



MIT Open Access Articles

Cache-conscious scheduling of streaming applications

The MIT Faculty has made this article openly available. ***Please share*** how this access benefits you. Your story matters.

Citation	Kunal Agrawal, Jeremy T. Fineman, Jordan Krage, Charles E. Leiserson, and Sivan Toledo. 2012. Cache-conscious scheduling of streaming applications. In Proceedings of the twenty-fourth annual ACM symposium on Parallelism in algorithms and architectures (SPAA '12). ACM, New York, NY, USA, 236-245.
As Published	http://dx.doi.org/10.1145/2312005.2312049
Publisher	Association for Computing Machinery (ACM)
Version	Author's final manuscript
Accessed	Wed Jan 23 02:41:58 EST 2019
Citable Link	http://hdl.handle.net/1721.1/90261
Terms of Use	Creative Commons Attribution-Noncommercial-Share Alike
Detailed Terms	http://creativecommons.org/licenses/by-nc-sa/4.0/

Cache-Conscious Scheduling of Streaming Applications

Kunal Agrawal
Washington University in
St. Louis
kunal@wustl.edu

Jeremy T. Fineman
Georgetown University
jfineman@cs.georgetown.edu

Jordan Krage
Washington University in
St. Louis
jordan.krage@wustl.edu

Charles E. Leiserson
Massachusetts Institute of
Technology
cel@mit.edu

Sivan Toledo
Tel-Aviv University
stoledo@tau.ac.il

ABSTRACT

This paper considers the problem of scheduling streaming applications on uniprocessors in order to minimize the number of cache-misses. Streaming applications are represented as a directed graph (or multigraph), where nodes are *computation modules* and edges are *channels*. When a module fires, it consumes some data-items from its input channels and produces some items on its output channels. In addition, each module may have some state (either code or data) which represents the memory locations that must be loaded into cache in order to execute the module. We consider *synchronous dataflow graphs* where the input and output rates of modules are known in advance and do not change during execution. We also assume that the state size of modules is known in advance.

Our main contribution is to show that for a large and important class of streaming computations, cache-efficient scheduling is essentially equivalent to solving a constrained graph partitioning problem. A streaming computation from this class has a cache-efficient schedule if and only if its graph has a low-bandwidth partition of the modules into components (subgraphs) whose total state fits within the cache, where the *bandwidth* of the partition is the number of data items that cross intercomponent channels per data item that enters the graph.

Given a good partition, we describe a runtime strategy for scheduling two classes of streaming graphs: pipelines, where the graph consists of a single directed chain, and a fairly general class of directed acyclic graphs (dags) with some additional restrictions. The runtime scheduling strategy consists of adding large external buffers at the input and output edges of each component, allowing each component to be executed many times. Partitioning enables a reduction in cache misses in two ways. First, any items that are generated on edges internal to subgraphs are never written

out to memory, but remain in cache. Second, each subgraph is executed many times, allowing the state to be reused.

We prove the optimality of this runtime scheduling for all pipelines and for dags that meet certain conditions on buffer-size requirements. Specifically, we show that with constant-factor memory augmentation, partitioning on these graphs guarantees the optimal number of cache misses to within a constant factor. For the pipeline case, we also prove that such a partition can be found in polynomial time. For the dags we prove optimality if a good partition is provided; the partitioning problem itself is NP-complete.

Categories and Subject Descriptors

F.2 [Analysis of Algorithms and Problem Complexity]: General

General Terms

Algorithms, Performance, Theory

Keywords

Caching, dag, graph, partitioning, pipelining, scheduling, streaming, synchronous dataflow.

1. INTRODUCTION

Streaming programming models have received enormous attention in recent years, mostly because they are perceived as enablers of high-performance computing and because of the continuing shift from signal processing by analog hardware to digital signal processing. Examples include academic projects like StreamIt [27] and StreamC/KernelC [12], community-based open-source projects like GNU Radio [9], and commercial products including Simulink[®] [20] and LabVIEW [16].

Naturally, there has been extensive research on how to map these programming models to hardware and on how to schedule them so as to maximize various objective functions, often subject to constraints. Much of the research has focused on maximizing *throughput*, defined as the number of data items processed per unit time, and on minimizing delay or latency (the period between the time an input data item enters the computation and the time it affects an output

Permission to make digital or hard copies of all or part of this work for personal or classroom use is granted without fee provided that copies are not made or distributed for profit or commercial advantage and that copies bear this notice and the full citation on the first page. To copy otherwise, to republish, to post on servers or to redistribute to lists, requires prior specific permission and/or a fee.

SPAA '12, June 25–27, 2012, Pittsburgh, Pennsylvania, USA.
Copyright 2012 ACM 978-1-4503-1213-4/12/06 ...\$10.00.

data item) [6, 13]. More recently, additional metrics have been considered, such as power consumption [29], memory usage [4, 23], and cache efficiency [15, 21, 25].

Because streaming computational models are restricted, it is possible to map and schedule them efficiently. A streaming application can be represented as a graph (or multigraph) where nodes are *computation modules* and the edges are *channels* connecting the modules. Channels have buffers associated with them. The modules send data in the form of *tokens* (also called *messages* or *items*),¹ to each other via these channels. When a module *fires*, it consumes some tokens from its incoming channels, performs some computation, and produces some tokens on its outgoing channels. A module is *ready* to fire each time each of its input channels contains sufficient tokens to enable it to compute. Each module accesses some associated state, almost always of static size (but not identical across modules), when it fires. Designated channels stream (a possibly infinite number of) tokens into and out of the application.

This paper focuses on cache-efficient scheduling of streaming computations on a single processor. Because cache efficiency is such an important determinant of performance, there is a rich history of algorithms and data structures designed to minimize the number of cache misses incurred by a program (see [1, 3, 5, 7, 11, 24] for a sample). Cache efficiency is often overlooked in streaming models, since data items in channels are written exactly once and read exactly once, and they are not reused. Streaming applications exhibit two kinds of cache misses, however, that can be controlled using intelligent scheduling. First, since modules access their state when they fire, it may be advantageous to execute the same module many times once its state has been brought into the cache. Second, consider a module A which outputs some data item on a channel between itself and a module B . If modules A and B are scheduled in a quick succession, then this data item remains in cache. If they are not, then this data item may be spilled out to main memory, leading to a cache miss when module B eventually reads it. Therefore, it is advantageous to execute consecutive modules one after the other whenever possible. Minimizing one of these two kinds of cache misses may compete with the objective of minimizing the other, however, and we must balance the concerns intelligently while scheduling streaming applications.

The algorithms in this paper for minimizing the number of cache misses are based on the idea of partitioning. The streaming graph is divided into subgraphs, each of which fits in cache. That is, given a cache of size M , the sum of the state size in any single subgraph does not exceed M . Within subgraphs, we use small buffers (the minimum required on that edge), whereas on channels between subgraphs, we use large buffers. This approach reduces cache misses in two ways. After a particular subgraph is cached, it can be executed any number of times — as long as sufficient input data is available — without causing any cache misses to load or spill state. Additionally, each module of the subgraph is executed so that the data moving through the subgraph remains cached, with the exception of the initial input and final output. The key to minimizing cache misses using this method is to partition the modules to minimize the amount of data transferred through the channels that cross between subgraphs.

¹These terms are used interchangeably.

We assume that each time a module fires, the number of items it produces and consumes from each of its input and output channels is known in advance and that this value does not change during execution. Such computations are called *synchronous dataflow* graphs. Without loss of generality, we also assume that all data items are unit sized. Finally, we assume that the state size of each module is known in advance. These assumptions are satisfied by a wide range of streaming computations.²

This paper makes the following contributions:

- A theoretical argument that provides a lower bound for the cost of scheduling a directed acyclic graph (dag). Specifically, consider an optimal “well-ordered” partition \mathcal{P} into subgraphs each with $O(M)$ total state, where “well-ordered” means that the graph induced by contracting each component is acyclic. An optimal partition is one with minimum “bandwidth,” which is the total number of messages that must cross partition boundaries for each input item consumed. Then *every* schedule incurs an amortized cost of at least $\Omega((1/B) \text{bandwidth}(\mathcal{P}))$ cache misses per input item, where M is the cache size, B is the block size.
- An upper bound for certain dags. Specifically, suppose that we are given a well-ordered partition \mathcal{P} subject to certain additional restrictions. We show how to schedule the dag on a machine with cache size $O(M)$ and block size B with $O((1/B) \text{bandwidth}(\mathcal{P}))$ amortized cache misses per input item. If \mathcal{P} is optimal, the upper and lower bound match to within constant factors. Therefore, the partitioned scheduler incurs at most a constant factor more cache misses than the optimal scheduler if given a constant-factor larger cache.
- We also show how to find a good partition for the case of *pipelines*, where the streaming graph consists of a single directed chain of modules. For more general dag topologies, finding the minimum bandwidth well-ordered partition is NP-complete [8, ND15: Acyclic Partition].

The additional restrictions for the upper bound require the partition to have $O(M/B)$ edges leaving each subgraph, and each component to be schedulable with internal buffers having a total of $O(M)$ size. The latter condition holds when all modules have uniform input and output rates, and both conditions always hold for the pipeline even without uniform rates. Both conditions also hold for a wide class of dags without uniform rates.

2. MODEL AND DEFINITIONS

This section describes the analysis and streaming models. We discuss assumptions about the streaming graphs used throughout the paper and define needed terminology.

Analysis model

To analyze the cost of a schedule for a streaming application, we use the *external-memory model* or *I/O model* [1], sometimes also called the *disk-access model* (DAM). The

²Some streaming applications satisfy our assumptions except for very few modules, such as modules that extract symbols or packets from a waveform, modules that produce television frames from a compressed stream, and so on. Our techniques can still be used to schedule these computations, perhaps suboptimally, by forcing these models to the boundaries of subgraphs.

I/O model is a two-level memory hierarchy consisting of a fast internal memory (cache) of size M and a slow, arbitrarily large external memory (disk), organized into contiguous **blocks** of size B . An algorithm may only operate on data that resides in the cache — requesting data that is not in cache causes a **cache miss**, wherein the block containing the data must first be moved from disk to cache (likely causing other data to be evicted from the bounded cache). The cost in this model is the number of cache misses.

Streaming model

We model a streaming computation as a directed acyclic graph (dag) $G = (V, E)$. Each vertex $v \in V$ corresponds to a module, which has a predefined **state size** denoted by $s(v)$. Each edge corresponds to a directed channel between modules, with **buffers** (implementing FIFO queues) along each edge to store messages that have not yet been consumed by the receiving module. A module has (integral) parameters $in(u, v)$ and $out(v, w)$ specifying, respectively, the number of messages that must be consumed from the incoming edge (u, v) 's buffer each time v executes and the number of messages produced into outgoing edge (v, w) when v executes. When each module has at most 1 input and output (as in the pipeline case), we use $in(v)$ and $out(v)$ as a shorthand for the inputs/outputs consumed by that module. A streaming dag is **homogeneous** if $in(u, v) = out(u, v) = 1$ for every edge (u, v) , that is, each module consumes exactly one message from each of its input channels and produces exactly one message to each of its output channels.

In order to execute, or **fire** a module v , the entire state of that module must be loaded into the cache. Moreover, the module may only fire if all of its input buffers contain at least the requisite minimum number of messages.

Assumptions

Throughout the paper, we make the following assumptions about the streaming graph. Except for the last one, these assumptions are all either without loss of generality (and made only to simplify the exposition) or necessary to admit any reasonable solution.

We assume that all messages are unit size and that the state of size each module is at most M . The former assumption is without loss of generality given the arbitrary input and output rates. The latter is necessary to allow a module to be fully loaded into cache when fired.

We assume that the streaming graph contains a single source node s with no incoming edges that produces an infinite stream of input data. Similarly, the graph contains a single sink node t with no outgoing edges that consumes all terminal outputs. This assumption is without loss of generality, as a multisource or multisink dag can be transformed into one with a single source and sink.

We assume that the streaming dag is **rate matched**, by which we mean that the value $\prod_{(u,v) \in p} (out(u, v) / in(u, v))$ is identical for all directed paths p between a fixed pair of vertices. This property is necessary and sufficient to allow the dag to be scheduled without deadlocks with bounded buffers on all channels.

Finally, let $minBuf(e)$ denote the minimum buffer size required by channel e , which can be computed for rate-matched dags using the procedure described in [17]. We assume that for any subgraph $G_i = (V_i, E_i)$ induced by a subset of vertices $V_i \subseteq V$, we have $\sum_{e \in E_i} minBuf(e) =$

$O(\sum_{v \in V_i} s(v))$. In other words, for any subset of modules, the state size of the modules exceeds the minimum buffer size of channels connecting those modules. This assumption allows us to amortize the cost of reading/writing the buffers for modules against loading the module. For a large class of applications, such as pipelines and homogeneous dags, $minBuf(e) = in(e) + out(e)$, making this condition hold without loss of generality, since a module must regardless load this much of its input and output each time it fires.

Additional definitions

Finally, we define some terms and notation used throughout the paper. When there exists a directed path from u to v in the dag, we say that u **precedes** v , denoted by $u \prec v$. When referring to a particular execution of a module v , the **progeny** of this execution are those messages that may (eventually) appear later in the dag as a result of this execution. In other words, the progeny of a particular execution of v are those messages that may be produced only after v is fired, but before the next time v is fired. We similarly define the progeny of a particular message m along channel (u, v) as the progeny of the execution of v that consumes message m .

We use the term “gain,” defined as follows, to describe the rate of amplification of messages along paths through the dag. Gain is only well defined for rate-matched dags.

DEFINITION 1. For a vertex v , the **gain** of the module is the number of times v fires for each time the source s fires. That is, for any path $p : s = x_0 \rightarrow x_1 \rightarrow x_2 \rightarrow \dots \rightarrow x_{|p|} = v$, we define

$$gain(v) = \prod_{i=1}^{|p|} (out(x_{i-1}, x_i) / in(x_{i-1}, x_i)) .$$

For an edge (u, v) , the **gain** of the edge is defined as the number of messages produced along the edge for each time the source s fires, denoted by $gain(u, v) = gain(u) out(u, v)$.

3. SCHEDULING AND EXECUTION

In this paper, we concentrate on schedules induced by a “partition” of the dag. Sections 4 and 5 shows that partitioning is a good strategy for scheduling streaming dags of various topologies if the goal is to minimize the number of cache misses. This section describes partition scheduling and the execution model that ensues when streaming applications are scheduled using partition scheduling.

A **partition** of a streaming dag V is a collection of disjoint subsets of vertices $\{V_i\}$ such that $\bigcup_i V_i = V$. We call each of the sets V_i in the partition a **component** of the partition. The edges that are internal to a particular component are called **internal edges** and the edges that cross from one component to another are called **cross edges**. We are interested in “well-ordered” partitions:

DEFINITION 2. Consider a partition $\mathcal{P} = \{V_1, V_2, \dots, V_k\}$ of a streaming dag V . The partition \mathcal{P} is **well ordered** if the multigraph induced by contracting each component V_i to a single vertex is a dag.

In addition, we want the partition to have the property that each component V_i “fits” in cache. That is, the state of all the modules in the subset, as well as the buffers on internal edges, all fit in cache at the same time. We call

a partition of modules $\{V_1, V_2, \dots, V_k\}$ a ***c*-bounded partition** if the total state size $\sum_{u \in V_i} s(u)$ of all modules within each subset is at most cM . The quality of a partition depends on its “bandwidth”:

DEFINITION 3. Consider a partition $\mathcal{P} = \{V_1, V_2, \dots, V_k\}$ of a streaming dag V , and let C be the set of cross edges. The **bandwidth** of \mathcal{P} is the sum of the gains of the cross edges:

$$\text{bandwidth}(\mathcal{P}) = \sum_{(x,y) \in C} \text{gain}(x,y).$$

Thus, for homogeneous applications, the bandwidth is simply the total number of crossing edges.

We say that a well-ordered *c*-bounded partition is an **optimal *c*-bounded partition** if there is no other well-ordered *c*-bounded partition of smaller bandwidth. We denote the bandwidth of an optimal *c*-bounded partition of the graph G by $\text{minBW}_c(G)$.

Partition scheduling begins by finding a *c*-bounded partition with small bandwidth. The scheduling of a partitioned dag is now considered at two levels: the higher level corresponds coarsely to scheduling components, and the lower level corresponds to scheduling modules within each component.

At the lower level, once a component is brought into cache, the modules within it must be scheduled. Each internal edge e is allocated a buffer with size $\text{minBuf}(e)$, which is the minimum buffer size required by e to avoid deadlocks. Since we have

$$\sum_{e \text{ incident on } v} \text{minBuf}(e) = O(s(v))$$

for all v , the sum of internal buffers sizes of a component V_i is $O(s(V_i)) = O(M)$. Therefore, these internal buffers fit in cache. Since a component is a rate-matched dag, one can always schedule at the lower level without overflowing these buffers [17].

For the higher level, the main idea is that once a component is loaded into the cache, it may be executed many times without incurring any additional cache misses, except when reading from or writing to those cross edges leaving the component. Two properties of the scheduler and partition enable this approach: the addition of large buffers along cross edges, and the fact that the partition is well ordered.³

Since the component is executed many times, we can amortize the cost of loading the component’s state against the number of reads/writes to cross edges (the bandwidth of the partition). We say that a component is **schedulable** whenever (1) “enough” inputs exist, and (2) “enough” space remains in the output buffers. In particular, the component is schedulable if it can be executed continuously until it consumes a total of at least $\Omega(M)$ inputs (or produces at least $\Omega(M)$ outputs) along cross edges to pay for loading the $O(M)$ state. The buffers on the cross edges must be large enough to ensure that, at all times, some component is schedulable. By enforcing this condition, we not only guarantee that the scheduling algorithm is deadlock free, but we allow dynamic schedules to be computed easily. The sizes of

³If the partition were not well ordered, then a single component could not necessarily be scheduled repeatedly in isolation: some other module might need to be scheduled in between modules of the component.

buffers on cross edges to satisfy this schedulability constraint is different for different stream topologies.

Scheduling inhomogeneous graphs

Inhomogeneous graphs may require large buffers on cross edges. How to improve the buffer sizes for inhomogeneous graphs is an interesting problem, but since the size of the cross-edge buffers does not affect our cache bounds, we leave this problem open. To design a schedule for this case, first compute any value T such that for every edge (u, v) , the value $T \text{gain}(u, v)$ is integral, is divisible by $\text{out}(u, v)$ and by $\text{in}(u, v)$, and is at least M . Under these conditions, all progeny of the T executions of the source node s can pass through the entire dag and be consumed by the sink t without requiring any future inputs at any node in the dag. We thus schedule at a granularity of T inputs. For each cross edge (u, v) , allocate a buffer with size $T \text{gain}(u, v)$. For each internal edge, as already stated, allocate a buffer of size $\text{minBuf}(u, v)$. At the higher level, execute the components in a topological-sort order, loading each component V_i exactly once per T inputs. For the lower level, execute the modules within the component V_i until all relevant progeny of the T source-node executions have been consumed and no items are buffered by this component except in the outgoing cross edges. The low-level scheduling can be accomplished by repeatedly choosing any module that can be fired without exceeding output buffer size.

Note that at the high level, once a component V_i is loaded, it is executed fully before proceeding to a different component. Moreover, it is never executed again until the next batch of T inputs. In contrast, at the lower level, many different modules may be interleaved with a successive executions of a particular module due to the tighter size restrictions on internal buffers.

Scheduling homogeneous graphs

For stream graphs where every module reads and writes exactly one item from its input and output channels, the scheduling is simpler. In this case, setting $T = M$ and proceeding as above suffices, but the lower level also becomes even more straightforward as $\text{minBuf}(e) = 1$ along all edges. In particular, at the higher level, each time the component is loaded, it is fired repeatedly until consuming M inputs along all incoming cross edges and producing M outputs along all outgoing cross edges. At the lower level, the modules are topologically sorted and are each fired just once in order; this lower-level schedule repeats M times.

In this case, one can extend this approach to an asynchronous or parallel dynamic schedule. (Throughout the remainder of this paper, however, our analyses are performed assuming the uniprocessor case.) To schedule components, choose any component(s) with M data items on all incoming cross edges and empty outgoing cross edges. Then schedule each internal module M times as above until the incoming buffers are empty and the outgoing buffers are full. The homogeneity of the graph ensures that it is always possible to find a schedulable component.

Scheduling pipelines

In the case of **pipelines**, where modules form of a chain (single directed path) but can have nonunit input and output rates, it is again possible to reduce the buffer sizes on cross edges. Each cross edge is assigned a buffer size of $\Theta(M)$. The

schedule works a little differently, however. A component is schedulable whenever its input buffer is at least half full and its output buffer is at most half full. It is then executed until either the input buffer becomes empty or the output buffer becomes full. Thus, either $\Omega(M)$ inputs are read, or $\Omega(M)$ outputs are written. A continuity argument now suffices to show that some component can always be scheduled. Scan the cross edges in topological-sort order until the first cross edge that is at most half full is encountered. The preceding component has a more-than-half-full input buffer by construction, and so it is schedulable. (The sink of the graph is treated as though its output buffer is always empty.) This schedule also readily generalizes to the asynchronous or parallel case as with the homogeneous graphs.

4. STREAMING PIPELINES

This section proves that a good partitioning schedule is nearly optimal for the problem of scheduling a pipeline with arbitrary input and output rates. More precisely, a good partitioning schedule has $O(Q)$ cache misses when run on a cache of size $O(M)$, where Q is the minimum possible number of cache misses of any schedule on a size- M cache. In other words, this schedule is $O(1)$ -competitive when given $O(1)$ cache augmentation. Moreover, this schedule can be found in polynomial time. Proving the optimality of the schedule hinges on a strong lower bound, which is a substantial part of the technical content of this section.

The crux of the lower bound is to show that any schedule must “pay” for messages to pass certain edges in the pipeline, essentially showing that the bandwidth of the best partition provides a lower bound on the cost of any schedule. We later present a more general lower bound (Theorem 10 of Section 5), which extends this partitioning lower bound to the more general case of dags. Theorem 10 thus subsumes the lower bound of this section, albeit with different constant factors. Nevertheless, we include both, as the lower bound here is less complicated and provides different intuition. Moreover, although finding the optimal partition for general dags is NP-complete, and we are not aware of any approximation algorithms, the pipeline lower bound immediately suggests a polynomial algorithm for finding an asymptotically optimal schedule for minimizing cache misses on pipelines.

For a pipeline, the well-ordered partitions can be represented compactly as a collection of segments. A ***u-v segment***, denoted by $\langle u, v \rangle$, corresponds to the set of all modules x with $u \preceq x \preceq v$. We define the ***gain-minimizing edge*** $\text{gainMin}(u, v)$ to be an edge with minimum gain in the segment $\langle u, v \rangle$.

The following lemma states that if the first node u in a segment $\langle u, v \rangle$ is fired enough times, then at least one of two things happens: (1) many messages are buffered within the segment, or (2) some output is produced from v . In either case, a certain number of cache misses occur, either due to the buffered messages in the case of (1) or due to the loading of state of the entire segment in the case of (2). One subtle point about the proof is that although the lemma is stated with respect to the number of times the module u is fired, we shall in fact charge against the gain-minimizing edge. Specifically, the eventual lower-bound theorem does not charge for reading the inputs to u — cache misses are assessed only for loading the state of $\langle u, v \rangle$ or buffering inside $\langle u, v \rangle$.

LEMMA 1. *Consider a segment $\langle u, v \rangle$ of a pipeline graph with gain-minimizing edge (x, y) . Then module u may be fired at most $2M(\text{gain}(u)/\text{gain}(x, y))$ times before either*

- *some progeny of one of these executions of u is output from module v , or*
- *at least $2M$ progeny of these executions of u are stored in buffers between modules u and v .*

PROOF. If any progeny is output from the segment, the claim holds trivially. Consider the case where no progeny are output from module v . Then the progeny of u must be buffered somewhere between u and v . The term $\text{gain}(z_1, z_2)/\text{gain}(u)$ is by definition the number of messages that (eventually) pass through edge (z_1, z_2) each time u is fired. These messages may be buffered or they may be passed to the next module. The number of messages buffered is thus minimized when $\text{gain}(z_1, z_2)$ is minimized, implying that at least $\text{gain}(x, y)/\text{gain}(u)$ messages must be buffered each time u fires. Hence if u fires at least $2M(\text{gain}(u)/\text{gain}(x, y))$ times, then at least $2M$ messages must be buffered. \square

If we apply Lemma 1 to a large enough segment, we see that if u is fired enough times, then either $\Omega(M)$ messages must be buffered, or $\Omega(M)$ state must be loaded, thereby incurring $\Omega(M/B)$ cache misses implying a lower bound on the number of cache misses. The following corollary formalizes this observation.

COROLLARY 2. *Consider any segment $\langle u, v \rangle$ of a pipeline graph satisfying the constraint $\sum_{z \in \langle u, v \rangle} s(z) \geq 2M$, and suppose that (x, y) is the gain-minimizing edge for this segment. Then any (sub)schedule that fires module u at least $2M(\text{gain}(u)/\text{gain}(x, y))$ times must incur at least $\Omega(M/B)$ cache misses. In other words, $\Omega((1/B)\text{gain}(x, y)/\text{gain}(u))$ is a lower bound on the amortized cost of firing u . This bound holds even if u 's inputs and v 's outputs are not counted towards cache misses.*

PROOF. Between the time a new message enters the segment and some progeny of that message leaves the segment, each of the modules in the segment must be in cache. Since the total state of these modules is at least $2M$ and can only initially hold at most M of this state, there must be at least M/B cache misses.

Similarly, if $2M$ new messages are buffered, they must be buffered in different memory locations, at most M of which already reside in cache. Thus, there must be at least M/B cache misses.

Applying Lemma 1, we see that the maximum number of times u can be fired before either of these events occurs is $2M(\text{gain}(u)/\text{gain}(x, y))$, and each event causes $\Omega(M/B)$ cache misses. To calculate the amortized cost, divide the $\Omega(M/B)$ cache misses by the number of executions of u . \square

We conclude by showing that the total bandwidth of the gain-minimizing edges provide a lower bound on the schedule cost. In this theorem, we consider any schedule that fires the sink t of the graph, which is the latest node in the pipeline, at least $T \cdot \text{gain}(t)$ times, for T a large-enough integer. In other words, we consider schedules that produce a large-enough number of outputs from the pipeline. Note that in order for the sink node to fire at least $T \cdot \text{gain}(t)$ times, the source node s must also fire at least T times. The reason to consider outputs here is to avoid any accounting ambiguity due to buffered items throughout the rest of the pipeline.

THEOREM 3. Consider a pipeline graph in which $S = \{\langle u_i, v_i \rangle\}$ is any collection of disjoint segments such that each segment has total size at least $2M$. Then any schedule of the graph that fires the sink node t at least $T \cdot \text{gain}(t)$ times must incur at least $\Omega((T/B) \sum_{s \in S} \text{gain}(\text{gainMin}(s)))$ cache misses, as long as T is sufficiently large.

PROOF. If each segment can be considered separately, this theorem follows directly from Corollary 2, since firing t at least $T \cdot \text{gain}(t)$ times implies that module u_i must be fired at least $T \cdot \text{gain}(u_i)$ times.

To see that each $\langle u_i, v_i \rangle$ can be considered separately, notice that the only shared state across different segments is on the cross edges shared between two segments. Since Corollary 2 does not count those edges towards the number of cache misses, there is no double-counting of state/buffer cache misses. \square

To provide an upper bound, we show constructively that there indeed exists a partitioning schedule that has cost asymptotically matching the lower bound. The following lemma shows that the number of cache misses of a partitioned schedule is related to the bandwidth of a partition. This lemma uses unbounded buffers on cross edges, but this issue is resolved at the end of the section.

LEMMA 4. Consider any partition $\mathcal{P} = \{V_i\}$ of a pipeline graph into segments such that each segment contains at most M total state. Then on a machine with cache size $O(M)$, it is possible to schedule the pipeline such that the sink t fires $\lceil T \cdot \text{gain}(t) \rceil$ times with $O((T/B) \cdot \text{bandwidth}(\mathcal{P}))$ cache misses in total, as long as T is sufficiently large.

PROOF. Let C be the set of cross edges induced by partition \mathcal{P} . Consider the segments V_1, V_2, \dots in topologically sorted order, where $V_i = \langle u_i, v_i \rangle$ and the edge $(v_i, u_{i+1}) \in C$. In sorted order, fire u_i a total of $O(T \cdot \text{gain}(u_i))$ times using a local schedule for V_i that has small bounded buffers on internal edges as described in Section 3. Since each V_i fits in $O(M)$ space including the internal buffers, we can repeatedly execute the loaded modules while only paying for reading and writing to external buffers. The total cost is therefore $O(M/B)$ to load V_i 's state plus $O((1/B)T(\text{gain}(v_{i-1}, u_i) + \text{gain}(v_i, u_{i+1})))$ to read inputs for u_i and to write outputs from v_i . For sufficiently large $T = \Omega(M/\text{gain}(v_{i-1}, u_i))$, the number of items read by V_i is $\Omega(M)$, costing $\Omega(M/B)$ cache misses, which dominates the cost. (The value T must also be large enough that t can be fired $\lceil T \cdot \text{gain}(t) \rceil$ times after firing s at most $O(T)$ times, which occurs when T is at least as large as the gain of every edge in the graph.) Summing across all V_i completes the proof. \square

The following theorem concludes that our lower bound is tight, modulo constant factors in both the number of cache misses and cache augmentation. Specifically, if Q is the optimal cache cost provided by any schedule (partitioned or not) on a machine with M cache, then there exists a partitioned schedule with cost $O(Q)$ cache cost on a machine with $O(M)$ cache. The upper-bound schedule is exhibited constructively.

THEOREM 5. There exists a partition $\mathcal{P} = \{V_i\}$ of any pipeline graph into segments such that for sufficiently large integer T ,

- every schedule (not necessarily a partitioned schedule) that fires t at least $T \cdot \text{gain}(t)$ times must cost $\Omega((T/B) \cdot \text{bandwidth}(\mathcal{P}))$ cache misses in total for a machine with a size- M cache, and
- there exists a partitioned schedule (based on \mathcal{P}) that fires t a total of $\lceil T \cdot \text{gain}(t) \rceil$ times and costs $O((T/B) \cdot \text{bandwidth}(\mathcal{P}))$ cache misses in total for a machine with a size- $O(M)$ cache.

Moreover, such a partition can be found in polynomial time.

PROOF. Construct segments W_i as follows. These segments will be used to build V_i later. Start at the beginning of the pipeline, with the current segment initially being W_1 , and consider all modules in topologically sorted order. Add modules to the current segment W_i until the total state size of W_i exceeds $2M$. If there is more than $2M$ state remaining, finish with W_i and proceed with initially empty W_{i+1} as the current segment. If, on the other hand, less than $2M$ state remains, add all remaining modules to the current segment. By construction, each W_i has total state at least $2M$. Moreover, since all modules have state size at most M , each W_i contains at most $5M$ state. (In fact, only the last segment can be this large, as all preceding segments have state between $2M$ and $3M$.)

For each W_i produced through this process, let $(x_i, y_i) = \text{gainMin}(W_i)$ be the gain-minimizing edge for the segment. Then let $C = \bigcup_i \{(x_i, y_i)\}$ be the set of cross edges. These cross edges induce a partition $\mathcal{P} = \{V_i\}$, that is, with $V_0 = \langle s, x_i \rangle$, $V_i = \langle y_i, x_{i+1} \rangle$ for $1 \leq i < |C|$, and $V_{|C|} = \langle y_{|C|}, t \rangle$, where s and t are the source and sink of the pipeline, respectively. Since each W_i contains at most $5M$ state, and each V_i spans at most 2 segments W_i and W_{i+1} , it follows that each V_i contains at most $10M$ state. In fact, this bound may be tightened to $8M$, since only the last segment can be so large. Therefore, for $c = 8$, this partition has the property that each segment is of size at most cM .

To achieve the lower bound, apply Theorem 3 with $S = \{W_i\}$. To achieve the upper bound, apply Lemma 4 with partition $\{V_i\}$. \square

COROLLARY 6. For a given pipeline graph and a sufficiently large number of outputs to produce, one can construct a schedule that incurs at most $O(1)$ times as many cache misses as the optimal schedule, as long as the schedule is allowed to use a constant factor more cache than the optimal schedule. \square

Although the partition described in Theorem 5 provides an asymptotically optimal upper bound on the number of cache misses (modulo memory augmentation), this partition is not the minimum bandwidth c -bounded partition. As it turns out, one can find the minimum bandwidth c -bounded partition for pipelines using a simple dynamic program. This (minimum bandwidth) partition provides no more cache misses than the partition described in Theorem 5, but not asymptotically fewer. Moreover, this optimal partition does not guarantee optimal cache performance. It still only guarantees *asymptotically* optimal cache performance.

Producing an optimal dynamic schedule

Lemma 4 (and Theorem 5) give schedules for the pipeline that rely on the number of outputs to produce. In particular, as specified, each segment V_i in the partition is considered in order and fired repeatedly until enough outputs are

produced. For the pipeline case, these schedules can be easily transformed into dynamic schedules, where the number $T \cdot \text{gain}(T)$ of times the sink must be fired is not specified *a priori*. Specifically, Lemma 4 requires only that each V_i consume $\Omega(M)$ inputs or produce $\Omega(M)$ outputs each time it is loaded. Thus, the segments can be scheduled as described in Section 3 in order to get the same bounds.

5. STREAMING DAGS

This section considers the case when the stream graph is a general directed acyclic graph (dag). A natural generalization of the pipeline schedule is the following: partition the dag into regions that fit in cache while minimizing the total gain of edges crossing the partitions. The key to showing that this type of strategy yields an asymptotically optimal cache bound is to provide a lower bound on the cache complexity of any schedule, showing that any schedule must incur at least as many misses (asymptotically) as some partition-based schedule. We first prove this claim for the homogeneous dataflow case (all gains are 1), and then we generalize to nonunit gains later in the section.

Proving a lower bound for the cache complexity of scheduling a streaming dag is significantly more complicated than for a pipeline. The following theorem shows that the bandwidth of an optimal c -bounded partition does indeed provide such a lower bound for the case of homogeneous dataflow. Recall that the term $\text{minBW}_3(G)$ used in the statement of the next theorem denotes the minimum possible bandwidth of a well-ordered partition into components such that each component has total state size at most $3M$. Without loss of generality, we assume a single input node (source) and a single output node (sink) to simplify exposition of the proof.

THEOREM 7. *Any schedule for a homogeneous dataflow dag that fires the sink node at least $T \geq B$ times must incur at least $\Omega((T/B) \cdot \text{minBW}_3(G))$ cache misses, where B is the block size.*

PROOF. The main idea of this proof is to look at (the progeny of) a single message passing through the entire dag from the source to the sink. We will argue that any schedule of this dag must pay at least $\Omega(1/B)$ block transfers for each edge crossing some 3-bounded partition (in particular, an optimal 3-bounded partition). Since modules may be executed more than once each time they are loaded, we must be careful about the way we count the block transfers incurred by loading the state of a module. It is not obvious that we can analyze a single message in isolation, although the proof will essentially do just this.

Consider any schedule π of modules that fires the sink T times. A schedule π is a list of module executions $\pi = u_1, u_2, \dots, u_m$ for some m . The same module can occur many times in the list (i.e., $u_i = u_j$ for some $j \neq i$). In particular, with all gains equal to 1, every module occurs at least T times in any schedule that produces T outputs. This interpretation of a schedule ignores whether messages are stored in intermodule buffers or not, but the proof charges for buffering where necessary.

Starting at the beginning of the schedule π , partition the schedule into contiguous subschedules $\pi = \pi_1, \dots, \pi_r$, each containing between $2M$ and $3M$ distinct state (except for the last subschedule which may be smaller). Such a partition is possible because each module has size at most M .

We first claim that beginning each subschedule π_i with an empty cache does not increase the overall cost of the schedule π by more than a constant factor. This fact follows because each π_i contains at least $2M$ state, and thus it must incur $\Omega(M/B)$ cache misses regardless of what is in the cache at the beginning of that subschedule. We can therefore afford to pay $O(M/B)$ to load any arbitrary initial cache state at the beginning of each subschedule without asymptotically increasing the cost.

Now look at any subschedule π_i . Let K_i be the total number of outputs produced by modules in π_i that are not consumed as input by another module within π_i . Each of these outputs must be written to cache or memory, taking a total of K_i space. Since we can assume that the cache is flushed at the end of each subschedule, the cost of executing π_i becomes $\Omega(K_i/B)$.

Showing that $\sum_i K_i \geq T \cdot \text{minBW}_3(G)$ will complete the proof. We do so by arguing that the progeny of a single source-node firing must contribute at least $\text{minBW}_3(G)$ edges to $\sum_i K_i$ (and hence the total for T firings is at least T times more). Proving this fact has two components. We first show that π can be reduced to a schedule π' , tracking only the progeny of one firing, such that every edge crossing a subschedule boundary in π' also crosses the same boundary in π . Then we argue that π' is a well-ordered, 3-bounded partition. If both of these claims hold, then the number of edges that a single firing contributes to $\sum_i K_i$ is at least the number of edges crossing some well-ordered 3-bounded partition, and hence at least $\text{minBW}_3(G)$.

To reduce π to π' , consider a single source-node execution and all of its progeny throughout the dag, and consider the schedule $\pi' = \pi'_1, \dots, \pi'_r$ induced by removing all but the corresponding modules from π . This schedule corresponds to every module firing exactly once within π' . All edges in π' also occur in π . Thus, each edge crossing subschedule boundaries in π' also crosses subschedule boundaries in π and contribute to some K_i .

To show that π' is a well-ordered 3-bounded partition, observe that since each π_i (and hence π'_i) contains at most $3M$ state, π' corresponds to a 3-bounded partition. Moreover, since every module is included just once in the schedule and the schedule obeys precedence constraints, it follows that modules π'_1, \dots, π'_k have no edges pointing to earlier modules. Hence each contracted π'_i has no edges pointing to the earlier components, and the contracted graph is a dag, meaning that it is well-ordered. \square

We now prove the upper bound, that is, if a good partition exists, then it is possible to schedule the pipeline with asymptotically optimal cache performance on a machine with $O(1)$ cache-size augmentation. In order to schedule a dag using partitioning, we need the partition to be *degree-limited*, meaning each component of the partition has degree $O(M/B)$, for the following reason. When a component is loaded into cache and executed, the cache should also be large enough to accommodate at least one block from the buffers on cross edges that are incident on this component. For a wide class of graphs, all $O(1)$ -bounded partitions are degree limited. For example, if every module u contains at least $\Omega(Bd_u)$ state, where d_u is its degree, then all $O(1)$ -bounded partitions are degree limited. The following lemma shows that the bandwidth of the partition determines the number of cache misses by a partitioned schedule.

LEMMA 8. Consider any degree-limited $O(1)$ -bounded partition $\mathcal{P} = \{V_i\}$ of a homogeneous dataflow dag. Then on a machine with $O(M)$ cache size and for sufficiently large T , there exists a schedule of the dag that fires the sink node T times with $O((T/B) \cdot \text{bandwidth}(\mathcal{P}))$ cache misses in total.

PROOF. As mentioned in Section 3, we add (large) buffers of size $\Theta(M)$ to each cross edge and then load each component into cache one at a time, executing each $O(M)$ times once loaded. Since each V_i fits in $O(M)$ space including the internal buffers, we can repeatedly execute the loaded modules while only paying for reading all the inputs from and writing all the outputs to external buffers. The total cost is then $O(M/B)$ to load V_i 's state plus the cost of reading the inputs from the incoming edges and writing outputs to outgoing edges. For large enough T , i.e., $T > M$, the number of inputs read is $\Theta(M)$ from each input edge (similarly for outputs) each time a component is scheduled. Since the degree of each partition is $O(M/B)$, we can afford to keep at least 1 block for each external buffer in cache at the same time, and thus reading/writing these external messages has cost $\Theta(1/B)$ times the number of message read or written. Therefore, reading the inputs and writing the outputs costs $\Theta(M/B)$ per cross edge, which dominates the cost of reading the state. Summing across all V_i and repeating T/M times completes the proof. \square

The following corollary (which follows directly from Theorem 7 and Lemma 8) says that a partitioned schedule is asymptotically optimal given $O(1)$ -memory augmentation. More generally, since finding the optimal well-ordered partition is NP-complete [8], we show that if we have an α -approximation for the partitioning problem, then we have an $O(\alpha)$ -competitive schedule.

COROLLARY 9. Let \mathcal{P}_{OPT} be an optimal 3-bounded partition of a homogeneous streaming dag G , and let \mathcal{P} be any 3-bounded degree-limited partition of G . Suppose that we have $\text{bandwidth}(\mathcal{P}) \leq \alpha \cdot \text{bandwidth}(\mathcal{P}_{\text{OPT}})$ for some $\alpha \geq 1$. Then for any sufficiently large number of outputs produced, the partition schedule using \mathcal{P} incurs at most $O(\alpha)$ times as many cache misses as the optimal schedule, as long as the partitioning schedule is allowed to use a constant times larger cache than the optimal schedule uses. \square

Notes on the upper bound

Corollary 9 is most useful for those dags with the following property: there exists a degree-limited 3-bounded partition with similar bandwidth to the optimal (not necessarily degree-limited) 3-bounded partition. When this property does not hold, the cost of the upper-bound partition may be worse by a factor of B (every read and write to cross-edge buffers may cause a cache miss), implying only that our schedule is $O(\alpha B)$ -competitive. The cost of a naive schedule, however, may be much worse, since only the messages moving along cross edges are charged. Moreover, we have not discussed how to find such a partition. Like most partitioning problems, finding an optimal well-formed 3-bounded partition is NP-hard. Since the partitioning occurs at compile time, however, and the streaming application is intended to be longer running, it may be reasonable to use an exponential-time algorithm for constructing a good partition. In any event, we have reduced the problem of scheduling to that of partitioning.

Generalizing to inhomogeneous streaming dags

Theorem 7 generalizes to the inhomogeneous case, but the reduction is not immediately obvious. Our approach is based on tracking fractional progeny through the dag and charging corresponding fractional costs to each edge crossing a subschedule boundary. The most natural fractional transformation is to treat a module v as though each time it fires, it consumes $\text{in}(u_i, v) \text{gain}(v)$ inputs on each input edge (u_i, v) and produces $\text{out}(v, w_j) \text{gain}(v)$ outputs on each output edge (v, w_j) . The problem with this transformation is that modification may restrict the optimal schedule, since for $\text{gain}(v) > 1$, the ‘‘fractional’’ module is forced to consume more inputs than the original module. Therefore, these obvious fractional amounts do not work. This approach of using fractional items is sound only if working with fractional messages makes the lower bound looser, that is, we give more power to the optimal schedule.

Instead, we track a tiny fractional message, with size $1/\gamma$, where γ is the product of all output rates in the entire dag. (This value γ is much larger than it needs be, but since this value is only employed for a proof and not in an algorithm, there is no need to reduce it.) A fractional module v then fires $\text{gain}(v)/\gamma$ times for each fractional firing of the source node of the dag.

THEOREM 10. Any schedule for a dataflow dag that fires the sink node at least $T \cdot \text{gain}(t) \geq B$ times must incur at least $\Omega((T/B) \cdot \text{minBW}_3(G))$ cache misses, where B is the block size.

PROOF SKETCH. The proof is nearly identical to that of Theorem 7, except that we track the $1/\gamma$ fractional progeny through the network, where γ is the product of all output rates in the dag. When nonunit outputs are produced, they contribute to some K_i in the same way, but they only contribute according to the size of the message written, which is $1/\gamma$ times the gain along the edge. When reducing π to π' , we consider each module firing once as described above, consuming inputs and producing outputs proportional to gains times $1/\gamma$. Thus, π' still contains each module once.

The one additional point in this proof is to argue that a fractional schedule is more powerful than every feasible nonfractional schedule, which follows from the fact that each module can be fired an integral number of times to simulate the nonfractional schedule. In particular, firing the fractional module v a total of $\gamma/\text{gain}(v)$ times is equivalent to firing the original module once. Since $1/\text{gain}(v)$ is the product of input rates divided by output rates, and γ is the product of all output rates, the resulting number is the product of the remaining integral output rates and some integral input rates. Thus, $\gamma/\text{gain}(v)$ is integral. \square

6. RELATED WORK

Most of the work on scheduling streaming applications aims to maximize throughput and/or maximize latency [2, 6, 18], minimize buffer sizes [4, 23, 28], or to optimize both while avoiding deadlocks [13, 19]. A few papers address other issues, such as power consumption [29]. None of these relates directly to our results.

Heuristic cache-aware scheduling of streaming programs on both single processors and multiprocessors has been studied by several research groups [15, 21, 25], but the proposed heuristics are all evaluated empirically. To the best of our

knowledge, no previous work provides any theoretical guarantees comparable to ours. Kohli [15] proposes a greedy scheduling heuristic for streaming dags that have a unique topological ordering (this class is a minor generalization of pipelines). The heuristic makes local decisions as to whether to continue to execute one module or to move on to its successor in the topological ordering based on an estimate of the number of cache misses that either decision will generate. Since this heuristic makes only local decisions, it does not provide an asymptotically optimal number of cache misses. Another difference from our work is that Kohli considers data-cache and instruction-cache misses separately, while we assume either that instruction cache is not an important bottleneck or that instructions are part of the state of the module. Sermulins et al. [25] also try to heuristically optimize for both the data cache and the instruction cache. Their schedule-optimization method starts from some given steady-state schedule that maintains bounded buffer sizes. Their optimizer takes this schedule and produces a new schedule which replaces each module invocation in the steady-state schedule by s back-to-back invocations. This scaling leads to state reuse, but it may cause spilling of buffers into main memory. Their method computes the largest s that avoids catastrophic spills. Since this method generates a small range of schedules, all of which are derived from the given steady-state schedule, it is suboptimal in many cases. Sermulins et al. also propose a module-fusion heuristic that can also reduce cache misses. This heuristic can be viewed as a special case of our partitioning method. Moonen et al. [21] also scale a given schedule in the same way, but in a more general setup that includes multiple processors and computational graphs that allow module to change their gains in a cyclic fashion. They do not provide any theoretical guarantees (it is unlikely that any can be proved for the method), but they do show convincing empirical results, demonstrating a cache-miss reduction of over a factor of 4 on a real-world application. Taken together, the empirical results strongly support our fundamental claim that effective scheduling can dramatically improve the performance of streaming applications.

Graph partitioning has also been used to improve cache efficiency under computational models other than streaming models. Examples include mesh processing for visualization [26], which provides theoretical guarantees, and repeated sparse matrix-vector multiplication [30], which employs an empirical heuristic methodology.

7. CONCLUSIONS

We have shown that in many cases the problem of cache-efficient scheduling of streaming applications can be reduced to the problem of partitioning the computational graph into components that have a small boundary and have a small-enough state to fit within the cache. On some classes of graphs, such as pipelines, solving the resulting partitioning problem is computationally easy. For more complex applications where the streaming graph is a synchronous dag, we can still show that the scheduling problem reduces to a partitioning problem. The problem of finding the optimal well-ordered partition for general dags is NP-complete [8], however. This situation is common to many mapping and scheduling problems in high-performance computing.

There are a few ways way to address this difficulty. One practical approach is to use an exact integer-programming

graph partitioner when the dag is relatively small. This strategy has proved effective in scheduling streaming computations on distributed memory-limited systems [22], for example. Another approach is to use a heuristic graph partitioner (see, for example, [10, 14]). Heuristic strategies are widely used to map and schedule large-scale parallel applications. Alternatively, since our results are approximation preserving, one can try to find a provably good approximation algorithm for the partitioning problem. We plan to address this issue in future work.

Another direction for future research is to study the cache-efficient scheduling of streaming computations on multiprocessors. If the number of cache misses is the only criterion, then the optimal uniprocessor schedule is trivially the optimal multiprocessor schedule. When considering multiprocessors, however, we must consider both load balancing and the number cache misses simultaneously.

Our work has also delineated the tougher theoretical problems in scheduling streaming applications. These include, not surprisingly, feedback (cycles in the graph) and modules whose output rates are not simple functions of their input rates (such as modules that sift through data and produce an output when they find something interesting, modules that make routing decisions, etc.). These computational structures also raise theoretical and practical difficulties unrelated to caching, such as the possibility of deadlocks due to insufficient buffer space. These issues, some of which are clearly online-scheduling problems, represent good opportunities for future research.

Acknowledgments

We would like to thank Fanny Dufossé of ENS Lyon for helpful discussions. This research was supported in part by NSF grants CCF-1150036, CNS-1017058, and CCF-0937860, by grant 1045/09 from the Israel Science Foundation (founded by the Israel Academy of Sciences and Humanities), and by grant 2010231 from the United-States-Israel Binational Science Foundation.

8. REFERENCES

- [1] Alok Aggarwal and Jeffrey Scott Vitter. The input/output complexity of sorting and related problems. *Communications of the ACM*, 31(9):1116–1127, September 1988.
- [2] Kunal Agrawal, Anne Benoit, Fanny Dufossé, and Yves Robert. Mapping filtering streaming applications. *Algorithmica*, pages 1–51, September 2010.
- [3] D. H. Bailey. FFTs in external or hierarchical memory. *Journal of Supercomputing*, 4(1):23–35, May 1990.
- [4] Mohamed Benazouz, Olivier Marchetti, Alix Munier-kordon, and Thierry Michel. A new method for minimizing buffer sizes for cyclo-static dataflow graphs. In *Proceedings of the 8th IEEE Workshop on Embedded Systems for Real-Time Multimedia (ESTIMedia)*, pages 11–20, 2010.
- [5] M. A. Bender, E. Demaine, and M. Farach-Colton. Cache-oblivious B-trees. In *41st Annual Symposium on Foundations of Computer Science (FOCS)*, pages 399–409, 2000.
- [6] Sardar M. Farhad, Yousun Ko, Bernd Burgstaller, and Bernhard Scholz. Orchestration by approximation mapping stream programs onto multicore

- architectures. *ACM SIGPLAN Notices*, 46:357–368, 2011.
- [7] Matteo Frigo, Charles E. Leiserson, Harald Prokop, and Sridhar Ramachandran. Cache-oblivious algorithms. *ACM Transactions on Algorithms*, January 2012.
- [8] Michael R. Garey and David S. Johnson. *Computers and Intractability: A Guide to the Theory of NP-Completeness*. W. H. Freeman & Co., New York, NY, USA, 1979.
- [9] GNU Radio, 2001. Software, gnuradio.org.
- [10] Bruce Hendrickson and Robert Leland. A multilevel algorithm for partitioning graphs. In *Proceedings of the 1995 ACM/IEEE conference on Supercomputing*, page 14 pages on CDROM, 1995.
- [11] Jia-Wei Hong and H. T. Kung. I/O complexity: the red-blue pebbling game. In *Proceedings of the 13th Annual ACM Symposium on Theory of Computing (STOC)*, pages 326–333, Milwaukee, 1981.
- [12] Ujval J. Kapasi, Peter Mattson, William J. Dally, John D. Owens, and Brian Towles. Stream scheduling. Concurrent VLSI Architecture Tech Report 122, Stanford University, Computer Systems Laboratory, March 2002.
- [13] Michal Karczmarek, William Thies, and Saman Amarasinghe. Phased scheduling of stream programs. *ACM SIGPLAN Notices*, 38(7):103–112, 2003.
- [14] George Karypis and Vipin Kumar. A fast and high quality multilevel scheme for partitioning irregular graphs. *SIAM Journal on Scientific Computing*, 20:359–392, 1998.
- [15] Sanjeev Kohli. Cache aware scheduling for synchronous dataflow programs. Technical Report UCB/ERL M04/3, EECS Department, University of California, Berkeley, 2004.
- [16] LabVIEW, 2011. Software, www.ni.com/labview.
- [17] E.A. Lee and D.G. Messerschmitt. Synchronous data flow. *Proceedings of the IEEE*, 75(9):1235–1245, 1987.
- [18] Edward Ashford Lee and David G Messerschmitt. Static scheduling of synchronous data flow programs for digital signal processing. *IEEE Transactions on Computers*, C-36:24–35, 1987.
- [19] Peng Li, Kunal Agrawal, Jeremy Buhler, and Roger D. Chamberlain. Deadlock avoidance for streaming computations with filtering. In *Proceedings of the 22nd ACM symposium on Parallelism in algorithms and architectures, SPAA '10*, pages 243–252, New York, NY, USA, 2010. ACM.
- [20] Mathworks. *Simulink User's Guide*, 2011. Release 2011b.
- [21] Arno Moonen, Marco Bekooij, Rene Van Den Berg, and Jef Van Meerbergen. Cache aware mapping of streaming applications on a multiprocessor system-on-chip. In *Proceedings of the conference on Design, Automation and Test in Europe (DATE '08)*, pages 300–305, 2008.
- [22] Ryan Newton, Sivan Toledo, Lewis Girod, Hari Balakrishnan, and Samuel Madden. Wishbone: Profile-based partitioning for sensor network applications. In *Proceedings of the ACM SIGCOMM Conference on Networked Systems Design and Implementation (NSDI 2009)*, April 2009.
- [23] Jongsoo Park and William J Dally. Buffer-space efficient and deadlock-free scheduling of stream applications on multi-core architectures. In *Proceedings of the 22nd ACM symposium on Parallelism in Algorithms and Architectures (SPAA '10)*, pages 1–10, 2010.
- [24] J. E. Savage. Extending the Hong-Kung model to memory hierarchies. In Ding-Zhu Du and Ming Li, editors, *Computing and Combinatorics*, volume 959 of *Lecture Notes in Computer Science*, pages 270–281. Springer Verlag, 1995.
- [25] Janis Sermulins, William Thies, Rodric Rabbah, and Saman Amarasinghe. Cache aware optimization of stream programs. *ACM SIGPLAN Notices*, 40(7):115–126, 2005.
- [26] M. Tchiboukdjian, V. Danjean, and B. Raffin. Binary mesh partitioning for cache-efficient visualization. *IEEE Transactions on Visualization and Computer Graphics*, 16:815–828, 2010.
- [27] William Thies, Michal Karczmarek, and Saman P. Amarasinghe. Streamit: A language for streaming applications. In *Proceedings of the 11th International Conference on Compiler Construction, CC '02*, pages 179–196, London, UK, 2002. Springer-Verlag.
- [28] Maarten H Wiggers, Marco J G Bekooij, and Gerard J M Smit. Efficient computation of buffer capacities for cyclo-static dataflow graphs. In *Proceedings of the 44th ACM/IEEE Design Automation Conference*, pages 658–663, 2007.
- [29] Ruibin Xu. *Energy-aware scheduling for streaming applications*. PhD thesis, University of Pittsburgh, 2010.
- [30] A. N. Yzelman and Rob H. Bisseling. Cache-oblivious sparse matrix-vector multiplication by using sparse matrix partitioning methods. *SIAM Journal on Scientific Computing*, 31:3128–3154, 2009.