitude in such a way that, for example, n^{α} may represent $\Theta(n)$ and n^{γ} may represent $\Theta(n \log n)$. In such a case, γ represents a higher order of complexity than α , and this is denoted as $\alpha < \gamma$. Some conventional notations are used in the analysis:

 $g_1(n) = O(g_2(n))$ means that $g_1(n)$ is bounded above by $g_2(n)$, i.e., there are constants c and n_0 such that $g_1(n) \le cg_2(n)$ for all $n \ge n_0$;

 $g_1(n) = \Theta(g_2(n))$ means $g_1(n)$ and $g_2(n)$ are of the same order of magnitude, i.e., both $g_1(n) = O(g_2(n))$ and $g_2(n) = O(g_1(n))$;

 $g_1(n) = o(g_2(n))$ means that $g_1(n)$ is of lower order of magnitude than that of $g_2(n)$, i.e., for any constant c > 0, there is a constant n_0 such that $g_1(n) < cg_2(n)$ for all $n \ge n_0$.

We assume a shared-nothing architecture. Each site has its own main memory and secondary memory (disks). The input size *n* can be very large, while the number of processors *K* is limited. So, it is assumed that $n \gg K$. The main memory at a site is not large enough to keep all data needed, even for a subproblem with a smaller input size after partitioning. Thus, *IO* time becomes the dominating factor in the response time and it is chosen as the performance metric.

The output (or the answer) W may be a single element or a set. When W is a set, we assume that the output size is a function of the input size, i.e., $|W| = A|S|^{\alpha}$ for some constant A > 0 and some order of magnitude α . When W is a single element, |W| = 1, i.e., A = 1 and $\alpha = 0$. Then, in any case, we have $|W| = A|S|^{\alpha}$.

To reach workload balance, the input set *S* should be partitioned into *K* subsets of the same size $\frac{n}{K}$. But, in practice, this is not guaranteed, or the partitioning procedure requires more time than we want to spend. So, we assume that any subset obtained in the dividing phase has size between $\frac{1}{2}\frac{n}{K}$ and $\frac{n}{K}$ for some constant $\lambda > 1$.

We assume that the merging phase can be carried out in log K rounds in parallel according to a binary tree. The merging function f takes two inputs, say W^1 and W^2 , to produce an output denoted $W^{1,2}$, and the time complexity T_f of the merging function should be a function of $|W^1|$ and $|W^2|$. In our analysis, each W^j , j = 1, 2, is an intermediate result produced from an original input subset S^j of size n_j and, by our assumption, has a size An_j^α , and $W^{1,2}$ is the result produced from the original input subset $S_1 \cup S_2$ of size $n = n_1 + n_2$. Then the time complexity of the merge function is a function of the sizes of the two original subsets, n_1 and n_2 , i.e.,

$T_{f}(|W^{1}|, |W^{2}|) = g_{1}(|W^{1}|, |W^{2}|) = g_{1}(An_{1}^{\alpha}, An_{2}^{\alpha}) = g_{2}(n_{1}, n_{2}).$

As indicated above, each subset obtained in the dividing phase has a size between $\frac{1}{k}\frac{n}{K}$ and $\lambda \frac{n}{K}$ for some constant $\lambda > 1$. As a consequence, the difference among n_1, n_2 and n is up to a constant factor, and T_f can be expressed as a function of n after replacing both n_1 and n_2 by n, i.e.,

 $T_{f}(|W^{1}|, |W^{2}|) = T_{f}(n).$

So we assume that the complexity of the merging function is a function of the original input size. Let $T_f(n) = \Theta(n^{\beta})$ for some order of magnitude β . Then, $\beta \ge \alpha$, since the output of the merge function has a size An_i^{α} .

The parallel execution need not be synchronized. To simplify the analysis, we assume the execution is synchronized, i.e., after the partitioning is done, all processors begin local processing at the same time, and each round of the merging phase starts after the previous round is finished at all involved sites. To compute the response time of the parallel scheme, we will take the maximum time for each phase and each round at all involved sites. This will worsen the response time of the parallel scheme and serves as a lower bound for our computed speedup.

We have chosen I/O cost as the performance metric, because I/O operations are much slower than main memory CPU operations. In parallel execution, communication across different processors is required. To send a package X from one site to another costs at least

THE DIVIDE-AND-CONQUER METHOD

as much as to read X at the receiving site. Thus, the communication cost is not negligible and should be part of the response time of a parallel execution. The time needed to transfer a package between two sites is assumed to be proportional to the size of the package, that is, it takes time $C_{cm}|X|$, where C_{cm} is a constant, to send a package X between two sites.

All assumptions we are making in this section are

- * The input size is much larger than the number of available processors, i.e., $n \gg K$.
- * The output size is a function of the input size, and $|W| = A|S|^{\alpha}$. * Each subset obtained in the dividing law of the set of
- * Each subset obtained in the dividing phase of the parallel scheme has a size between $\frac{1}{\lambda} \frac{n}{K}$ and $\frac{\lambda}{R}$ for some constant λ .
- * The merging phase takes log K rounds, and the execution is synchronized.
- * The time complexity of the merging function is a function of the original input size, i.e., $T_f = T_f(n)$.
- * The communication cost is proportional to the size of the package transferred.

Some notations used in the analysis are listed in the following.

n: the input size, i.e., |S| = n;

- K: the number of available processors;
- λ : the skew constant, i.e., each subset obtained in the dividing phase has a size between $\frac{1}{\lambda} \frac{\pi}{K}$ and $\lambda \frac{\pi}{K}$, and $\lambda > 1$;
- α : the order of magnitude of the output size in the original input size, i.e., $|W| = A|S|^{\alpha}$;
- β: the order of magnitude of the merging function in the original input size, i.e., $T_f(n) = \Theta(n^{\beta})$;
- y: the order of magnitude of the time complexity of a fastest sequential algorithm (see the next sub-section).

4.2. The speedup of the PADAC scheme

Let the time complexity of a fastest sequential algorithm be $RT_{\epsilon}(n) = \Theta(n^{\epsilon_1} n)$ log^{*c*₁} *n*) for some constants c_1 and c_2 . As mentioned earlier, the I/O time is the dominating factor in the response time when processing large amount of data and it is chosen as the performance metric in our analysis. It is clear that $\gamma \ge 1$, since the entire input set has to be read into the main memory at least once. This implies $c_1 \ge 1$. For the parallel PADAC Scheme, the communication cost should be included in addition to the I/O cost at each site. Let Div(n, K) be the time for the dividing phase including the I/O time in partitioning the input set and the communication time in sending the partitioned input subsets to their destination sites, Proc(n, K) the I/O time for the local processing phase, and Merge(n, K) the time for the merging phase including the communication time to send intermediate results across site and the I/O time at those merging sites. Then, the response time of the PADAC Scheme is

$$RT_p(n, K) = Div(n, K) + Proc(n, K) + Merge(n, K)$$

420

The merging phase of the PADAC Scheme consists of log K rounds, and the time for the merging phase depends on the communication cost and the cost for executing the merging function. As discussed earlier, the time complexity of the merging function is a function of the original input size, i.e., $T_f(n) = \Theta(n^\beta)$. The following lemma says that the time for the merging phase is of the same order of magnitude as the complexity of the merging function, that is, $Merge(n, K) = \Theta(T_f(n))$.

Lemma 4.1. When $\beta > 0$, the merging phase of the PADAC Scheme has the same time complexity as the merging function, that is, $Merge(n, K) = \Theta(T_f(n))$.

Proof: The merging phase consists of $\log K$ rounds. When the merging function is applied in the last round, the entire original input is involved. That is,

$$T_t(n) = O(Merge(n, K)).$$

In the following, we prove $Merge(n, K) = O(T_f(n))$. In each round, some intermediate results are sent out from sending sites, then the merging function f is applied to two intermediate results at each merging site to produce a larger intermediate result. So the time for the *i*th round consists of two parts, $CM_i(n, K)$, the time for communication, and $MG_i(n, K)$, the time for merging. The execution is assumed to be synchronized. As a consequence, CM_i should be taken as that needed to transfer a largest possible intermediate result from all sending sites, and MG_i as the maximum time needed to execute the merging function at all merging sites. Then,

$$Merge(n, K) = \sum_{i=1}^{\log K} \{CM_i(n, K) + MG_i(n, K)\}$$

In the local processing phase, one processor has a subset of size at most $\lambda \frac{\pi}{K}$ and produces an intermediate result of size at most $A(\lambda \frac{\pi}{K})^{\alpha}$. In the first round of the merging phase, each sending site sends out a package of size at most $A(\lambda \frac{\pi}{K})^{\alpha}$; at each merging site, the merging function takes two arguments and produces an intermediate result of size at most $A(\lambda \frac{\pi}{K} \times 2)^{\alpha}$, which is the output produced from an input subset of size at most $A(\lambda \frac{\pi}{K} \times 2)^{\alpha}$ general, in the *i*th round, each sending site sends a package of size at most $A(\lambda \frac{\pi}{K} \times 2^{i-1})^{\alpha}$; at a merging site, the merging function *f* is applied to produce an intermediate result for an original input subset of size at most $\lambda \frac{\pi}{K} \times 2^{i}$. Then the time for the merging function for the *i*th round is

$$MG_i(n, K) = O\left(T_f\left(\lambda \frac{n}{K} 2^i\right)\right) = O\left(\left(\lambda \frac{n}{K} \times 2^i\right)^{\beta}\right).$$

The communication time for the ith round is

$$CM_i(n, K) = O\left(A\left(\lambda \frac{n}{K} \times 2^{i-1}\right)^{\alpha}\right) = O(MG_i(n, K)),$$

THE DIVIDE-AND-CONQUER METHOD

423

since the communication cost is proportional to the size of the package sent out and $\beta \ge \alpha$. The total time for the merging phase is

$$\begin{split} Merge(n, K) &= \sum_{i=1}^{\log K} \{ CM_i(n, K) + MG_i(n, K) \} \\ &= \sum_{i=1}^{\log K} \{ O(MG_i(n, K)) + MG_i(n, K) \} = \sum_{i=1}^{\log K} O\left(\left(\lambda \frac{n}{K} \times 2^i \right)^{\beta} \right) \\ &= O\left(\left(\lambda \frac{n}{K} \right)^{\beta} \sum_{i=1}^{\log K} (2^{\beta})^i \right) = O\left(\lambda^{\beta} n^{\beta} \times \frac{1}{K^{\beta}} \frac{2^{\beta} (K^{\beta} - 1)}{(2^{\beta} - 1)} \right) \\ &= O\left(\lambda^{\beta} n^{\beta} \times \frac{2^{\beta}}{2^{\beta} - 1} \frac{K^{\beta} - 1}{K^{\beta}} \right) = O(n^{\beta}) = O(T_f(n)). \end{split}$$

Since $K \ge 2$, the factor $\frac{K^{\beta}-1}{K^{\beta}}$ is less than 1 and can be dropped from the above expressions. The magnitude order β is treated as a constant number in the above, but the result holds when it is of the form $n^{C_1} \log^{C_2} n$ for some constants $C_1 \ge 1$ and C_2 . The log K rounds do not contribute a higher order complexity, because the summation $\frac{1}{K^{\beta}} \sum_{i=1}^{\log K} (2^{\beta})^i$ is bounded by a constant $\frac{2^{\beta}}{2^{\beta}-1}$.

Unlike the merging phase, the time complexity for the dividing phase may involve both the input size *n* and the number of processors *K*. Let the time complexity of the dividing phase be $\Theta(n^{p_1}K^{p_2})$. We prove in the following lemma that if the dividing phase takes time more than $\Theta(\frac{RT_i(n)}{K})$, then the PADAC Scheme can not reach linear speedup. The value of *K* is assumed to be limited, and the value of *n* may increase to arbitrarily large. So, both n^{p_1} and n^p are the dominating factor in the corresponding expression. That is, if $p_1 < \gamma$, then $n^{p_1}K^{p_2} < C'(n^pK^{-1})$ for constant *C'* when *n* is large, and vice versa. We say that $n^{p_1}K^{p_2}$ is of a higher order of magnitude than n^pK^{-1} if $p_1 > \gamma$ or $p_1 = \gamma$ and $p_2 > -1$.

Lemma 4.2. The PADAC Scheme can not reach linear speedup if the dividing phase has time complexity of a higher order than $\Theta(\frac{RT_i(n)}{K})$.

Proof:

Speedup =
$$\frac{RT_s(n)}{RT_n(n,K)} < \frac{RT_s(n)}{Div(n,K)} = \frac{n^{\gamma}}{n^{\gamma_1}K^{\gamma_2}}$$

If $\gamma_1 > \gamma$, then the speedup will approach zero when *n* is increasing and it is impossible to reach linear speedup. Otherwise, $\gamma_1 = \gamma$ and $\gamma_2 > -1$. Then the speedup is

Speedup
$$< \frac{1}{K^{\gamma_2}} = K^{-\gamma_2} = o(K).$$

Since $-\gamma_2 < 1$, linear speedup can not be reached.

For example, let $RT_s(n) = \Theta(n \log n)$ and $Div(n, K) = \Theta(\frac{n \log n}{\log K})$. Since $\log K = o(K)$, we have $\frac{1}{K} = o(\frac{1}{\log K})$, and Div(n, K) has a higher order of magnitude than $\frac{RT_s(n)}{K}$. The speedup will be at most $\log K$.

$$Speedup = \frac{RT_s(n)}{RT_\rho(n,K)} < \frac{RT_s(n)}{Div(n,K)} = \frac{n\log n}{\frac{n\log n}{\log K}} = \log K.$$

The time complexity of a fastest sequential algorithm is $RT_s(n) = \Theta(n^{\nu}) = \Theta(n^{c_1})$ log^{c₂} n) for some constant $c_1 \ge 1$. The merging function in the PADAC Scheme has time complexity $T_f(n) = \Theta(n^{\beta})$. The following lemma uncovers the fact that if $c_1 > 1$ then there exists no efficient merge function, i.e., for any merge function f, $T_f(n)$ is of at least the same time complexity as $RT_s(n)$.

Lemma 4.3. Assume $RT_s(n) = \Theta(n^{c_1} \log^{c_2} n)$ and $c_1 > 1$. Then, for any merging function f, $T_f(n) = \Omega(RT_s(n))$.

Proof: The fact that $c_1 > 1$ implies the fastest sequential algorithm has higher than linear time complexity, i.e., $RT_s(n) = \Theta(n^{\gamma})$ and $\gamma > 1$. We prove that if there exists a merging function f of time complexity $T_f(n) = \Theta(n^{\beta}) = o(n^{\gamma})$, then there is a sequential algorithm of time complexity $o(RT_s(n))$, which is faster than the given fastest algorithm.

We apply the divide-and-conquer method to construct such a faster sequential algorithm. We first sort the input set S in a non-decreasing order, then compute a result u_i from each single element x_i in S, and finally apply the merging function f according to a binary tree to compute the output W for S (see figure 1).

Since the input size *n* is very large, external sorting is required. The I/O time becomes the dominating factor in sorting and the CPU processing time can be ignored. For a uniprocessor system, when the main memory size is moderate (contains at least as many pages as the square root of the total number of pages), the sorting can be completed in linear I/O cost (in two passes) by serial Fastsort [33]. However, as we will see later (Section 3.3), when *n* is extremely large (or the main memory is not large enough), the I/O cost will still be $O(n \log n)$. So, the sorting cost is of $\Theta(n \log n) = o(RT_s(n))$, since $RT_s(n) = \Theta(n^{\gamma})$ and $\gamma > 1$.



Figure 1. $r_1 \prec r_2$, $sml(r_1) \prec sml(r_2)$, but $lrg(r_2) \prec lrg(r_1)$.

THE DIVIDE-AND-CONQUER METHOD

The entire input set can not reside in the main memory. The pages of *S* are loaded into the main memory in the sorted order. The pages residing in the main memory at the same time form a fragment, and we assume that the computation of the partial result from each fragment can be carried out in the main memory. Each fragment corresponds to a subtree of the merging tree from the bottom level. Within such a subtree, the I/O cost of the merging function can be ignored, since no I/O cost is involved. In the following we assume that the cost of merging data within a fragment is the same as $\Theta((n_1 + n_2)^{\beta})$ and we will obtain an upper bound of the time complexity for the constructed algorithm.

Without loss of generality, assume $n = 2^L$ for some integer L. Then the binary merging tree has L + 1 levels. We label these levels as in figure 1. At level 0, a result u_i is produced from x_i for each element x_i in S. To produce u_i from x_i takes constant time, since only a single element x_i is involved. Then, it takes time O(n) to produce all single results u_i from x_i , including the I/O cost.

At level l > 0, there are $\frac{\pi}{2^{j}}$ internal nodes. At each such node, the merging function f takes two intermediate results to produce a larger intermediate result. Each input to the merging function represents the result for an original input subset of size 2^{l-1} , and the output of the merging function represents the result for an original input subset of size 2^{l} . Then, the cost of the merging function at a node of level l is $T_f(2^l) = \Theta((2^l)^{d})$. Then, the total cost for all internal nodes in the entire merging tree is

$$\sum_{l=1}^{L} \left(\frac{n}{2^{l}} \right) \times (2^{l})^{\beta} = n \sum_{l=1}^{L} (2^{\beta-1})^{l}.$$

n

When $\beta = 1$, the sum is $n \times L = n \log n = o(RT_s(n))$, since $RT_s(n) = \Theta(n^{c_1} \log^{c_2} n)$ and $c_1 > 1$. When $\beta \neq 1$, the sum is

$$\times \left(\frac{(2^{\beta-1})^{L+1}-1}{2^{\beta-1}-1}-1\right) = n \times \frac{2^{\beta-1} \times (2^L)^{\beta-1}-2^{\beta-1}}{2^{\beta-1}-1}$$
$$= n \times \frac{2^{\beta-1}n^{\beta-1}-2^{\beta-1}}{2^{\beta-1}-1}.$$

When $\beta > 1$, the sum is $\frac{2^{n-1}n^{d}-2^{n-1}n}{2^{d-1}n} = \Theta(n^{\beta}) = o(RT_s(n))$. When $\beta < 1$, the sum is $\frac{2^{n-1}n-2^{d-1}n^{d}}{1-2^{n-1}} = O(n) = o(RT_s(n))$, since $RT_s(n) = \Theta(n^{c_1} \log^{c_2} n)$ and $c_1 > 1$. In any case, the cost for merging all nodes in the binary tree is of $o(RT_s(n))$.

The total cost of the constructed algorithm is the sum of the cost for sorting, the cost for computing all u_i (the cost for level 0 in the merging tree) and the cost for merging all internal nodes in the tree. Since each part in the sum is of $o(RT_x(n))$, the constructed algorithm has time complexity $o(RT_x(n))$.

The proof for Lemma 4.3 looks similar to that for Lemma 4.1. But there are $\frac{n}{2^l}$ nodes at level *l* and the costs at all these nodes are taken in account. So the result can not be derived directly from Lemma 4.1.

Theorem 4.1. Assume the fastest sequential algorithm has time complexity $RT_s(n) = \Theta(n^{e_1} \log^{e_2} n)$ for some constants $c_1 \ge 1$ and c_2 . Then the PADAC Scheme achieves linear

424

speedup if and only if (1) $RT_s(n) = \Theta(n \log^{c_2} n)$, i.e., $c_1 = 1$;

(2) The dividing phase takes time at most $O(\frac{RT_{\epsilon}(n)}{\kappa})$, i.e., $Div(n, K) = O(\frac{n \log^2 n}{\kappa})$; and

(2) The arrange place lakes line at most $O(\frac{1}{K})$, i.e., $Dr(n,n) = O(\frac{1}{K})$, and (3) The merging function has time complexity of a lower order of magnitude than $RT_s(n)$,

i.e., $T_f(n) = o(n \log^{c_2} n)$.

Proof: SUFFICIENCY.

426

The merging function has time complexity $T_f(n) = \Theta(n^\beta)$. If $\beta = 0$ (e.g., the merging function is trivial such as list concatenation), then the total time for the merging phase will be log *K*. As mentioned earlier, *K* is assumed to be fixed and *n* can increase to very large. Thus, log $K = o(RT_s(n))$, since $RT_s(n) = \Theta(n^\gamma)$ and $\gamma \ge 1$. Otherwise, $\beta > 0$ and by Lemma 4.1, the merging phase takes time $\Theta(T_f(n))$; by condition 3, $T_f(n) = o(RT_s(n))$. In any case, we have $Merge(n, K) = o(RT_s(n))$.

Let $RT_s(n) = \Theta(Cn \log^{c_2} n)$ (i.e., the leading term has a coefficient *C*). Since *K* is assumed to be fixed while *n* is increasing, $\log (\lambda \frac{n}{K}) = \log n + \log \lambda - \log K = \Theta(\log n)$. Then the processing phase takes time

$$Proc(n, K) = \Theta\left(RT_s\left(\lambda \frac{n}{K}\right)\right) = \Theta\left(C\left(\frac{\lambda n}{K}\right)\log^{c_2}\left(\lambda \frac{n}{K}\right)\right) = \Theta\left(\frac{C\lambda}{K}n\log^{c_2}n\right)$$

Let $Div(n, K) = O(\frac{C_{atx}}{K}n \log^{c_2} n)$ (i.e., the leading term has a coefficient C_{div}). Thus, the response time of the PADAC Scheme is

$$\begin{aligned} RT_p(n, K) &= Div(n, K) + Proc(n, K) + Merge(n, K) \\ &= O\left(\frac{C_{div}}{K} n \log^{c_2} n\right) + \Theta\left(\frac{C\lambda}{K} n \log^{c_2} n\right) + o(n \log^{c_2} n) \\ &= \Theta\left(\frac{C_{div} + C\lambda}{K} n \log^{c_2} n\right). \end{aligned}$$

The PADAC Scheme reaches linear speedup, because

$$Speedup = \frac{RT_s(n)}{RT_n(n, K)} = \frac{Cn\log^{c_2}n}{\frac{C_{du}+C\lambda}{n\log^{c_2}n}} = \frac{C}{C_{div}+C\lambda} \times K.$$

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By Lemma 4.2, condition (2) is necessary. If condition (1) is not true, then $c_1 > 1$. By Lemma 4.3, $T_f(n) = \Omega(RT_s(n))$ and condition (3) is not true. The merging function has time complexity $T_f(n) = \Theta(n^\beta)$, and the sequential algorithm has time complexity $RT_s(n) = \Theta(n^\gamma)$, where $\gamma \ge 1$. If (3) does not hold, then $\beta \ge \gamma > 0$. By Lemma 4.1, $Merge(n, K) = \Omega(T_f(n))$. Let $Merge(n, K) = \Theta(C_{mg}n^\beta)$ and $RT_s(n) = \Theta(Cn^\gamma)$. Then, the speedup is

$$Speedup = \frac{RT_s(n)}{RT_p(n, K)} < \frac{RT_s(n)}{Merge(n, K)} = \frac{Cn^{\gamma}}{C_{mg}n^{\beta}} \le \frac{C}{C_{mg}}.$$

THE DIVIDE-AND-CONQUER METHOD

427

Thus the speedup is limited by a constant, since $\beta \geq \gamma$, and hence linear speedup can not be reached.

4.3. An example

We use the sorting problem to illustrate the performance of the PADAC Scheme and discuss the problem in single-input-and-single-output and multi-input-and-multi-output environments. The transitive closure problem will be discussed in Section 6 with the generalized PADAC Scheme.

It has been realized that the I/O cost is the dominating factor in the response time for external sorting. Some algorithms have been proposed to reduce the I/O cost in both sequential and parallel sorting e.g. [6, 7, 10, 16, 33]. Most of them assume that the main memory has a moderate size (contains at least as many pages as the square root of the total number of pages of the input set), and external sorting can be completed with linear I/O cost (in two passes). The requirement of a moderate main memory size is equivalent to that the input size *n* is large but not too large. As will be seen after this paragraph, when *n* is extremely large relative to the main memory size, the I/O cost for sorting will still be $O(n \log n)$. Our analysis shows that, under the assumption that sequential sorting can be done in linear I/O cost, linear speedup is not possible in the single-input-single-output environment. This agrees with previous results, e.g., in [16, 31, 33]. However, under the asymptotic assumption that sequential sorting takes $O(n \log n)$ I/O cost, linear speedup can be achieved in the single-input-single-output environment. It seems that this has not been observed before. In the multi-input-and-multi-output environments, linear speedup can be achieved under either assumptions.

Assume that the main memory can contain M pages of data. Then 2M pages of data on the average can be sorted by replacement selection (based on a tournament or heapsort algorithm [24]). Assume that one page contains P elements. Let $N = \frac{n}{P}$, i.e., the input set occupies N pages. Suppose that $R = \frac{N}{2M}$. Then the input set can be sorted by replacement selection into R runs. After all R runs have been sorted, a sorted set can be computed by combining these sorted runs. If there are at most M runs, i.e., $R \leq M$, then the first page of each run is loaded into the main memory and replacement selection can again be used to merge all the R runs. When the first page of a run becomes empty, the second page of that run is loaded in, and so on. The sorted set is written out one page at a time. In this way, the entire input set is scanned twice: one pass for sorting all R runs and one pages for merging all sorted runs. This is how serial Fastsort [33] sorts in linear I/O time.

When $M < R \le M^2$, the original R runs are divided into M larger runs, each of them has at most M original runs and can be sorted in two passes. After that, the M large runs can be merged together in the same way. The entire input set will be scanned three times. In general, if $M^{k-1} < R \le M^k$, the entire input set need to be scanned k times before it is completely sorted. The I/O cost is

$$\log_M R = n \log_M \frac{N}{2M} = n \log_M \frac{n}{2PM} = n (\log_M n - \log_M 2PM).$$

Both P and M are constants. Thus the I/O cost for sorting is still $O(n \log n)$ when n is extremely large.

Single-input-and-single-output environment. If the partial order < is used, then a nondecreasing partitioning will generate K subsets such that $x \le y$ if $x \in S_i$, $y \in S_j$ and i < j. We need K - 1 different numbers x_i ; then an element x of S is in subset S_i if and only if $x_{i-1} < x \le x_i$ (x_0 and x_K are the possible smallest and largest numbers respectively). We call these K - 1 numbers partitioning numbers. A set of partitioning numbers can be obtained easily if it is known that the elements uniformly distributed within a range. Otherwise, it can be computed by sampling (e.g., [16, 18, 19]), or pre-computation. In any case, the cost for computing these partitioning numbers can be ignored. After the K - 1partitioning numbers are determined and sorted, an element can be decided to belong to a unique subset S_i by binary search. The entire input set has to be scanned at the starting site, and the dividing phase takes linear I/O cost, i.e., $Div(n, K) = \Theta(n)$.

428

There are two cases. When the input set is extremely large, $RT_s(n) = O(n \log n)$. Then linear speedup will be achieved, since the merging function (concatenation) takes at most linear I/O time and Merge(n, K) = O(n). When the main memory is moderate (or that the input set is large but not too large), $RT_s(n) = O(n)$; by Lemma 4.2, linear speedup is impossible, since $Div(n, K) = \Theta(n)$.

If the trivial order is used, then the merging function has linear I/O cost, since the two sorted subsets to be merged have to be scanned to accomplish the merging. The dividing phase takes at most linear I/O cost. Assume that $RT_s(n) = O(n \log n)$, linear speedup will be achieved. On the other hand, when $RT_s(n) = O(n)$, linear speedup can not be achieved even with the dividing cost ignored, since the last step in merging takes linear I/O cost already. This is why some previous experiment results show very small or zero speedup [31, 33].

Multi-input-and-multi-output environment. We assume that the partial order *less than* is used. (If the trivial order is used, the final merging step has to be carried out in a single site and multi-output does not make any sense.) In this case, linear speedup will be achieved both when $RT_s(n) = \Theta(n)$ and when $RT_s(n) = \Theta(n)$.

The input set is initially distributed across all K sites, but not in a non-decreasing manner. Let S^j be the fragment of S at site j before sorting. Then, the elements of S^j are not sorted; for two elements x and y in two different sites, each of the three cases may be true: x < y, x = y or x > y. After sorting, the entire set is still distributed across all K sites, but in a non-decreasing manner. Let S_k be the fragment at site k after sorting. Then, all elements in S_k are sorted; for two different sites $k_1 \neq k_2$, $x \le y$ if $x \in S_{k_1}$, $y \in S_{k_2}$ and $k_1 < k_2$.

After the K - 1 partitioning numbers are decided, each site j can partition the initial fragment S^j into K subsets S_k^j according to these partitioning numbers, where $S_k^j = S^j \cap S_k = \{x: x \in S^j \text{ and } x_{k-1} < x \le x_k\}$, and send S_k^j to site k for each $k \ne j$. The fragment after partition at site k is $S_k = \bigcup_{1 \le j \le K} S_k^j$. After S_k is sorted at site k for each k, the entire input is sorted and resides at the K sites.

Assume that the fragment at each site has a size about $\frac{n}{K}$ both before and after the partitioning. Then, the dividing can be completed in time $O(\frac{R}{K})$ at a site, including the I/O cost and the communication cost. That is, $Div(n, K) = O(\frac{R}{K})$. The merging function is trivial (concatenation of two sorted sets) and the cost can be ignored, since the output is not required to be sent to a single site. That is, $T_f(n) = O(1)$. Then by Theorem 4.1, linear speedup will be achieved for either $RT_s(n) = O(n)$ or $RT_s(n) = O(n \log n)$.

THE DIVIDE-AND-CONQUER METHOD

5. A generalization of the PADAC scheme

Some operations on sets need parameters. In some cases, it is convenient to consider part of the input set as a parameter set. This allows data duplication among different sites. The following Program 2 is a generalization of the basic linear recursive Program 1 and uses a parameter set Z to produce an answer W from an input set S.

Program 2 $p(\{x\}, Z, W) : - sgl(x, Z, W).$ p(S, Z, W) : - sel(x, S), dif(S, x, SS), sgl(x, Z, u), prm(x, Z, ZS), p(SS, ZS, WS),mrg(u, WS, W).

The predicate prm(x, Z, ZS) was not present in Program 1, each other predicate in Program 2 was present in Program 1 and has a similar meaning as in Program 1. The new predicate prm(x, Z, ZS) is used to compute a parameter subset for the remaining input subset after an element is selected. When the input set is a singleton, a result is produced by the predicate sgl(x, Z, W). Otherwise, an element x is selected from S, and a result u is produced from x using the parameter set Z. To produce a result from the remaining subset SS = S - x, a parameter subset ZS need to be computed. This is carried out by the predicate prm(x, Z, ZS). After the result WS is produced from SS using ZS, the final result W is formed from u and WS by the predicate mrg(u, WS, W).

The following Algorithm 2 is our generalized PADAC Scheme which is developed to evaluate Program 2 in parallel based on the *divide-and-conquer* method. As before we assume that a partial order \prec is defined on a given input set S. Then, in the dividing phase (PHASE 1), the input set S is partitioned into K subsets non-decreasingly according to \prec . In addition, a parameter subset Z_k need to be computed for each subset S_k . In the processing phase (PHASE 2), a partial result W_k is produced from each subset S_k with the parameter subset Z_k . Finally, the partial results are combined to form the answer W corresponding to the entire input set S with the parameter set Z.

Algorithm 2

PHASE 1: The input set S is partitioned non-decreasingly into K non-empty subsets S_k ; the corresponding parameter sets Z_k are computed.

PHASE 2: Each processor executes independently Program 2 with S_k as the input set and Z_k as the parameter set to produce W_k .

PHASE 3: The final answer W is computed from these partial results W_1, W_2, \ldots, W_K by the merging function.

Algorithm 2 is similar to Algorithm 1 except that a parameter subset Z_k need to be computed for each input subset S_k in the dividing phase. The intention of non-decreasing



partition of the input set is that, for any input subset S_k , $1 < k \le K$, the elements in all earlier subsets, i.e., in $\bigcup_{1 \le l < k} S_l$, might be selected before any element in S_k (and in any later subset) is selected in the sequential linear program. Program 2. As a result, the parameter subset Z_k for S_k should be the same as that obtained in Program 2 after the elements in $\bigcup_{1 \le l < k} S_l$ have been applied one by one and the predicate prm(x, Z, ZS) has been applied accordingly. The elements in $\bigcup_{1 \le l < k} S_l$ may be selected in different orders by the predicate sel(x, S) in Program 2. To guarantee that a unique parameter subset Z_k is to be computed for each input subset S_k , we assume that, for any input subset S_l , if the elements in S_l can be selected consecutively in different orders by the predicate sel(x, S) the parameter subset for the remaining input subset $S - S_i$ is the same. When this is true, we say that the predicate prm(x, ZS, Z) is order-independent. Then, in the dividing phase of Algorithm 2, Z_k is to be computed as the parameter subset for $S - \bigcup_{1 \le l < k} S_l$. In the next section, we will see that in many applications it is rather simple and straightforward to compute these parameter subsets.

All concepts defined earlier for Program 1 and Algorithm 1 can be defined for Program 2 and Algorithm 2 in the same way or with minor changes, and we can prove similar results regarding the *computational equivalence* between the sequential Program 2 and the parallel Algorithm 2.

Theorem 5.1. Assume the predicate prm(x, ZS, Z) is order-independent. Algorithm 2 is computationally equivalent to Program 2 if and only if

P1. There exists a partial order \prec on any finite subset S of D such that for any $x \in S$, sel(x, S) is true if and only if x is a minimal element of S with respect to \prec .

P2. The function f is associative with respect to valid lists.

Corollary 5.1. Assume the predicate prm(x, ZS, Z) is order-independent. Algorithm 2 is computationally equivalent to Program 2 if

P1. and

P2'. The merging function f is associative on R.

6. Applications

In [41], we have shown that the transitive closure problem can be parallelized by the generalized PADAC Scheme. The set of vertices V is considered as the input set, while the set of edges E is treated as the parameter set. The partial order on V is a topological order (We discuss acyclic graphs only.) The predicate sgl(x, Z, u) produces the descendant set of vertex x (i.e., the set of all tuples (x, y) in the transitive closure with x at the first position), and requires significant computation, while the merge function is the set union and trievial. Our simulation results show that in sequential computation our algorithm is superior to other existing algorithms in most cases, and that in parallel computation linear speedup is achieved.

In the following, we apply the generalized PADAC Scheme to the fuzzy join problem and show that linear speedup will also be obtained in this application. From the two examples, we can see that more complicated problems can be parallelized by the generalized PADAC

THE DIVIDE-AND-CONQUER METHOD

Scheme. However, the introducing of parameter sets makes it much more difficult to analyze the performance of the scheme than in the case without parameter sets.

6.1. Fuzzy joins

In a fuzzy relation [12, 29, 20, 42–44], the value of an attribute of a tuple may represent either a single number or an interval. For example, a relation about people may have an attribute AGE. A tuple with a single number, say 28, as the value for the AGE attribute says the age of the person is 28. A tuple with "about 35" as the value for the AGE attribute says that the age of the person is, for instance, between 30 and 40, which represents an interval [30, 40]. Such a value "about 35" is called a fuzzy number. A tuple may have another kind of value such as "young" for the AGE attribute, which represents, for instance, an interval [21, 35]. Such a value "young" is called a fuzzy label. Both fuzzy number and fuzzy label represent existing but uncertain information. An attribute is called a *fuzzy attribute* if its domain contains single values as well as fuzzy numbers or/and fuzzy label; a relation is called a *fuzzy relation* if at least one attribute is a fuzzy attribute.

Consider the join $R \bowtie_{R,B=T,B} T$, where R and T are two fuzzy relations and B is a common fuzzy attribute. When B is the attribute AGE, the above join requests all pairs of persons, one from R and another from T, who have the same age. Two persons can not have the same age if one is 28 and the other is "about 35", since "about 35" means between 30 and 40. However two persons may have the same age if one is 28 and the other is "young", or one is "about 35" and the other is "young", since "young" means between 20 and 35. But we do not know for sure that they have the same age. So, unlike joins of ordinary relations, the condition r.B = t.B can only be checked with some uncertainty, since the join values may represent some uncertain information. We use degrees, which are numbers between 0 and 1, to indicate the uncertainty that the join condition r.B = t.B holds.

We give some notations to define fuzzy joins formally. In general, a value in the domain of a fuzzy attribute represents a (finite or infinite) subset of the background domain. For example, AGE = 28 represents a singleton set {28}, "about 35" represents a set of { $x : 30 \le x \le 40$ }. For a tuple r of R and a tuple t of T, we use r.B and t.B to denote either the fuzzy value or the corresponding subset, and $r.B \cap t.B$ represents the ordinary set intersection. Formally, the join $R \bowtie_{B=T,B} T$ is carried out as follows.

For any tuple r of R and any tuple t of T,

(1) When $r.B \cap t.B = \emptyset$, no tuple is generated from r and t.

(2) When r.B ∩ t.B ≠ Ø, a degree, which is a number between 0 and 1 is computed, and a tuple will be generated from r and t only when the degree is positive.

We do not discuss the issue in this paper how to compute the degree and assume that the degree can be computed in constant time. Interested readers are referenced to [12, 42, 44]. To compute the join $R \bowtie_{R,B=T,B} T$ by nested loops, we need to compare each tuple $r \in R$ with all tuples in T. There is analytical and experimental evidence that hashed join is the most effective method for joins of ordinary relations, but it is unlikely that the method can be applied to joins of fuzzy relations effectively, since a value may be an interval as

430

THE DIVIDE-AND-CONQUER METHOD

well as a single number. Sort-merge join is a good candidate for such joins. Suppose that both R and T are sorted according to an appropriate partial order based on the join values. Then, the join proceeds according to the partial order. It is possible that the join of the two fuzzy relations can be carried out in linear I/O cost after both relations are sorted according to the partial order (see performance in Section 5.4).

Fuzzy joins are similar to "band joins" (e.g. [15]). In "band joins", each value of the joining attribute is a simple value and represents an interval. However, all intervals for different joining values are of the same length. Fuzzy joins are much more general than "band joins", since each joining value may be an interval as well as a simple value and different intervals may have different lengths. The method for computing band joins can not be adopted directly for fuzzy joins, while our algorithm for fuzzy joins includes band joins as a special case.

6.2. Formalizing fuzzy joins by Program 2

Let the domain of the joining attribute *B* be *D*. A value *v* in *D* may be either a single number or an interval. When *v* is an interval, we use b(v) and e(v) to denote the two (beginning and ending) points, i.e., v = [b(v), e(v)]. When *v* is a single number, we let b(v) = e(v) = v. Then, each value corresponds to a pair of ordinary numbers. We define a linear order on *D* as follows.

Definition.

- 1. A linear order on D is defined as follows: for two values v_1 and v_2 in D, $v_1 \prec v_2$ if $b(v_1) < b(v_2)$, or $b(v_1) = b(v_2)$ and $e(v_1) < e(v_2)$. For example, $[2, 5] \prec [3, 4]$ and $[3, 4] \prec [3, 5]$.
- 2. A partial order on R (or T) is defined as follows: for two tuples r_1 and r_2 , $r_1 \prec r_2$ if $r_1.B \prec r_2.B$.

For two values v_1 and v_2 in D, we use $v_1 \leq v_2$ for $v_1 \prec v_2$ or $v_1 = v_2$. For two tuples r_1 and r_2 , we use $r_1 \leq r_2$ for $r_1 \prec r_2$ or $r_1 \cdot B = r_2 \cdot B$.

Definition.

- For any tuple r in R, sml(r) is the smallest value v of D that appears in T and r. B ∩ v ≠ Ø, and lrg(r) is the largest value v of D that appears in T and r. B ∩ v ≠ Ø.
- 4. The range of a tuple r of R is a subset of T defined by $Rng(r) = \{t : t \in T \text{ and } sml(r) \leq t.B \leq lrg(r)\}.$

For two tuples of R, $r_1 \prec r_2$, we have $sml(r_1) \preceq sml(r_2)$, but it is possible that $lrg(r_2) \prec lrg(r_1)$ (see figure 1). This is because that $r_1 \prec r_2$ implies $b(r_1) \leq b(r_2)$, but it does not imply $e(r_1) \leq e(r_2)$.

A tuple t of T can not join with a tuple r of R if $r.B \cap t.B = \emptyset$. By the definition of sml(r) and lrg(r), a tuple t of T can not join with a tuple r of R if t is not in Rng(r), i.e., either $t.B \prec sml(r)$ or $lrg(r) \prec t.B$. However, it is possible that t can not join with r even t is in Rng(r) (see figure 2).



Figure 2. $sml(r) \prec t.B \prec lrg(r)$, but $r.B \cap t.B = \emptyset$.

We now formalize the join of two fuzzy relations $R \bowtie_{R,B=T,B} T$ by Program 2. Relation R is the original input set, relation T is the original parameter set, and the final answer W is the set containing all tuples generated in the join. (Each tuple in W has a degree of satisfaction greater than zero.) The predicate sel(r, R) selects a minimul tuple r from R according to the partial order \prec on R. The predicate sgl(r, T, u) carries out the join between tuple r and all tuples of T in Rng(r) and can be written as sgl(r, Rng(r), u), and u is the set containing all tuples generated from the join between r and Rng(r). The remaining input subset RS is obtained by the predicate dif(R, r, RS) by removing from R the selected tuple r. The parameter set TS for RS is obtained by the predicate prm(r, T, TS) by removing from T those tuples that are not in Rng(r) and precede the tuples in Rng(r). These removed tuples will not be used for the join with any later tuples of R, since r is a minimal tuple of R and $sml(r) \leq sml(r')$ for any tuple r' in the remaining subset RS. The predicate mrg(u, WS, W) forms the final answer W by taking the union of u and WS. Then the following program computes the join of two fuzzy relations.

```
Program FuzzyJoin

p(\{r\}, T, W) : - sgl(r, Rng(r), W).

p(R, T, W) : - sel(r, R),

dif(R, r, RS),

sgl(r, Rng(r), u),

prm(r, T, TS),

p(RS, TS, WS),

mrg(u, WS, W).
```

It can be seen that Program FuzzyJoin performs the join of two fuzzy relations in a sort-merge manner.

6.3. Parallelizing fuzzy joins by the generalized PADAC scheme

A partial order \prec is defined on the input set *R*, the predicate *sel(r, R)* selects a minimal tuple from *R*, and the predicate *mrg(u, WS, W)* represents the set union which is apparently associative. By Theorem 5.1, the join of two fuzzy relations can be parallelized by the generalized PADAC Scheme.

We discuss the parallel scheme in a multi-input and multi-output environment. Initially, both R and T are distributed across the K sites, but not according to the partial order \prec .

432

The partition of R is carried out the same way as in sorting, and in processing phase each subset of R will be sorted. We discuss the dividing of T in the following. We use R^j (T^j) to denote the subset of R (T) at site j before re-distribution, and R_k (T_k) the subset of R (T) at site k after.

After K - 1 partitioning tuples for R are determined, R is non-decreasingly partitioned into K subsets R_k and a tuple r of R is in R_k if $x_{k-1} \prec r \preceq x_k$, $1 \le k \le K$, where x_0 and x_K are the possible minimal and maximal tuples of R respectively. For each tuple r in a subset R_k , the tuples of T in Rng(r) are needed to perform the join. The parameter set T_k for R_k will be $\bigcup_{r \in R_k} Rng(r)$. Let $b(k) = min\{b(r, B) : r \in R_k\}$ and $e(k) = max\{e(r, B) : r \in R_k\}$. Then each subset R_k of R determines an interval [b(k), e(k)] and the corresponding parameter subset $T_k = [t \in T : t.B \cap [b(k), e(k)] \neq \emptyset]$. We call these K intervals dividing intervals for T. Notice that the K intervals [b(k), e(k)] may intersect each other, and those subsets T_k may intersect too. Even the intervals [b(k), e(k)] do not intersect, there still may be some overlap between different subsets T_k .

The computation of the dividing intervals for T and the dividing of T can also be carried out in parallel at all sites. Let $R_k^j = R^j \cap R_k$, i.e., R_k^j is the set of tuples that will be sent from site j to site k. Let $b^j(k) = min\{b(r,B) : r \in R_k^j\}$ and $e^j(k) = max\{e(r,B) : r \in R_k^j\}$. At each site j, the values of $b^j(k)$ and $e^j(k)$ for all k are computed when R^j is partitioned into R_k^j , and they are sent to all other sites after the partitioning of R^j is completed. Then at each site k, the K dividing intervals [b(k), e(k)] for T are computed by $b(k) = min\{b'(k), e(k)\}$ are computed at all sites, because each site has the same $b^j(k)$ and $e^j(k)$. Let R_k^j be defined in the same way as for R_k^j . Then at site j, T^j is divided into K subsets T_k^j . At tuple t of T_j is in T_k^j if $t.B \cap [b(k), e(k)] \neq \emptyset$. The K dividing intervals can also be sorted according to the linear order \prec on the domain D, and a binary search can be employed on the K intervals. After a tuple is determined to belong to T_k^j , the preceding and following intervals should be examined since the tuple may be in multiple intervals.

We give a parallel algorithm derived from our PADAC Scheme in the multi-input-andmulti-output environment. Step (1) through (6) form the dividing phase of the PADAC Scheme, Step (7) is the processing phase of the PADAC Scheme, and the merging phase of the PADAC Scheme is not shown in the algorithm. Some Steps, say Step 2 and 3, can be interleaved.

Algorithm MIMO-fuzzy join

- (1) Decide K 1 partitioning tuples for R.
- (2) At each site j, partition R^{j} into K subsets R_{k}^{j} according to the partitioning tuples decided in (1); compute $b^{j}(k)$ and $e^{j}(k)$ for all k.
- (3) At each site j, send R_k^j to site k for each k; send all $b^j(k)$ and $e^j(k)$ to all sites $k \neq j$. (4) At each site j, compute the same K dividing intervals [b(k), (e(k))] for T by
- $b(k) = min\{b^j(k) : 1 \le j \le K\}$ and $e(k) = max\{e^j(k) : 1 \le j \le K\}$.
- (5) At each site j, divide T^{j} into K subsets T_{k}^{j} according to the K dividing intervals decided in (4).
- (6) At each site j, send T_k^j to site k for each k.

THE DIVIDE-AND-CONQUER METHOD

(7) At each site k, perform the sort-merge join between R_k and T_k , where $R_k = \bigcup_{1 \le j \le K} R_k^j$ and $T_k = \bigcup_{1 \le j \le K} T_k^j$.

6.4. Performance

After both *R* and *T* are sorted, the merge join can be completed in linear I/O cost (see next paragraph). If we assume that both *R* and *T* are very large and that sorting takes $O(n \log n)$ I/O cost, then the join of two fuzzy relations needs $O(n_1 \log n_1 + n_2 \log n_2)$ I/O cost, where $n_1 = |R|$ and $n_2 = |T|$. That is, the sorting cost is the dominating factor. Then, fuzzy join can be parallelized with linear speedup under this assumption, since sorting can be parallelized by the PADAC Scheme with linear speedup in this case (Section 3.3). In the following, we assume that the main memory has a moderate size, or both *R* and *T* are large but not too large. As explained earlier in Section 3.3, sorting takes linear I/O cost. Thus, the join of two fuzzy relations can be carried out in linear I/O time in sequential computation. That is, $RT_r = O(n_1 + n_2)$.

To perform the merge-join in linear I/O cost, the tuples of R are loaded into the main memory one page at a time in the sorted order. Let CR be the current page of R in the memory, and $b(CR) = min\{b(r,B) : r \in CR\}$ and $e(CR) = max\{e(r,B) : r \in CR\}$. The value of b(CR) is equal to b(r'.B), where r' is the first (minimal) tuple of CR; the value of e(CR) is obtained by scanning all tuples in CR and can be computed when R is being sorted. The tuples of T that may join with any tuples in CR are in $\bigcup_{r \in CR} Rng(r)$. A tuple t of T precedes the tuples in $\bigcup_{r \in CR} Rng(r)$ if e(t.B) < b(CR); it follows the tuples in $\bigcup_{r \in CR} Rng(r)$ if b(t,B) > e(CR). The pages of T are loaded into the main memory one by one in the sorted order. If a page of T contains only tuples that precede those tuples in $\bigcup_{r \in CR} Rng(r)$, then the page is useless in current as well as in later joining and should be discarded. This is because $sml(r_1) \leq sml(r_2)$ if $r_1 \leq r_2$. If a page contains some tuples in the set $\bigcup_{r \in CR} Rng(r)$, then the join is performed and the page stays in the main memory. since some tuples in the page may join with some tuples in the next page of R. If a page contains one tuple that follows the tuples in $\bigcup_{r \in CR} Rng(r)$, then no later pages of T are needed for the page CR of R and should not be loaded in. Then, the join w.r.t. the current page of R is completed, and several pages of T may reside in the main memory. For the next current page of R, the values of b(CR) and e(CR) are modified before the joining. The pages of T in the main memory are examined first, and the first few pages may be discarded immediately if they contain only tuples preceding the tuples in $\bigcup_{r \in CR} Rng(r)$. Some later pages of T may need to be loaded in. It is clear that the join can be completed in linear I/O cost as long as the main memory can keep one page of R and the maximal number of pages of T that cover $\bigcup_{r \in CR} Rng(r)$ for any single page of R.

We discuss parallel execution under assumptions similar to that in Section 3. We assume that each site k does not have enough main memory to keep the entire subset R_k or T_k , and the I/O cost is again the dominating factor for local processing at each site. The communication cost should be included. We assume that there is no contention in the network, e.g., there is a link between any pair of sites, and the communication cost at a site is proportional to the amount of data sent out from and received at that site. The cost in the merging phase is

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YANG FT AL

ignored, that is, the produced tuples are scattered across all sites and not required to send to a single site. For example, another operation, say another join, is to be performed in parallel. For ordinary (non-fuzzy) relations, many parallel joining algorithms are evaluated under a uniform distribution assumption, e.g., [34, 36]. The following are similar assumptions for fuzzy relations

- Both relations R and T are initially distributed evenly across the K sites.
- * A set of K 1 partitioning tuples that partitions R into K subsets of about the same size exists or can be decided efficiently
- The K dividing intervals [b(k), (e)] divide T into K subsets of about the same size.
- * The interval of each tuple of T is relative small w.r.t. the length of those dividing intervals [(k), e(k)] so that one tuple of T will appear in at most two subsets of T. This is usually true for fuzzy applications.

Under these assumptions, the cost of the dividing phase (Step 1 through 6 in Algorithm MIMO-FuzzyJoin) is $Div(n_1, n_2, K) = O(\frac{n_1+n_2}{K})$.

- Step (1): The set of partitioning tuples for R can be computed by sampling; the sampling cost is small and can be ignored, since the sample size is usually small. Alternatively, the set of partitioning tuples for R can be pre-computed and updated periodically. Thus, the cost of the pre-computation can be amortized over many queries.
- Step (2): the I/O cost is $O(\frac{n_1}{K})$ at a site;
- Step (3): the communication cost is $O(\frac{n_1}{K})$ at a site;
- Step (4): the computation of the K intervals does not incur extra I/O cost;
- Step (5): the I/O cost is $O(\frac{n_2}{\nu})$ at a site;
- Step (6): the communication cost is $O(\frac{n_2}{K})$ at a site.

For the processing phase (Step (7)), each site has an input subset R_k of size $O(\frac{n_1}{K})$ and a parameter subset T_k of size $O(\frac{n_1}{\kappa})$, and the merge-join can be carried out in parallel with $O(\frac{n_1+n_2}{\kappa})$ I/O cost at a site provided the main memory at each site has a moderate size. Thus, the total I/O cost at each site is $O(\frac{n_1+n_2}{K})$, and Algorithm MIMO-FuzzyJoin will achieve linear speedup.

7. Summary

436

A linear recursive program is used to formalize problems to be parallelized by the divideand-conquer method. A necessary and sufficient condition which characterizes problems solvable by the divide-and-conquer method in parallel is obtained, and a parallel scheme is developed. The performance of the parallel scheme is analyzed, and a necessary and sufficient condition is obtained as to when linear speedup can be achieved. A generalization of the PADAC Scheme is developed, and a real application, the fuzzy join problem, is parallelized by the generalized PADAC Scheme with linear speedup.

THE DIVIDE-AND-CONQUER METHOD

437

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References

- R. Agrawal and H.V. Jagadish, "Multiprocessor transitive closure algorithms," Proc. Int'l. Symp. on Databases in Parallel and Distributed Systems, Austin, Tex., Dec. 1988, pp. 56-66.
- R. Agrawal and H.V. Jagadish, "Hybrid transitive closure algorithms," Proc. 16th Int'l. Conf. Very Large Data Bases, Brisbane, Australia, Aug. 1990, pp. 326-334.
- 3 A.V. Aho, J.E. Hopcropt, and J.D. Ullman, The Design and Analysis of Computer Algorithms. Addison-Wesley, 1974.
- S. Baase, Computer Algorithms: Introduction to Design and Analysis, 2nd ed., Addison-Wesley, 1988.
- F. Bancilhon and R. Ramakrishnan, "An amateur's introduction to recursive query processing strategies, Proc. ACM-SIGMOD Intel. Conf. on the Management of Data, 1986. 6. B. Baugstø and J. Greipsland, "Parallel sorting methods for large data on a hypercube database computer,"
- Proc. the Sixth Int'l Workshop on Database Machine, pp. 127-141. 1989. 7
- M. Beck, D. Bitton, and W.K. Wilkinson, "Sorting large files on a backend multiprocessor," IEEE Trans. on Computers, vol. 37, no. 7, pp. 769-778, 1988.
- 8. J.L. Bentley and M.I. Shamos, "Divide-and-conquer for linear expected time," Info. Proc. Letts, pp. 87-91. Feb. 1978.
- D. Bitton, H. Boral, D.J. DeWitt, and W.K. Wilkinson, "Parallel algorithms for the execution of relational 9.
- D. Billor, H. Dolar, D.J. Collin, and W.M. Hurmann, Fundated agrinding to the decident of relational database operations," ACM Trans. on Database Systems, vol. 8, no. 3, 1983.
 G.E. Blelloch, C.E. Leiserson, B.M. Maggs, C.G. Plaxton, S.J. Smith, and M. Zagha, "A comparison of sorting algorithms for the connection machine CM-2," Proc. the 3rd Annual ACM SPAA, 1991.
- S. Bondeli, "Divide and conquer: A parallel algorithm for the solution of a tridiagonal linear system of equations," Parallel Computing, vol. 17, July 1991. P. Bosc, M. Galibourg, and G. Hamon, "Fuzzy querying with SQL: Extensions and implementation aspects," 12
- Fuzzy Sets and Systems, vol. 28, pp. 333-349, 1988.
- S.S. Cosmadakis and P.C. Kanellakis, "Parallel evaluation of recursive rule queries," Proc. 5th ACM Symp. 13. on Principles of Database Systems, 1986.
- 14 S. Dar and H.V. Jagadish, "A spanning tree transitive closure algorithm," Proc. 8th Int'l Conf. Data Engineering. Tempe, Arizona, Feb. 1992, pp. 2–11.
- 15. D. DeWitt, J. Naughton, and D.A. Schneider, "An evaluation of non-equijoin algorithms," VLDB 1991. nn 443-452
- D. DeWitt, J. Naughton, and D.A. Schneider, "Parallel sorting on a shared-nothing architecture using proba-16. bilistic splitting," IEEE PDIS, pp. 280-291, 1991.
- 17 D. DeWitt and J. Gray, "Parallel database systems: The future of high performance database systems," CACM. pp. 85-98, July 1992
- 18 D. DeWitt, J. Naughton, D.A. Schneider, and S. Seshadri, "Practical skew handling in parallel joins," VLDB. 1992, pp. 27-40.
- 19 W.D. Frazer and A.C. McKellar, "Samplesort: A sampling approach to minimal storage tree sorting," JACM. vol. 17, no. 3, pp. 496-507, July 1970. "Fuzzy LUNA—Fuzzy database system library user's manual and fuzzy LUNA—fuzzy database system
- 20 library reference manual," OMRON Corporation, 1992.
- 21 S. Ganguly, A. Silberschatz, and S. Tsur, "A framework for the parallel processing of datalog queries," Proc. ACM SIGMOD Intel. Conf. on the Management of Data, Atlantic City, NJ, 1990.
- 22. K.-C. Guh and C. Yu, "Efficient management of materialized generalized transitive closure in centralized and parallel environment," IEEE Transaction on Knowledge and Data Engineering, vol. 4, no. 4, pp. 371-381. August 1992.

- 23. L.J. Henschen and S.A. Nagvi, "On compiling queries in recursive first-order databases," JACM, vol. 31, no 1 1984
- 24. D.E. Knuth, The Art of Computer Programming: Sorting and Searching, Reading, Mass., vol. 3, Addison-Wesley, 1973. 25. F.T. Leighton, Introduction to Parallel Algorithms and Architectures, Morgan Kaufmann, 1991
- 26. J.W. Lloyd, Foundations of Logic Programming, 2nd edition, Springer-Verlag, 1987.
- 27. D. Maier, The Theory of Relational Databases, Computer Science Press: Rockvill, MD, 1983.
- 28. S. Nagvi and S. Tsur, A Logical Language for Data and Knowledge Bases, Computer Science Press, 1989.
- 29. H. Nakajima, T. Sogoh, and M. Arao, "Fuzzy database language and library: Fuzzy extension to SQL," Proc. of the Second IEEE International Conference on Fuzzy Systems, pp. 477-482, 1993.
- 30. D.S. Parker, E. Simon, and P. Valduriez, "SVP-a model capturing sets, streams, and parallelism," Proc. 18th VIDB, pp. 115-126, 1992.
- M.J. Quinn, "Parallel sorting algorithms for tightly coupled multiprocessors," Parallel Computing, vol. 6, pp. 349–367, 1988.
- 32. G. Salton, Automatic Text Processing, Addison Wesley, 1989.
- 33. B. Salzberg, A. Tsukerman, J. Gray, M. Stewart, S. Uren, and B. Vaughan, "FastSort: A distributed single-input single-output external sort," SIGMOD, pp. 94-101, 1990.
- D.A. Schneider and D.J. DeWitt, "A performance evaluation of four parallel algorithms in a shared-nothing multiprocessor environment," Proc. ACM SIGMOD, pp. 110–121, 1989.
- 35. J. Ullman, Database and Knowledge-Base Systems, Computer Science Press, Inc., 1988. P. Valduriez and G. Gardarin, "Join and semijoin algorithms for a multiprocessor database machine," ACM TODS, vol. 9, no. 1, pp. 133–161, March 1984.
- 37. B.W. Wah and G. Li, "Optimal parallel evaluation of AND trees," Journal of Parallel and Distributed Com-
- puting, vol. 10, pp. 1–17, 1990.
 38. O. Wolfson and A. Silberschatz, "Distributed processing of logic programs," Proc. ACM-SIGMOD Intl. Conf. on the Management of Data, 1988.
- 39. O. Wolfson and A. Ozeri, "A new paradigm for parallel and distributed rule-processing." Proc. ACM SIGMOD Intel. Conf. on the Management of Data, 1990.
- Q. Yang and C. Yu, "Parallelization by the divide-and-conquer method," IEEE Systems, Man and Cybernetics Conference, Chicago, 1992, pp. 1265–1270.
- 41. Q. Yang, C. Yu, C. Liu, S. Dao, and T. Pham, "A hybrid transitive closure algorithm for sequential and parallel processing," IEEE Data Engineering, Houston, pp. 348–355, 1994. 42. Q. Yang, C. Liu, J. Wu, C. Yu, S. Dao, and H. Nakajima, "Efficient processing of nested fuzzy SQL queries,"
- L.A. Zadeh, "Fuzzy logic," IEEE Computer, pp. 83–93, April, 1995, pp. 131–138.
 L.A. Zadeh, "Fuzzy logic," IEEE Computer, pp. 83–93, April, 1988.
 W. Zhang, C. Yu, G. Wang, T. Pham, and H. Nakajima, "A relational model for imprecise queries," International
- Symposium on Methodologies in Intelligent Systems, Trondheim, Norway, 1993.