iOverlay: A Lightweight Middleware Infrastructure for Overlay Application Implementations

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Abstract. The very nature of implementing and evaluating fully distributed algorithms or protocols in application-layer overlay networks involves certain programming tasks that are at best mundane and tedious — and at worst challenging — even at the application level. These include multi-threaded message switching engines at the application layer, failure detections and reactions, measurements of QoS metrics such as loss rates and per-link throughput, application deployment and terminations, debugging and monitoring facilities, virtualizing distributed nodes, as well as emulating resource bottlenecks and asymmetric network connections. Unfortunately, such a significant set of programming tasks is inevitable when implementing a diverse variety of application-layer overlay protocols and algorithms.

In this paper, we present *iOverlay*, a lightweight and high-performance middleware infrastructure that addresses these problems in a novel way by providing clean, well-documented layers of middleware components. The interface between iOverlay and overlay applications is designed to maximize the usefulness of iOverlay, and to minimize the programming burden of application developers. The internals of iOverlay are carefully designed and implemented to maximize its performance, without sacrificing the simplicity of application implementations using iOverlay. We illustrate the effectiveness of iOverlay by rapidly implementing a set of overlay applications, and report our findings and experiences by deploying them on PlanetLab, the wide-area overlay network testbed that iOverlay conveniently supports.

1 Introduction

Existing research in the area of application-layer overlay protocols has produced a sizable collection of real-world implementations of protocols and distributed applications in overlay networks. Examples include implementations of structured search protocols such as Pastry [1] and Chord [2], as well as overlay data dissemination such as Narada [3], NICE [4], SplitStream [5] and Bullet [6]. However, an interesting observation is that most of the existing work has resorted to simulations to evaluate the effectiveness of the proposed protocols. This phenomenon is certainly not surprising, since it is generally difficult to federate a large number of physical nodes that are globally distributed across the Internet, such that application implementations may be deployed globally to show their real-world performance and quality.

The recent emergence of global-scale implementation testbeds for application-layer overlay protocols comes to our rescue. Both PlanetLab [7] and Netbed's wide-area testbed [8] have been designed and implemented precisely for the purpose of evaluating new protocols and distributed applications over a wide-area overlay network. The availability of these testbed platforms makes it feasible to design, implement and deploy overlay protocols in a wide-area network, so that they may be evaluated in realistic environments rather than simulations. However, there still exist roadblocks that make it impractical for a small research group to deliver a high-quality, highperformance and fully distributed real-world implementation of overlay applications entirely from scratch: such an implementation involves many software components that must work together, including certain programming tasks that are at best mundane and tedious — and at worst challenging — to code.

We observe that, among all the components of a distributed application or protocol implementation, only a few specific areas are interesting for research purposes, and are subject to changes and innovations. On the other hand, any realistic implementation of overlay applications — in order to be useful even for collecting the first set of performance data — must include a significant number of largely uninteresting elements, such as bootstrapping wide-area nodes from a centralized authority, implementing a multi-threaded message forwarding engine, as well as monitoring facilities to control, debug, and record the performance of distributed algorithms. The necessity of writing this kind of supporting infrastructure not only slows down the pace of prototyping new applications and protocols, but also greatly increases the cost of entry to application-layer overlay research in realistic overlay testbeds, such that many small but useful experiments of newly conceived ideas are simply not viable.

In this paper, we present *iOverlay*, a lightweight and high-performance middleware infrastructure that is specifically designed from scratch to support rapid development of distributed applications and protocols over realistic testbeds. By *distributed applications*, we refer to both specific applications such as multimedia streaming or service composition, and application-layer overlay protocols such as multicast protocols for rapid data dissemination. The design objectives of *iOverlay* are as follows. First, it seeks to provide a high-quality and high-performance implementation of a carefully selected number of features that are common or useful to most of the overlay application implementations. Second, it seeks to be as *generic* as possible, and minimizes the set of assumptions with respect to the objectives and nature of new applications. Third, it seeks to significantly simplify the implementation of distributed applications, to the extent that only the logics and semantics specific to the application itself need to be implemented by the application developer. In addition, it should not be necessary for the application developer to have any prior knowledge about the internal details of iOverlay, before starting a successful implementation. Finally, it seeks to design a well-documented, straightforward and clean interface between the application and iOverlay.

The remainder of this paper is organized as follows. In Section 2, we open our discussions with an overview of the iOverlay architecture and highlights (Section 2.1), and proceed with a detailed account of various aspects of iOverlay: (1) the design of the message switching engine (Section 2.2); (3) the interface between iOverlay and algorithms (Section 2.3); and (4) the achievable performance with iOverlay (Section 2.4). In Section 3, we support our observations by presenting our own experiences with rapidly prototyping a set of overlay applications as case studies. Finally, we discuss iOverlay in light of related work (Section 4), and conclude the paper in Section 5.



Fig. 1. The iOverlay architecture.

2 iOverlay: Design and Performance

iOverlay considers three layers in a distributed application: (1) the message switching engine, which performs indispensable tasks of switching application-layer messages. (2) the algorithm, which implements the application-specific distributed protocol beyond mundane tasks in the engine; and (3) the application, which produces and interprets the data portion of application-layer messages at both the sending and the receiving ends. This may include global storage systems that respond to queries, or publish-subscribe applications that produce events and interests. The ultimate objective is for the application developer to build new algorithms based on the engine, and to select an application to be deployed on top of the algorithm.

Architecturally, the iOverlay middleware infrastructure provides support to the application developer in all of these aspects. First, it implements a fully functional, virtualizable and high performance message switching engine, upon which the application-specific algorithm is built. Second, it implements common elements of selected categories of algorithms that are completely optional for the application developer to use. Third, it implements typical applications, which the algorithm developer may choose to deploy. Finally, it provides a centralized Windows-based graphical utility, referred to as the *observer*, for the purpose of monitoring, debugging, visualizing and logging various aspects of the distributed application. The iOverlay architecture, as discussed, is illustrated in Fig. 1.

2.1 Highlights

The fundamental contribution that the iOverlay middleware infrastructure has brought is that it eliminates the need of "reinventing the wheel" with respect to uninteresting or challenging components that different overlay applications share. We now show a few highlights of the iOverlay architectural design.



Fig. 2. The observer in action with 10 PlanetLab nodes across the Internet. The black node is the current selection, while the gray (green) node is its selected downstream. The current outgoing throughput to this downstream is shown, along with the buffer size. The map may be conveniently switched to the North American map using the controls at the lower right corner.

Simplified interface. iOverlay is designed to have the simplest interface possible between the applicationspecific algorithm and the engine on each overlay node, in order to minimize the cost of entry to use iOverlay. The application developer only needs to be aware of one function of the engine: the send function, used for sending data or protocol messages to downstream or peer nodes. In addition to this function, the entire interface is designed to be completely message driven, in the sense that the algorithm only needs to passively process messages when they arrive or are produced by the engine. Since messages are distinguished by their types, a message handler that handles possible types is all that is required for the algorithm implementation. Further, the entire implementation of the application-specific algorithm is guaranteed to be executed in a single thread, and therefore does not need to use thread-safe data structures (those guarded with semaphores and locks).

Virtualized nodes. iOverlay features complete virtualization of overlay nodes in a distributed application. Each physical node in the wide-area network may easily accommodate from one to up to dozens of iOverlay nodes, depending on available physical resources such as CPU. Each iOverlay node has its own bandwidth specifications, such as the total bandwidth available to and from the node, separate upload and download available bandwidth, or per-link bandwidth limits. This adds to the flexibility of iOverlay deployment: if necessary, iOverlay may be entirely deployed in a local area network with a cluster of servers; or, for small-scale tests, on just a single server.

Maximized performance and portability. Finally, iOverlay is designed to maximize its performance. The engine is implemented from scratch with the C++ programming language and the native POSIX thread library in UNIX. It is also portable across UNIX variants, and may be compiled without changes on Linux, FreeBSD, or even Cygwin-based Windows environments. On the other hand, for the sake of simplicity of extensions and modifications of the graphical user interface, the observer is implemented in Windows using the C# programming

language in Visual Studio .NET, guaranteeing rapid development of additional interface elements, as well as the most impressive visual effects possible. Fig. 2 shows the current graphical user interface of the observer.

2.2 Internal Design

In iOverlay, we assume that all communication is in the form of *application-layer messages* (referred henceforth as *messages*), containing application data (or *payload*) of a maximum (but not necessarily fixed) length, in terms of bytes. The message structure is illustrated in Fig. 3. We strive to minimize the overhead (in terms of number of bytes) of using application-layer headers. To keep it simple, the content of a message is mostly immutable, and is initialized at the time of construction. In addition, the notion of a *node* in iOverlay is uniquely identified by its IP address and port number. The port number may be explicitly specified at start-up time; otherwise, the engine chooses one of the available ports.



Fig. 3. The application-layer message in iOverlay, with a fixed 24-byte header.



Fig. 4. The internal design of the engine. In this illustration, the engine has three *receiver* threads, three *sender* threads, and one *engine* thread that encapsulates the application-specific algorithm and the switch.

The message switching engine: a close examination

The engine of iOverlay is an application-layer message switch. We seek to design the engine such that it supports multiple competing traffic sessions, so that the application developer may easily test the performance of distributed algorithms under heavy cross traffic. It also has the capability to concurrently process both application data and protocol-specific messages.

We deploy a multi-threaded architecture to concurrently handle multiple incoming and outgoing connections, application-specific messages, as well as messages to and from the observer. Specifically, we use a thread-perreceiver and a thread-per-sender design, along with a separate *engine* thread for processing and switching messages using the application-specific algorithm. All receiver and sender threads use blocking receive and send operations, and the sender thread is suspended when the buffer is empty, to be signaled by the engine thread. We use a thread-safe circular queue to implement the shared buffers between the threads. Such a design is illustrated in Fig. 4.

We adopt such a design to avoid the complex wait/signal scenario where the receiver or sender buffer is shared by more than one reader or writer threads. Unlike the receiver and sender threads that "sleep" when the the buffer is full (receiver) or empty (sender), the engine thread constantly monitors the publicized port of the node (by using the non-blocking select() function) for incoming control messages from the observer, or from the algorithms of other nodes. If they exist, they are either processed within the engine, or sent to the algorithm to be processed, by calling the Algorithm::process() function. Next, it switches data messages from the receiver buffers to the sender buffers in a weighted round-robin fashion, with dynamically tunable weights (implemented in the Engine::switch() function). The skeleton of the engine thread is shown in Table 1.

Table 1. Design of the engine thread

start the TCP server on the publicized port; bootstrap from observer; while not terminated if there are incoming messages on the port detected using non-blocking select() if the message is engine-related call Engine::process(); else call Algorithm::process(); call Engine::switch(); stop the TCP server.

Obviously, when the switch attempts to forward messages to downstreams, the choice of downstream nodes is at the sole discretion of the algorithm. Therefore, the engine consults with the algorithm by calling Algorithm::process(). There are two possibilities. First, the algorithm may locally process and consume the message. Second, the algorithm continues to forward the message to one or more downstream nodes, by calling the Engine::send() function. Only in the latter case does the engine forward the message to the sender buffers.

The tight coupling of the algorithm's and the engine's message processing components is intentional by design. First, they must reside in the same thread, since we prefer to avoid the cases where the developer needs to use thread-safe data structures when algorithms are developed with iOverlay. It is impossible to design a typical two-thread solution — where the engine processes control messages in one thread, and switches data messages in another — and still achieve such a favorable property of accommodating thread-unaware algorithms. Second, the seemingly complex "paradox" — at times the engine calls the algorithm, and at other times the algorithm calls the engine — is in fact straightforward, since the algorithm is always *reactive* and never *proactive*.

There are further complexities involved in the design of a switch. As a first example, there may be cases where messages are successfully forwarded to only a subset of the intended senders, but fail to be forwarded to the remaining ones, since their buffers are full. In this case, we label each message with its *set of remaining senders*, so that they may be tried in the next round. As a second example, in some scenarios a set of states needs to be shared and exchanged between active threads. For example, a receiver thread needs to notify the engine when a failed upstream node has been detected, such that the engine thread can clear up its data structures related to this node. To avoid complex thread synchronization between active threads, we extensively take advantage of the mechanism of passing application-layer messages across thread boundaries via the publicized port. Without a doubt, these complexities are completely transparent to the algorithm developer.

Finally, we may not only wish to forward verbatim messages in an application-layer switch, but also wish to merge or code multiple incoming messages into one outgoing message. In order to implement the most generic *n*-to-*m* mapping (such as coding messages from *n* multiple incoming connections to *m* downstreams), we allow **Algorithm::process()** to return a *hold* type, instructing the engine that the message is buffered in the algorithm, but its processing should be put on hold to wait for other messages from other incoming connections. It is up to the algorithm to implement the logic of merging or coding multiple messages after requesting a hold on them, and eventually producing a new message to be sent to downstreams. Using the *hold* mechanism, we have successfully implemented algorithms that perform overlay multicast with merging or network coding [9].

Salient features

Handling of failures. In iOverlay, we assume that the nodes themselves, the virtual link between nodes, as well as the application data sources may all fail prematurely. Transparent to the algorithm developer, iOverlay supports the automatic detection of failed nodes and links, and the automatic tear-down of relevant links after such failures. For example, if an upstream link in a multicast tree has failed, it causes a "Domino Effect" that fails all downstream links from this point. The engine is able to appropriately tear down these links without affecting any of the other active links, and to notify the algorithm of such failures. All terminations are graceful, and all affected links are smoothly dropped without side effects.

We have implemented a collection of exception handling mechanisms to detect and process such failures. Depending on the state of the sockets at the time of premature failures, we rely on a combination of mechanisms to detect that a node or a link may have failed: (1) exceptions thrown and timeouts at the socket level; (2) abnormal signals caught by the engine, such as the *Broken Pipe* signal; and (3) long consecutive periods of traffic inactivity, detected by throughput measurements. To avoid overhead, we do not use any forms of active probes or "heartbeat updates" for this purpose. Still, we are able to implement very responsive detections of link and node failures in most cases. In addition, the observer may choose to terminate a node at will, in which case all the data structures and threads in both the engine and the algorithm will be cleared up, and the program terminates gracefully.

Measurement of QoS metrics. At the socket level, we have implemented mechanisms to measure the TCP throughput of a connection, as well as the round-trip latency and the number of bytes (or messages) lost due to failures. The results of these measurements are periodically reported to the algorithm and the observer. Upon requests from the algorithm, the available bandwidth and latency to any overlay nodes can be measured.

Emulation of bandwidth availability. In some cases, the algorithm developer prefers to test a preliminary algorithm under controlled environments, in which node characteristics are more predictable. iOverlay explicitly supports the emulation of bandwidth availability in three categories: (1) per-node total bandwidth: the total incoming and outgoing bandwidth available; (2) per-link bandwidth: the bandwidth available on a certain point-to-point virtual link; and (3) per-node incoming and outgoing bandwidth: iOverlay is able to emulate asymmetric nodes (such as nodes on DSL or cable modem connections) featuring disparate outgoing and incoming bandwidth availability. The emulated values may be specified at node start-up time, or within the observer at runtime. In the latter case, artificially emulated bottlenecks may be produced or relieved on the fly, in order to evaluate the adaptivity of the algorithm. To implement such emulations, we have wrapped the socket send and recv functions to include multiple timers in order to precisely control the bandwidth used per interval (the length of which may be specified by the algorithm).

Performance considerations

The performance objective of the engine design is to "push" messages through the engine as quickly as possible, with the lowest possible overhead at the switch. Towards this objective, we have considered three directions of performance optimizations, and successfully implemented them in the current engine.

Persistent connections. In order to avoid the unacceptable overhead of thread-level context switching at the operating system when a large number of threads are used, we implement both incoming and outgoing socket connections as *persistent connections*, in the sense that all the messages between two nodes are carried with the same connection, regardless of the applications they belong to. With persistent connections, we have avoided the creation of more threads when new distributed applications are deployed; instead, existing connections are reused.

Zero copying of messages. In order to avoid deep copying of entire messages when they pass through the engine, we have implemented a collection of mechanisms to ensure that only the references of messages are passed from the incoming socket all the way to the outgoing socket, and no messages will be copied in the engine at all. The algorithm may choose to copy messages, if necessary, supported by the copy constructor of the Msg

class. In order to appropriately destruct messages whose references are shared by multiple threads, an elaborate thread-safe reference counting mechanism is in place in the core of the engine.

Footprint. The engine is meticulously designed and tested so that the memory footprint is minimized and stable (without leaks). For example, with a message size of 5 KB and a buffer capacity of 10 messages, the footprint of the engine is only 4 MB per active connection¹. The optimized binary executable of the engine (with a simple testing algorithm) is only 100 KB. Such a footprint guarantees the scalability of iOverlay, especially when a large number of virtualized nodes are deployed on the same physical server.

The observer and its proxy

As a centralized monitoring facility, we have implemented the observer as a graphical tool in Windows, as illustrated previously in Fig. 2. The observer implements the first level of bootstrap support, by responding to any bootstrap requests (messages of type *boot*) with a random subset of existing nodes that are alive. The number of initial nodes in such a subset is configurable. Once a node is bootstrapped, the observer periodically sends it a *request* message to request for status updates, which include lengths of all engine buffers, measurements of QoS metrics, and the list of upstream and downstream nodes. With these status updates, the observer may visually illustrate the current network topology of each of the applications with geographical locations of all nodes, on either the world or the North American map.

Further, the observer serves as a control panel and may take the following actions to control the status of the network: (1) controlling the emulated per-link and per-node bandwidth availabilities; (2) deploying an application; (3) asking a node to join or leave a particular application; and (4) terminating an application data source or a node. For the sake of flexibility, the observer is also able to send new types of algorithm-specific control messages to the nodes, with two optional integer parameters embedded in the header.

Finally, the observer is able to record the content of any messages with the type *trace* in its log files. This mechanism serves as a centralized facility to collect and record debugging information, performance data and other traces. Alternatively, if the volume of traces becomes large, it may be more favorable to log them locally at each node, in which case iOverlay provides scripts to collect them after algorithm execution.

Initially, the observer is designed as a traditional multi-threaded TCP server on Windows. Our initial experiences with such a design have shown two problems. First, Windows XP Professional poses a very tight limit on the number of concurrently backlogged connections, such that when there are more than a few nodes reporting their states concurrently, the connection requests of some of them may be refused. Second, most of the Windows desktops are installed behind firewalls, preventing the updates to arrive from wide-area overlay nodes (e.g., on PlanetLab). To address both problems, we have implemented an efficient proxy to be executed in an UNIX environment outside of the firewall on the same local area network, such as on PlanetLab nodes or firewall gateways. In this case, the status updates from overlay nodes are submitted to the proxy, who relay them with a *single connection* to the observer. With the addition of the proxy, we have tested the observer handling incoming messages from thousands of virtualized nodes without problems.

Basic elements of algorithms

Despite the tight coupling between the algorithm and the engine, the algorithm is placed in its own namespace with an object-oriented design. The basic and commonly used elements of an algorithm is defined and implemented in a generic base class referred to as *iAlgorithm*. We present two examples. First, it implements a default message handler, that handles known messages from the observer and the engine with a default behavior. For example, upon receiving the bootstrap message from the observer, it records the set of initial nodes in a local data structure referred to as KnownHosts. Second, *iAlgorithm* implements a disseminate function, which disseminates a message to a list of overlay nodes, with a specific probability p. This resembles the gossiping behavior in distributed systems. The default implementations of a library of functions in the *iAlgorithm* class serve as a set of basic utilities, and since application-specific algorithms are classes that inherit from *iAlgorithm*, the developer may choose to override any default behavior with application-specific implementations.

2.3 Interface between iOverlay and Algorithms

Given the iOverlay design we have presented, how do we rapidly develop an application using iOverlay? Many design choices are made to reduce the complexity of developing new application-specific algorithms. First, the

¹ This is the case in Linux, which may be inferior with respect to footprint since clone() is usually used to support user-level POSIX threads.

algorithm namespace extensively uses object orientation such that new algorithms may be built based on existing algorithm implementations. As we have discussed, a few basic elements of algorithms have already been provided by iOverlay. Second, the algorithm only needs to call *one* function of the engine: the *send* function. This greatly improves the learning curve of the interface. Finally, the algorithm is designed as a message handler, in the form of a *switch* statement on different types of messages. While processing each incoming message, internal states of the algorithm may be modified. The message handler should reside in the **process()** function. The skeleton of an algorithm is shown in Table 2.

Table 2.	Skeleton	of the	algorithm	using	iOverlay
					-/

process(Msg * m)
<pre>switch (m -> type())</pre>
case sDeploy: (from observer)
deploy an application source;
case request : (from observer)
send algorithm status updates to observer;
case sTerminate: (from observer)
terminate an application source;
case BrokenSource: (from upstream)
clear up internal states corresponding to the application
source at upstream, since it has failed;
case data: (from the engine)
process, consume or forward the message using
<pre>send(Msg * m, Node dest);</pre>
case UpThroughput: (from the engine)
record or process the throughput from an upstream;
(process other engine or algorithm-specific types)
default: (use the default behavior from iAlgorithm)
iAlgorithm::process(m);

In such a skeleton, it is not necessary for an algorithm to handle all the known message types from the engine or the observer. If a message type is not handled in the algorithm, the default process() function provided by the base *iAlgorithm* class takes this responsibility. In fact, the only message type that the algorithm must handle is the type data, indicating a data message. iAlgorithm provides default handlers for all other types of messages. It is also not necessary for an algorithm to handle abnormal return values when invoking the send() function. In fact, send() has a return type of void, and all abnormal results of sending a message are handled by the engine transparently. For example, if the destination node of the message fails, the algorithm is notified appropriately, again via messages produced by the engine.

Another important design decision is related to the destruction of messages. In order to completely eliminate memory leaks, we need to carefully assign the responsibilities of message destruction. Particularly, consider a message passed to the algorithm (by pointers) as a parameter in the *process* function. Should the engine or the algorithm be responsible for destructing the message after it has been processed? Further, when a message is constructed in the algorithm and passed to the *send* function of the engine, should the engine or the algorithm be responsible for destructing the message after it is sent? To simplify the tasks of algorithm developers, we stipulate that all message destructions are the responsibility of the engine. The algorithm developer should never destruct messages, even if they have been constructed in the algorithm.

However, there exist a subtle problem with this solution even it works well at most times. When the algorithm receives a pointer to an engine-created message as a parameter of the *process* function, what if the algorithm passes the pointer back to the engine by using the *send* function? We distinguish treatments of this scenario depending on the type of the message. If the message is of type data, we have developed the engine carefully such that the algorithm can directly invoke *send* with the same message, guaranteeing zero copying of data messages. However, if the message is of any other type, we require the algorithm developer to clone the message before invoking *send* on the new copy. Performance-wise this is not a problem, since most protocol messages are very small in size.

2.4 iOverlay: Performance and Correctness

With C++ on Linux, C# on Windows, and around 19,000 lines of code in total, we have completed a stable implementation of the entire iOverlay middleware infrastructure that we have presented. We now evaluate the results of such an implementation, focusing on the baseline correctness, accuracy and performance aspects. For these purposes, we execute iOverlay nodes on a single dual-CPU server with two Pentium III 1GHz processors, 1.5GB of memory, and Linux 2.4.25. The iOverlay engine is compiled with gcc 3.3.3 with the most aggressive optimizations.

We first evaluate the raw message switching performance of iOverlay nodes, especially when they are virtualized nodes on the same server. Since iOverlay nodes are multi-threaded user-level programs, the bottleneck of such switching performance under heavy load is the overhead of context switching among a large number of threads. We create such a load using a chain topology, and we test iOverlay with different number of nodes in the network. Before we deploy an application on the chain topology, we observe that the CPU load is 0.00, which shows that iOverlay does not consume CPU resources without traffic. After we deploy an application that sends back-to-back traffic from one end of the chain to the other as fast as possible, we measure the end-to-end throughput, as well as the total bandwidth in the chain, calculated by the end-to-end throughput multiplied by the number of links. The total bandwidth represents the actual number of messages per second that have been switched or in transit in the network. Fig. 5 shows the iOverlay engine performance in this test, with a chain from two nodes to 32 nodes.

We have two noteworthy observations from this experiment. First, if we compare the two-node total bandwidth of 48.4 MBps and the three-node bandwidth of 46.8 MBps, the overhead of one user-level message switch is only 3.3%. Second, as the number of nodes increases, the overhead of context switching becomes more significant, due to the Linux implementation of POSIX threads using clone(). Still, even with a 32-node configuration, the sustained throughput is still 424 KBps, which is higher than the typical throughput of wide-area connections. This implies that we may potentially deploy dozens of nodes on a single physical node in a local-area or wide-area testbed, making it feasible to test the scalability of new applications in terms of the number of participants. Such performance is simply not achievable if, for example, Java is used rather than C++, or zero message copying is not enforced.



Fig. 5. The raw performance of the iOverlay engine.

In order to verify the correctness of the engine, we have constructed a seven-node topology as in Fig. 6(a), and deployed an application source at node A, so that it sends back-to-back traffic as rapidly as possible to all the remaining receivers. When the number of downstream nodes is more than one, we use the simple algorithm that identical copies of the messages are sent to all downstream nodes. When more than one upstream node exists, no merging is performed.

We first verify per-node total bandwidth emulation and the baseline correctness of message forwarding switches. For this purpose, we have set the buffers of all nodes to be 5 messages at start-up, and specified the per-node available bandwidth on node A as 400 KBps, after deploying the application source using the observer. We observed that the throughput values on all the links have converged to correct values, as shown in Fig. 6(a). At this point, we proceed to set the uplink available bandwidth of node D to 30 KBps, which is much smaller than its current measurements. In a few seconds, the throughput values of all the links except EF and EG have converged to those shown in Fig. 6(b). At node D, both incoming links have converged to 15 KBps due to the flow conservation property (no merging performed); while at node B, since BD is currently the bottleneck and messages have to be copied to both downstreams, both AB and BF are therefore throttled to the same throughput as BD. This demonstrated the accuracy of bandwidth emulation, and the correctness of the basic switching behavior of the engine.



(a) The traffic topology. A is the application data source, with a per-node total available bandwidth of 400 KBps, and copies are made when forwarding to multiple downstream nodes. The measured throughput values are marked at the edges, in KBytes per second.(b) of the transmission of the transmission





Fig. 6. Correctness of the engine: verified with a seven-node topology.

Next, we verify the correctness of node terminations. In the case shown in Fig. 6(c), we terminate node B with the observer. After termination, links AB, BF and BD are closed automatically, while the other nodes are undisturbed, except that the link throughput of CD has converged to 30 KBps. When we continue to terminate node G (Fig. 6(d)), node F is still able to receive application data from A, via the nodes C, D, and E, completely undisturbed.

For some applications, the "back pressure" effect² that bandwidth emulations have on upstream nodes in the topology (as shown in Fig. 6b) is not desirable or realistic. For example, while video streaming and conferencing applications on the overlay may cause such a "back pressure" effect due to its strict latency requirements (and therefore small per-node buffers), data dissemination applications, in general, should allow very large buffers on overlay nodes.

 $^{^2}$ With limited buffer space on a particular node, throughput from upstream nodes eventually converges to the same as the smallest throughput to downstream nodes. This is referred to as the "back pressure" effect.

To show the effects on throughput when large buffers are used, we use the same seven-node topology and the same bandwidth emulations as Fig. 6(b), but set the buffer size to 10000 messages, with each message carrying 5 KB of data. The results with respect to link throughput are shown in Fig. 7(a). In this case, the smaller uplink bandwidth on D has only affected its downstream links, rather than the entire network. We then updated the emulated bandwidth of link EF to 15 KBps, which has *not* affected the link EG, shown in Fig. 7(b). This is because that with large sender thread buffers, the throttling effects on other more capable downstreams are significantly delayed. Of course, when the node buffers do become full after running for a prolonged period of time, the back pressures from full buffers are still effective to decrease link throughputs to those shown in Fig. 6(b). With these experiments, we are confident that iOverlay is able to meet the demands of both delay-sensitive and bandwidth-aggressive applications, by adjusting per-node buffer sizes.



Fig. 7. The effects of bottleneck per-node or per-link available bandwidth: the case of large buffers

3 Case Studies of Application Implementations using iOverlay

We believe that iOverlay is useful to support the rapid implementation of a wide range of applications and distributed algorithms in application overlay networks. We briefly illustrate a few potential research directions to informally justify our observations, and then undertake several case studies to highlight our own experiences of rapidly prototyping new algorithms and ideas using iOverlay as the middleware infrastructure.

3.1 How Useful is iOverlay?

How useful is iOverlay to assist application implementations after all? We now present a few examples.

Content-based networking. Content-based networks, which consist of a collection of client and router nodes in an application-layer overlay, is a *natural fit* to be supported by iOverlay. In content-based networks, messages are not addressed to any specific node; rather, a node advertises *predicates* that define messages of interest that the node intends to receive. The content-based service consists of delivering a message to all the client nodes that advertised predicates matching the message. Any algorithm in content-based networks boils down to one that makes decisions on which nodes should a message be forwarded to, and this may be implemented as a derived class from *iAlgorithm* in iOverlay. The engine passes messages to the content-based decision-making algorithm; and once decisions are made, it forwards the message to all selected downstreams.

Load balancing, rationality and self-interests. There have been recent interests on applying economic or gamebased models to study per-node behavior motivated by self-interests and rationality. In this case, nodes may not be able to relay messages, accept new child nodes in a topology, or give precedence to certain traffic flows, due to the lack of incentives. iOverlay naturally supports such algorithms that seek to engineer and exchange incentives across nodes. For example, an algorithm may perform an elaborate local calculation to determine whether or not a data message should be forwarded, or a new join request should be acknowledged. Since bandwidth and latency measurements are already in place, the load balancing aspects of such algorithms may be straightforwardly evaluated.

Fault tolerance, robustness and availability. Due to the transparent detection of link and node failures in iOverlay, it is easy to design experiments consisting of a certain number of failures, and evaluate the robustness and dependability of proposed algorithms with the presence of failures. For example, the availability of application services may be evaluated by measuring the received throughput at all participating clients, and observe whether the quality of service has been degraded. The faults are all injected by the observer in a controlled fashion, while any possible exceptions are handled by the engine, transparent to the algorithm.

The potential of iOverlay is not limited to these informal discussions. We have envisioned tremendous opportunities of undertaking future research directions using iOverlay as a middleware infrastructure. We now proceed to discuss our own experiences with three case studies.

3.2 Network Coding

The advantages of application-layer overlay networks arise from the fundamental property that overlay nodes, as opposed to lower-layer network elements such as routers and switches, are end systems and have capabilities far beyond basic operations of storing and forwarding. In the first case study, we implement a novel message processing algorithm that performs *network coding* on overlay nodes, using iOverlay. In such an algorithm, messages from multiple incoming streams are coded into one stream using linear codes in the Galois Field (and more specifically, with $GF(2^8)$). We are pleasantly surprised that, with one developer, such a non-trivial task is completed within a few days. We have evaluated the network coding algorithm in the same topologies as those shown in Fig. 6, and we show the performance of the algorithm in Fig. 8.



Fig. 8. Performance of network coding: an iOverlay case study.

Fig. 8(a) shows the results without using network coding. Node A is the data source with per-node bandwidth of 400 KBps, and node D has an uplink bandwidth of 200 KBps. Node A splits its data into two streams sent to B and C, respectively. In this case, B and C are not able to receive both streams, and are referred to as *helper* nodes. Based on iOverlay throughput measurements, the nodes D, E, F and G have received 400, 200, 300, 300 KBps, respectively. In comparison, Fig. 8(b) shows the case where the coding algorithm a + b in GF(2⁸) is applied at node D on the two incoming streams. In this case, the nodes F and G are able to receive both streams aand b by decoding a + b with a, achieving a throughput of 400 KBps. The trade-off, however, is that node E becomes a helper node, in addition to B and C. Our experiences with this case study have demonstrated both the advantages and the trade-offs of applying network coding on overlay nodes. We believe that such an experimentbased evaluation of network coding algorithms is not possible within such a short time frame, if iOverlay is not available as a substrate.

3.3 Construction of Data Dissemination Trees

In this case study, we are interested in the development and evaluation of new algorithms that construct data dissemination multicast trees in overlay networks, particularly in the scenario that the "last-mile" available bandwidth on overlay nodes is the bottleneck. With iOverlay, we have implemented a *node stress* aware algorithm to construct such multicast trees, where *node stress* is defined as the degree of a node in a data dissemination topology divided by the available "last-mile" bandwidth of the node.

The outline of this algorithm is as follows. Periodically, each node in the existing multicast session exchanges node stress information with its parent and child nodes. As a node A joins the multicast session, it first locates a node that is currently in the tree by using one of the utility functions supported in iOverlay, which disseminates a sQuery message. As the message is relayed to the first such node B in the tree, B compares its own node stress with its parent and child nodes. If B itself has the minimum node stress, it responds with an sQueryAck message, so that A becomes a new child of B in the tree. Otherwise, it recursively forwards the message to the node with the minimum node stress (parent or children), until the message reaches the minimum-stress node who sends the acknowledgment.

In order to evaluate such an algorithm in a comparative study, we have also implemented the *all-unicast* and *randomized* tree construction algorithms as control. In the all-unicast algorithm, node B — or any node who is aware of the source of the session (e.g., from the *sAnnounce* message in iOverlay) — simply forwards the *sQuery* to the data source of the session. In the randomized algorithm, node B directly sends the *sQueryAck* acknowledgment to A, and A will join the tree on receiving the first such acknowledgment.

Node	node degree			node stress $(1/100 \text{ KBps})$		
	unicast	random	ns-aware	unicast	random	ns-aware
\mathbf{S}	4	2	2	2.0	1.0	1.0
Α	1	1	3	0.2	0.2	0.6
В	1	1	1	1.0	0.98	0.97
\mathbf{C}	1	2	1	0.5	1.0	0.51
D	1	2	1	1.0	1.98	1.0

Table 3. Tree construction algorithms: node degree and stress

We first experiment with a five-node data dissemination session, shown in Fig. 9, in which the data source is deployed on node S, and nodes A – D joins the session in the order of D, A, C, and B. The figure has been annotated with the per-node available bandwidth, as well as the throughput that we have obtained in our experiments. The node degree and stress are summarized in Table 3. It is very clear that, with respect to endto-end throughput, our new algorithm has the upper hand. We have also observed that the topology of the node stress aware tree is not *optimal*, there may be better trees with respect to throughput. For example, in Fig. 9(g), if D is a child of A rather than S, throughput may be further improved, leaving possibilities for further research. Such experiment-based insights would not be possible without the substrate that iOverlay provides.

In the next experiment, we choose to evaluate the performance and stress tolerance of the node stress aware algorithm in large-scale overlay networks, by deploying it to a total of 81 wide-area nodes in PlanetLab. The per-node available bandwidth has been specified to a uniform distribution of 50 to 200 KBps for all the nodes, with the source node set at 100 KBps. By taking advantage of the deployment scripts in iOverlay, we are able to deploy, run, terminate and collect data from all 81 nodes, with one command for each operation. Fig. 10 shows the North American portion of the wide-area topology after 30 nodes have joined the data dissemination session.

The results we have obtained from these PlanetLab experiments are illustrated in Fig. 11. With respect to node stress, we may observe that the node stress aware algorithm has managed to approach the ideal case (*i.e.*, the vertical line at node stress 20) much better than the other cases. With respect to end-to-end throughput, we may observe that the throughput is much higher with the node stress aware algorithm. Finally, a 10-node topology generated by the node stress aware algorithm implementation is illustrated in Fig. 12 (labeled with per-node IP addresses), while the 81-node topology is shown in Fig. 13.



Fig. 9. Tree construction algorithms: throughput (in KBytes per second).



Fig. 10. The real-time wide-area topology produced by the node stress aware algorithm after 30 nodes have joined (only nodes that reside in North America have been shown, some nodes may reside in the same geographical location).



Fig. 11. Performance of the node stress aware algorithm using 81 wide-area nodes in PlanetLab. (a) end-to-end throughput; and (b) the cumulative distribution of node stress.



Fig. 12. A 10-node topology generated by the node stress aware algorithm.



Fig. 13. The 81-node topology generated by the node stress aware algorithm.

3.4 Service Federation in Service Overlay Networks

In some applications, data messages may need to be transformed (such as media or web data transcoding) by a series of third-party nodes (or *services*) before they reach their destinations. The process of provisioning a complex service by constructing a topology of a selected group of primitive services is known as *service federation* (or *composition*), within what is referred to as *service overlay networks* consisting of instances of primitive services. In order to start a service federation process, a specific *service requirement* needs to be specified, which includes the required primitive services in order to compose the federated service. As a case study, we have designed and implemented a new distributed algorithm, referred to as *sFlow*, to federate complex services that require service requirements in the generic form of directed acyclic graphs, with the aid of iOverlay and over a period of three weeks.

We outline the gist of the algorithm as follows. When a new service is established by the sAssign message from the observer, it locally maintains a *service graph* that represents the producer-consumer relationships among different types of services, and disseminates its existence to all its known hosts via the sAware message. The message is further relayed until an existing service node is reached, which forwards the message to the direct upstream and downstream nodes of the new service in its service graph. When a service federation session is started using the observer, the requirement for the complex service is specified in a sFederate message to the designated source service node. As this message is forwarded, each node applies a local algorithm to select the most bandwidth efficient downstream service node according to the requirement, until the sink service node is reached. The federation process is concluded with the deployment of actual data streams through the selected third-party services. In order to construct a high-quality service topology, the algorithm takes advantage of iOverlay's feature that measures point-to-point throughput to selected known hosts.

We start our experiments by implementing our new algorithm on 16 real-world nodes in PlanetLab, mostly in North America, to construct a service overlay network. The best-quality — *i.e.*, most bandwidth efficient federated service according to a particular service requirement is presented in Fig. 14. Each node in Fig. 14 is labeled with a service identifier assigned to them by the observer. The edges indicate a live service federation session where live data streams are being transmitted. The end-to-end delay of this service session is 934.547 milliseconds, and the last hop average throughput is measured as 69374 bytes per second.



Fig. 14. The constructed complex service in a service overlay network.

During the session, we record detailed statistics on bandwidth measurements and control message overhead on each of the 16 nodes, shown in Fig. 15. In this experiment, the sAware message overhead depends on the number of known hosts of each node, and the overhead of sFederate messages is sufficiently small, compared to that of sAware messages. The per-link and total per-node bandwidth are illustrated in Fig. 15(b) in descending order. Evidently, the overhead incurred by the algorithm is sufficiently small, and seven nodes are left untouched during the entire session of the protocol, since they do not host services or are not involved in the service federation process.

We further experiment with larger-scale service overlay networks, and when multiple service requirements are requested within a short period of time. We are mainly interested in two aspects of the sFlow algorithm. First, the message overhead that the algorithm has incurred, particularly during its phases of disseminating awareness of new services, and of federating existing services to construct the service topology. Second, the end-to-end throughput from the source service to the sink in the constructed service topology.



Fig. 15. Service federation: (a) control message overhead; (b) per-link and per-node bandwidth measurements on each of the overlay nodes. The overlay nodes are sorted by their per-node bandwidth availability.

The overhead of control messages is evaluated with respected to time and different network sizes. Fig. 16 illustrates the sAware message overhead over time, when establishing a 30-node service overlay network, with an average of three new services participating in the network every minute. We may observe that the sAware message overhead starts to significantly decrease after 10 minutes, and is moderate and acceptable over the entire period. Further, Fig.17 shows the results of evaluating the total communication overhead of the control messages as the network size varies. We may observe that the overhead of both types of control messages grows gradually as the network size increases. Especially, the overhead of sFederate messages grows at a slower rate compared to that of the sAware messages. Still, even in a 40-node overlay network, the total control message overhead is less than 1 KB over a 10-minute period, which is equivalent to less than 2 bytes per second.



Fig. 16. The total control message overhead in a 30-node service overlay network, within a period of 22 minutes.

We are further interested in the per-node overhead of control messages during the service federation session, especially when the network load is heavy. Fig. 18 provides more detailed insights with respect to per-node message overhead in a 30-node overlay network. We observe that the overhead caused by the sFederate messages has reached a maximum of 40 KB on three specific nodes. These are the nodes selected by the *observer* as the source service nodes for most the service requirements. Three other nodes have a sFederate message overhead of 17 KB, as they are either selected as the source service node, or executing services involved in most of the federated services. As we may observe, there are 11 nodes with very low overhead with respect to sFederate messages. This indicates that either their services are not required in the service requirements, or they have low available bandwidth, and are therefore not selected.

Finally, we present the end-to-end throughput of the federated complex services generated by the sFlow algorithm, as compared to alternative service composition algorithms. As control, we have implemented the *random* algorithm that randomly chooses a direct downstream node that leads to the corresponding downstream



Fig. 17. The total control message overhead under different network sizes, over a period of 10 minutes, with 50 new service requirements specified and requested every minute.



Fig. 18. The per-node control message overhead within a period of 22 minutes, with 50 new service requirements specified and requested every minute.

service node required in the service requirement. In addition, we have also implemented the *fixed* algorithm, which always chooses the direct downstream node with the highest available bandwidth to the corresponding downstream service in the requirement, rather than randomly choosing downstream nodes. As indicated in Fig. 19, compared to the random and the fixed algorithms, the *sFlow* algorithm consistently produces federated complex services with higher end-to-end throughput, regardless of the network size. Given the complexity of *sFlow*, its rapid implementation demonstrates the effectiveness of iOverlay in supporting realistic algorithm implementations.

4 Related Work

iOverlay was originally motivated by our own experiences of implementing distributed application implementations on overlays, when we have failed to locate a suitable middleware framework for such developments. The idea behind iOverlay originates from the *Flux OSKit* project [10] in operating system design, where a modular set of OS components are designed to be reusable, and to facilitate rapid development of experimental OS kernels. iOverlay provides a reusable set of components in the domain of overlay rather than OS implementations, and seeks to achieve similar design objectives that support rapid prototyping of new overlay-based distributed applications. Particularly, iOverlay is designed to minimize the bar of entry: in order for it to be useful, it is *not* required to have either knowledge about its internals, or extensive system-level programming skills. In addition, iOverlay is also designed to reside at a "higher level" than previous work on user-level network protocol stack implementations (e.g., Alpine [11]), and aims at the development of application-layer rather than network protocols, without the requirements of root privileges.

There exist previous work on using virtual machines (such as VMWare or User-Mode Linux) and support the deployment of full-fledged applications over a virtual network (e.g., [12]), as well as on emulation testbeds and environments to test network protocols in a virtualized and sandboxed environment (e.g., Netbed [8] and ModelNet [13]). In comparison, the objective of iOverlay is to facilitate the development of distributed applications and algorithms at the application layer, and iOverlay assumes the availability of a wide-area network testbed such as PlanetLab. Although iOverlay supports virtualizing multiple overlay nodes on a single physical node,



Fig. 19. End-to-end bandwidth of federated complex services under different network sizes, comparing *sFlow* to the *random* and the *fixed* algorithm.

all implementations are achieved at the user level beyond the abstraction of sockets. iOverlay is designed to be tightly coupled with applications and distributed algorithms, rather than a supporting infrastructure based on either virtual machines or emulation environments.

In particular, ModelNet [13] has introduced a set of ModelNet core nodes that serve as virtualized *kernel-level* packet switches with emulated bandwidth, latency and loss rates. Such kernel-level modifications may not be achievable in wide-area testbeds due to the lack of root privileges. The iOverlay engine, in contrast, implements application-layer message switches, that may be bundled with any new algorithms and deployed in the user space of any UNIX hosts. Thanks to the virtualization of iOverlay nodes, it is not required to have access to a large-scale network in order to experiment with large-scale application topologies.

To the best of our knowledge, there exist two previous papers that present similar objectives to iOverlay. First, the *PLUTO* project [14], an underlay topology service (or routing underlay) for overlay networks, based on PlanetLab. PLUTO is a layer between the overlay algorithms and the network, that exposes topological information to the algorithms. More specifically, it may expose information on connectivity, disjoint end-to-end paths between overlay nodes, as well as the distance between nodes in terms of a particular metric such as latency or router hops. We believe that iOverlay and PLUTO are completely *complementary* with each other, and that it is straightforward for the algorithm to simultaneously take advantage of both architectures. From the viewpoint of PLUTO, iOverlay is simply an overlay application. When it comes to measurement of metrics, iOverlay focuses on measuring the performance of active or potential overlay links, while PLUTO focuses on obtaining insights on the underlay physical topology. From this perspective, iOverlay operates at a higher level than PLUTO does, and PLUTO may be easily integrated into the overall iOverlay middleware architecture.

Second, the *Macedon* project [15] offers a common overlay network API by which any Macedon-created overlay implementation may be used. It features a new language to describe the behavior of an overlay algorithm, from which actual code can be generated using a code generator. As a result, Macedon allows algorithm designers to focus their attention on the algorithm itself, and less on tedious implementation details. Despite the similarities between the design objectives of Macedon and iOverlay, the design principles are drastically different. Macedon attempts to minimize the lines of code to be developed by the algorithm developer, by providing a new language to specify the characteristics of the algorithm. In contrast, iOverlay seeks to maximize the freedom and flexibility when designing new algorithms, by minimizing the API between the middleware and the application. While Macedon is able to support Distributed Hash Table based searching and overlay multicast algorithms, iOverlay is sufficiently generic to accommodate virtually any applications to be deployed on overlay networks, while still encapsulating tedious and common functional components such as message switching, throughput emulation, fault detection and recovery, as well as a centralized debugging facility. Our recent experiences of successfully and rapidly deploying a Windows-based MPEG-4 real-time streaming multicast application on iOverlay have verified our claims.

5 Concluding Remarks

We have been pleasantly surprised at how phenomenally rapidly one can develop fully distributed overlay applications using iOverlay. The evolution of features we have presented have been entirely demand-driven: rather than being designed *a priori*, with inevitably flawed vision of what new applications may need, iOverlay has been constantly refined and augmented, driven by the needs of new application implementations. From this experience, we conclude that research and implementation of overlay applications and algorithms are significantly aided by having reusable, extensible and customizable components that iOverlay provides. As a matter of fact, the burden on the application developer is completely shifted to the core portion of the application-specific algorithm, rather than subtle and mundane details that iOverlay has encapsulated.

We are convinced that the full potential of iOverlay has yet to be realized. First, the library of prefabricated algorithms may be significantly extended, in the form of new classes derived from the base *iAlgorithm* class. These new extensions may become foundations of similar categories of algorithms, which may further simplify the process of new application implementations. Second, the PLUTO routing underlay may be integrated into the iOverlay framework as additional reusable components in the form of libraries, in order to support algorithms that need topological knowledge of the underlying IP topology. Finally, we expect a growing user base of iOverlay clients to drive the continued growth in its performance, generality, power and simplicity, such that the journey from brainstorming sessions to performance evaluations may indeed become enjoyable rather than daunting.

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