

General Data Streaming *

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Abstract

This work presents a new I/O system design and implementation targeted at applications that perform *data streaming*. The approach yields true zero-copy transfers between I/O devices in many instances. We give a general characterization of I/O elements and provide a framework that allows analysis of the potential for zero-copy transfers. Finally, we describe the design, implementation, and performance of a prototype I/O system in a real-time, embeddable, 32-bit operating system that uses the presented analysis to minimize data copying. The prototype uses a *global buffer cache* that allows pass-by-reference transfers of buffers between distinct I/O subsystems.

1 Introduction

A variety of computer applications display a behavior termed *data streaming* where data is repetitively moved from one input/output (I/O) device to another. This work seeks to increase performance by providing mechanisms to tightly couple any pair of I/O devices together within the operating system kernel. In current systems, data copies serve to decouple the transfer characteristics of the I/O devices. However, these data copies have a detrimental effect on performance. The design and implementation of streaming presented here eliminates data copies by passing data between I/O devices by reference.

The proposed design is quite general, applying to the I/O systems of a wide range of operating system types, from embedded systems to personal (PC) class and workstation class computers. The initial implementation has been done in the context of an embeddable, 32-bit, real-time operating system. While predictability is the primary goal for the design of embedded, real-time operating systems, such systems are also generally resource constrained. The elimination of data copies frees both CPU and memory resources that can be used for other purposes. As such, the performance gains achieved using the techniques presented here should be of great value to real-time operating system designers.

This work makes several contributions. 1) A small set of general parameters that may be used to characterize *any* I/O channels to be used as stream endpoints is given. 2) A general analysis of data streaming based on an orthogonal matching of these parameters is provided. 3) A set of algorithms designed to minimize data copies for each of the analytical cases is presented. 4) We discuss mechanisms and policies used to manage a *global* buffer cache used in our prototype implementation. 5) A large set of performance results taken from the prototype operating system that enumerate a representative set of I/O channel pairings and uniformly show throughput improvements are presented.

The rest of the paper is organized as follows. Section 2 provides motivation and context in related work. Section 3 discusses a small set of parameters that can be used to characterize I/O channels for the streaming analysis presented in Section 4. Section 5 presents a set of algorithms that provide optimized streaming based on the analysis in Section 4. Section 6 discusses the im-

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plications of the global buffer cache used for buffer management in the *Roadrunner* operating system prototype. Section 7 provides a performance evaluation of a representative sample of the orthogonal matching of I/O channels. Finally, Section 8 discusses some observations and future work.

2 Related Work

The major motivation for the design and implementation presented in this work is an explosion in the number of different types of peripheral I/O devices that can be attached to modern day computer systems and the desire to support their data transfers as efficiently as possible. In embedded real-time systems, there is an almost infinite array of I/O devices in use as sensor/effector equipment. With respect to PCs and workstations, it was not so long ago that such a system consisted of a CPU, memory, disk, keyboard, display, and perhaps an Ethernet adapter. More recently, cameras, scanners, CD-ROMs and DVDs, microphones, sound cards with their associated amplifiers and/or speakers, all types of pointing devices, joysticks, projectors, modems, ISDN, all types of network adapters ({regular, fast, really fast} Ethernet, FDDI, ATM, Frame Relay, cable modems, etc.), virtual reality hardware, and more are available and in widespread use.

Data streaming can occur at a variety of levels. Architectures such as I₂O⁸ are designed to allow for *peer-to-peer streaming* across a backplane bus. The major advantage is performance but it comes at a price. System complexity and cost are significantly higher since code to manage the organization of data being moved across the backplane must be pushed into the devices and coordinated with the main system. For this reason, much of the work in data streaming has focused on *main memory streaming* where data passes through main memory on its way between peripheral devices.

In main memory streaming, the goal is to copy the data while it is in memory as few times as possible, with zero copy the ultimate goal. Much of the recent work in streaming has focused

on the problem of *cross address space* copies. Pai's work with *IO-lite*,¹³ which is based on Druschel's *fbufs*⁴ mechanism, and Brustolini and Steenkiste's *Genie*¹ are recent examples of the use of page remapping to move data residing in main memory across protection boundaries while avoiding data copies in streaming applications.

Reducing cross address space copies is an important problem to be sure. However, in embedded systems, a significant number of applications do not make use of virtual memory addressing. Furthermore, the cross address space works have not addressed the resulting situation which is that the data transfers between I/O devices are now tightly coupled. The analysis and mechanism presented in this work complement the solutions proposed to address the cross address space problem, i.e. it simply assumes that data is brought into memory from one device and sent out another. It works regardless of how many times the data is moved across address space boundaries.

The primary influence on the I/O system design has been the streaming work performed at UCSD.^{5,6} Fall's proof-of-concept when coupling I/O from similar channels together within a BSD-variant operating system kernel provided performance improvements that are echoed here. However, his approach has been expanded in this work to encompass orthogonal pairing between any I/O elements in the system. The central issue arising from this extension is the need to deal with the dislike characteristics of tightly coupled I/O channels.

3 I/O Channel Parameters

Define a device as any hardware that performs data movement. Typically, this movement is between a peripheral and main memory. A device driver is defined as the lowest level software that controls a device. The driver is generally the only method for accessing a hardware device. Each device can have *input channels*, *output channels*, or both. A device and one channel specify a stream endpoint, either a *source* or a *sink*.

Each device has a value, *B*, associated with

it that corresponds to the *maximum* contiguous block of memory required for a data transfer. Define b to be the amount of data moved during some transfer. If the transfer size associated with a device must be constant and equal to its block size, i.e. $b = B$ for all transfers, the device is said to require *fixed block transfers*. Hard disks and CD-ROM drives are examples of fixed block devices. If the block size can vary up to the maximum, i.e. $b \leq B$, the device is said to allow *variable block transfers*. Network adapters and video cards are examples of variable block devices.

These definitions yield two channel parameters, the *block type*, T , which can take on the values **fix** (fixed) or **var** (variable) and the *maximum block size*, B . Both parameters are static, i.e. they do not vary from transfer to transfer.

A *stream* is defined a half-duplex channel between a source and sink. The source provides a sequence of M transfers where each transfer has a size b^{in} . The stream delivers to the sink a sequence of N transfers each with size b^{out} .

4 Streaming Analysis

An analysis of streaming performance must be carried out with respect to a goal. As in most areas in software design, a time versus space trade-off is present. On one end of the spectrum, the throughput of a stream can be maximized by eliminating data copies in main memory where possible. At the other end, system resources can be minimized at the expense of throughput by performing data copies. Consider an analysis using the two static parameters to maximize throughput.

Figure 1 gives the list of all possible combinations that result when the parameters of source and sink channels are coupled. The input and output channels each have a block type and maximum block size. These twelve cases can be reduced to four that are used to optimize throughput for streaming.

Case 1 $(T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{fix} \wedge B^{in} = B^{out}) \vee$
 $(T^{out} = \mathbf{var} \wedge B^{in} \leq B^{out})$

case 1	$T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{fix} \wedge B^{in} = B^{out}$
	$T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{var} \wedge B^{in} = B^{out}$
	$T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{var} \wedge B^{in} < B^{out}$
	$T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{var} \wedge B^{in} = B^{out}$
	$T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{var} \wedge B^{in} < B^{out}$
case 2	$T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{var} \wedge B^{in} > B^{out}$
	$T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{var} \wedge B^{in} > B^{out}$
case 3	$T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{fix} \wedge B^{in} > B^{out}$
case 4	$T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{fix} \wedge B^{in} < B^{out}$
	$T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{fix} \wedge B^{in} = B^{out}$
	$T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{fix} \wedge B^{in} < B^{out}$
	$T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{fix} \wedge B^{in} > B^{out}$

Table 1: Static Parameter Combinations

In this first case, either both the source and sink have the same fixed block type or the output block type is variable and the input block size is smaller than the output block size. Input blocks can be output directly. We have $N = M$ and no data copies are required since the input buffer can always be output directly. An illustration of this streaming case is given in case 1 of Figure 1.

Case 2 $T^{out} = \mathbf{var} \wedge B^{in} > B^{out}$

In this case, the output block type is variable and the input block size is larger than the output block size. The stream divides the input block into $\lfloor \frac{B^{in}}{B^{out}} \rfloor$ output blocks of size B^{out} and a single output block holding the remainder, $b^{out} = B^{in} - \lfloor \frac{B^{in}}{B^{out}} \rfloor B^{out}$. Note that the solid lines indicate that *reference buffers* are used to generate multiple output blocks for each input block. Reference buffers are discussed in the section describing the common buffer representation. An illustration of this streaming case is given in case 2 of Figure 1.

Case 3 $T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{fix} \wedge B^{in} > B^{out}$

In this case, both the input and output block types are fixed and the input block size is greater than the output block size. The stream divides input blocks into smaller blocks for output. The number of output blocks required to empty an input block is

$\frac{B^{in}}{B^{out}}$ which may be non-integral. Since output blocks must be filled, they may contain fractions of input blocks. For every $\frac{L}{B^{out}}$ output blocks, $\frac{L}{B^{in}} - 1$ copies of input data to an output block must be performed. The segments of data framed by dashed lines must be copied from an input buffer to an output buffer. The algorithm presented later does not exactly match the mechanism presented here. An illustration of this streaming case is given in case 3 of Figure 1.

Case 4 ($T^{in} = \mathbf{fix} \wedge T^{out} = \mathbf{fix} \wedge B^{in} < B^{out}$) \vee ($T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{fix}$)

In the final case, either both the input and output block types are fixed and the input block size is smaller than the output block type or the input block type is variable and the output block type is fixed. In both situations, blocks must be aggregated to fill the larger output blocks. The number of input blocks that fill an output block is given by $\frac{B^{out}}{B^{in}}$. If this result is non-integral, input blocks must be split.

Define $L = lcm(B^{in}, B^{out})$ to the least common multiple of B^{in} and B^{out} . Observe that for every $\frac{L}{B^{in}}$ input blocks, $\frac{L}{B^{out}}$ blocks will be output. Furthermore, for every $\frac{L}{B^{out}}$ output blocks, *one* copy of data from an input block to an output block can be avoided. An illustration of this streaming case is given in case 4 of Figure 1.

This analysis does not apply to the second part of this case, where $T^{in} = \mathbf{var} \wedge T^{out} = \mathbf{fix}$. The analysis presented here is similar to that of case 3 in that it represents the best analytical case. The algorithm presented later does not adhere to this analysis, but does handle both parts.

5 Streaming Algorithms

A set of algorithms whose goal is to provide the efficient streaming presented in the previous section is now given. These algorithms make assumptions

Routine	Description
<code>open()</code>	Open a file
<code>close()</code>	Close a previously opened file
<code>ctl()</code>	Execute a control function for a file
<code>get()</code>	Get a character from a file
<code>put()</code>	Put a character to a file
<code>read()</code>	Read block of data from a file
<code>write()</code>	Write block of data to a file

Table 2: **fs** Layer Routines

about the design of the I/O system. First, access to all I/O elements is performed through a common applications programming interface (API). Second, a common buffer representation is used by all system I/O elements. The common API, common buffering and streaming algorithms have all been implemented in the *Roadrunner* operating system.^{12,3}

5.1 Common API

Figure 2 illustrates the architecture of the I/O system on which the streaming algorithms are implemented. The architecture provides a common interface to all I/O elements. This is accomplished using the **fs** layer. Table 2 lists some of the common routines provided by the **fs** layer that are available for moving data to and from all I/O elements.

This architecture is based on work from several sources. The use of the file system name space to address various elements of the system is reminiscent of Plan 9/Inferno.¹⁴ The internal implementation is based primarily on the **vnodes**^{10,15,16} architecture present in 4.4BSD¹¹ (among others) and the stackable architectures proposed by Heidemann and Popek.⁷

5.2 Common Buffers

In the analysis, there were several places where buffers are assumed to be passed by reference. This is accomplished by using a common buffer representation among all I/O elements. Figure 3 illustrates the design of the common buffer rep-

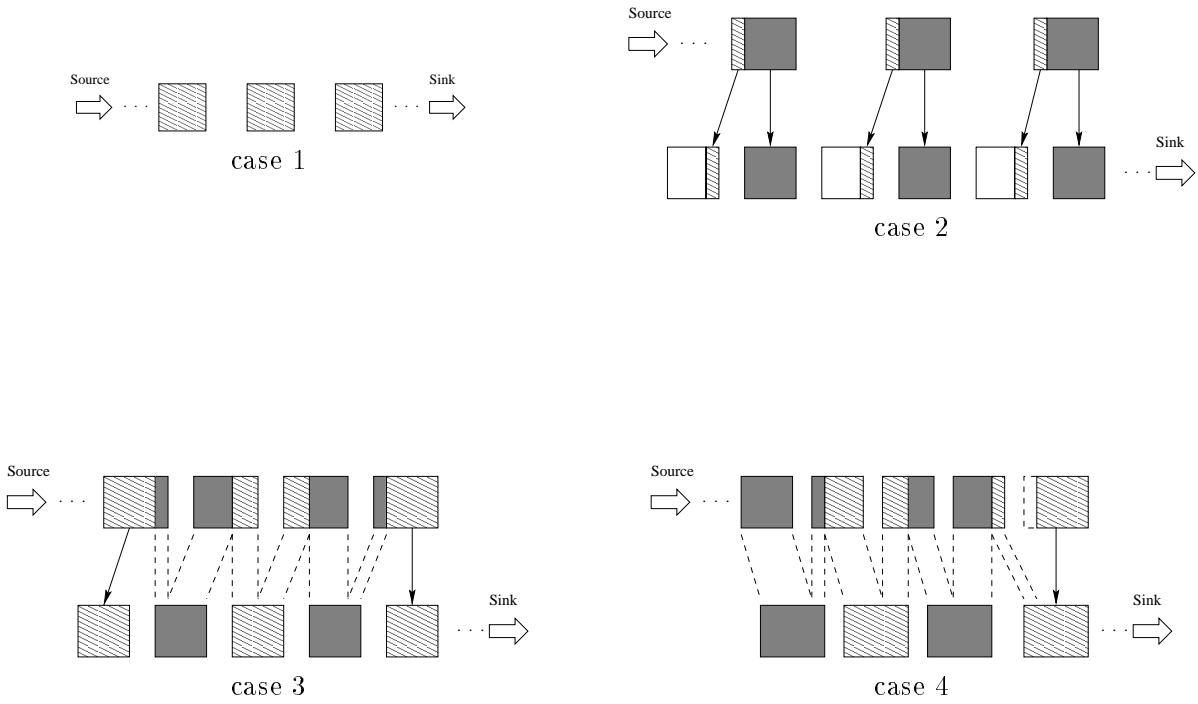


Figure 1: The set of streaming cases derived from the static parameter combinations.

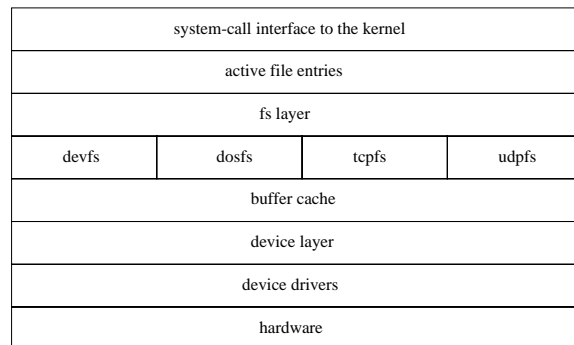


Figure 2: The *Roadrunner* I/O system architecture.

Routine	Description
<code>bread()</code>	Retrieve cached buffer
<code>bput()</code>	Enter buffer in cache
<code>balloc()</code>	Blocking get buffer descriptor <i>only</i>
<code>_balloc()</code>	Non-blocking get buffer descriptor <i>only</i>
<code>bget()</code>	Blocking get buffer with data block
<code>_bget()</code>	Non-blocking get buffer with data block
<code>bfree()</code>	Release buffer to the free list
<code>bre1()</code>	Enter buffer in cache or release to free list

Table 3: Common Buffer Management Routines

resentation. Buffers have two parts, a descriptor and an optional data block. The fields of the descriptor are partitioned into those used for referencing data in a data block and those used for caching. The `data` field points at the beginning of an allocated data block. The `size` field contains the length of the allocated block. The `start` and `len` fields mark where useful data is contained in the allocated block. The `pos` field is used to provide an offset into the start of the useful data. The `devno` (or device number), and `blkno` (or block number), fields are used for caching. These two fields are combined in a hash function that allows quick insertion and retrieval of buffers into and out of hash table managed buffer pools.

Table 3 gives the routines used by I/O elements to perform buffer management. I/O elements use *move semantics* when managing buffers, i.e. a buffer only resides in the global cache when it is *not* owned by some file descriptor. Furthermore, there are two conventions that must be followed. First, for source devices, from which data is read, a buffer where data is to be placed during an input transfer must be specified, however, *the actual buffer into which data is placed need not be the one specified*. The only requirement, if data is placed in a different buffer, is an indication that this has occurred, i. e. a reference to the actual buffer into which data was read. This assumption enables *all* devices to write data directly into common buffers. Second, it is assumed that all sink devices perform output transfers directly from the specified buffer and that they discard the buffers when the output transfer is complete, i.e. the

caller may not reuse the buffer after the output transfer.

5.3 The Overlay Algorithm

The algorithm presented here derives its name from the conceptual idea that it is “overlying” a stream of input buffers onto a stream of output buffers. The overall structure of the algorithm is given in Figure 4. Conceptually, this algorithm can be executed statically or dynamically. In the former case, the first step is performed once, prior to execution of the stream transfers. In the latter case, the execution of stream transfers is halted periodically in order to re-execute the first step of the algorithm. More discussion on dynamic execution of the algorithm is presented later.

5.3.1 Obtaining Channel Parameters

To obtain the parameters for the input and output channels that the algorithm uses to determine which stream case will be executed, a series of queries are executed for each file descriptor. These queries are illustrated in Figure 5.

5.3.2 Case 1

This simplest cases occur when either the input and output block types are fixed and the input and output buffer sizes are equal or the output block type is variable, and the input buffer size is less or equal to the output buffer size. In these instances, input buffers can be output directly using the simple algorithm in Figure 6. This algorithm implements exactly the behavior specified in case 1 of Figure 1. The system call, `read(in, b)`, reads data from the input channel, `in`, into the buffer, `b`. The system call, `write(out, b)`, writes data to the output channel, `out`, using the buffer, `b`.¹ Since input buffers are output directly, no data copies are required.

¹The specifications for `read()` and `write()` do not match exactly those in the actual *Roadrunner* implementation. Their use has been condensed for clarity in this presentation. Also note that the presentation of this and all the algorithm cases leaves a large number of details out, e.g. error handling, also for the sake of clarity.

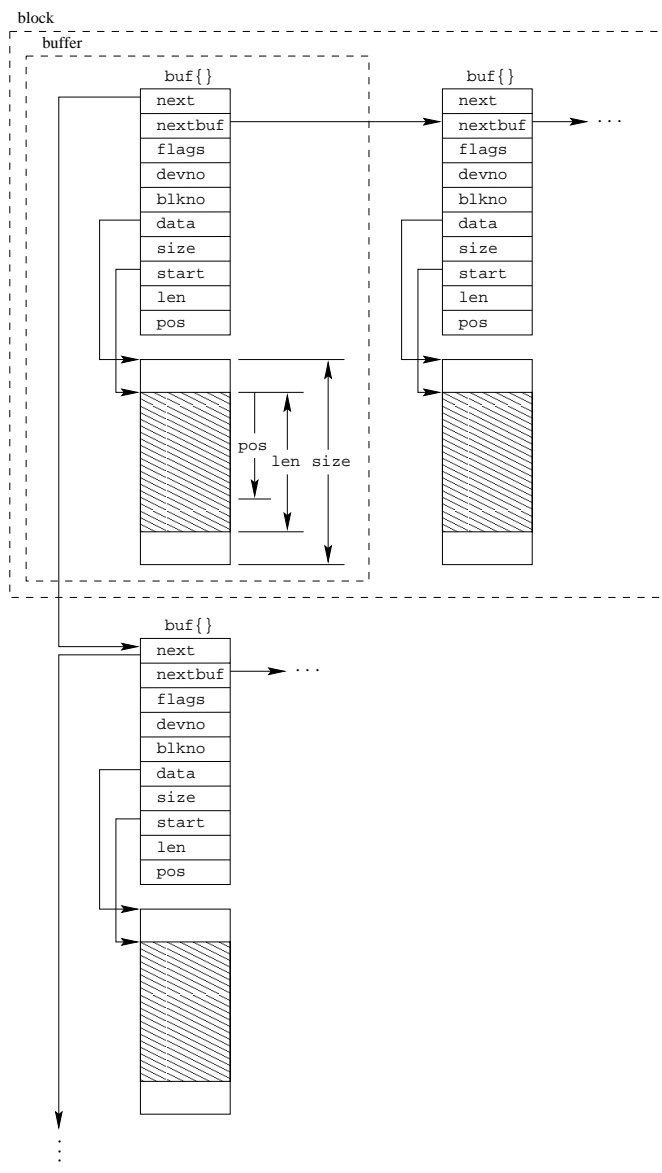


Figure 3: Common buffer representation.

```

int stream(int in, int out)
{
    int infilesize;
    int inblktype, outblktype;
    int inbufsize, outbufsize;

    Get channel paramters

    if ((inblktype == FIX && outblktype == FIX && inbufsize == outbufsize) ||
        (outblktype == VAR && inbufsize <= outbufsize)) {

        Case 1 stream

    } else if (outblktype == VAR && inbufsize > outbufsize) {

        Case 2 stream

    } else if (inblktype == FIX && outblktype == FIX && inbufsize > outbufsize) {

        Case 3 stream

    } else if ((inblktype == FIX && outblktype == FIX && inbufsize < outbufsize) ||
                (inblktype == VAR && outblktype == FIX)) {

        Case 4 stream

    }
}

```

Figure 4: The overall structure of the overlay algorithm

```

ctl(in, GET_FILE_SIZE, &infilesize);
ctl(in, GET_BLOCK_TYPE, &inblktype);
ctl(out, GET_BLOCK_TYPE, &outblktype);
ctl(in, GET_BUF_SIZE, &inbufsize);
ctl(out, GET_BUF_SIZE, &outbufsize);

```

Figure 5: Obtaining channel parameters

```

int n;
for (n = 0; n < infilesize;) {
    b = bget(inbufsize);
    read(in, b);
    write(out, b);
    n += inbufsize;
}

```

Figure 6: The case 1 stream algorithm

5.3.3 Case 2

This case occurs when the output block type is variable and the input block size is greater than the output block size. In these instances, the algorithm given in Figure 7 can be executed. It's operation matches exactly the analysis given in case 2 of Figure 1.

The algorithm repetitively fetches a buffer, `bin`, from the input channel and progresses through its data sending it to the output channel. The buffer, `bout` is a *reference buffer*, i.e. its descriptor is used to progress through areas of data in `bin`. The maximum block size for `bout` is used until the end of the data has been reached in `bin`. In this last output transfer, the amount of data sent will be `inbufsize mod outbufsize`. Since all output buffers are passed by reference, no data copies are required.

5.3.4 Case 3

This case occurs when the input and output block types are both fixed and the input buffer size is greater than the output buffer size. The algorithm given in Figure 8 is based on the analysis in case 3 of Figure 1 but does not match its operation exactly.

The key observation made when developing this algorithm is that it is very likely that the input buffer size is a multiple of the output buffer size. If this is so, it is desirable to have the algorithm behave similarly to case 2, i.e. passing partitions of the input blocks by reference. Only when the input block size is not a multiple of the output block size should copying be performed. Fortunately, the case where a multiple is present is common. An example is when a file is copied between disk file systems that utilize a different block size.

The algorithm again repetitively reads blocks from input channel. In this algorithm case, however, a separate inner `for` loop is used to progress through the data in each input block. There are four cases that occur as it progresses through data in each input block. They are defined by the guards on the inner and outer `if` statements.

The first two cases occur when no output block is currently being filled. In the first case, the amount of data in the input block is greater than or equal to the size of an output block. A reference buffer is defined and sent out the output channel. The second case occurs when the amount of data left in the input block is less than the size of an output buffer. The remainder of the input block is copied into a new output buffer and the algorithm proceeds to the next input block.

The second two cases occur when there is an as yet unfilled output block that was carried over. This can happen as a result of the second or fourth cases. In the third case, there is enough data in the input block to fill the remainder of the output block. The data is copied and the filled output block is sent to the output channel. The fourth case occurs when the data remaining in the input block will not fill the carried over output buffer. This action taken is similar to case 2.

If the input buffer size is a multiple of the output buffer size, the first case is the only one that is ever executed and no data copies are required. If however, the input buffer size is not a multiple of the output buffer size, the situation in case 2 of Figure 1 occurs, i.e. a copy is avoided only when the least common multiple of the two buffer sizes occurs.

5.3.5 Case 4

This case occurs when the input and output block types are both fixed and the input buffer size is less than the output buffer size or the input block type is variable and the output block type is fixed. The algorithm given in Figure 9 is based on the analysis in case 4 of Figure 1 but does not match its operation exactly.

In this case, one copy must always be performed. For the first part of this case, it is necessary to aggregate smaller input blocks into larger output blocks. In the second part, since it is impractical to expect that the the data in variable input blocks will have repetitive sizes, all the variable input data must be aggregated into the fixed output buffers.

```

int n;
for (n = 0; n < infilesize;) {
    if (bin == NULL) {
        bin = bget(inbufsize);
        read(in, bin);
    }
    bout = balloc();
    bout->start = bin->start + bin->pos;
    if (bin->len - bin->pos >= outbufsize) {
        bout->len = outbufsize;
        n += outbufsize;
        bin->pos += outbufsize;
    } else {
        bout->len = bin->len - bin->pos;
        n += bin->len - bin->pos;
        brel(bin);
        bin = NULL;
    }
    write(out, bout);
}

```

Figure 7: The case 2 stream algorithm

This case also proceeds by looping over input buffers. During each iteration, however, it is possible that an input or output buffer may have been carried over. If not, the required buffer is allocated. If there is enough input data to fill the output buffer, the data is copied and the buffer is sent to the output channel. The input buffer carries over. Otherwise, the output data is copied and the output buffer is carried over.

5.4 Dynamic Conditions

The goal of the static analysis just presented is to increase throughput by reducing data copies. This goal is addressed by the use of tailored algorithms based on static I/O channel parameters. However, there are also any number of dynamic conditions that can have an affect on a given streams performance.

Consider the following situation that can arise on some streams. In many cases, output transfer times are nominally shorter than data copy times. However, during transients, output transfer times can increase beyond the duration for data copies. When this occurs, copies can be used to fill otherwise partially full variable output blocks to reduce

throughput degradation. The key observation is that the streaming cases presented for static conditions represent a spectrum in the use of data copies. The first two cases never perform data copies. The third case uses copies only when the one of the input or output buffer sizes is not a factor of the other. The fourth case always makes one data copy in order to fill the output block. When output times increase during a transient, the overlay algorithm can force the stream to the last case, which fills output buffers using a single data copy. This has the effect of reducing the overall number output transfers. When the transient subsides, the algorithm can return to its nominal case processing.

It only makes sense to perform this action when the output blocks are partially filled and this only occurs when the case 1 algorithm is executing because $T^{out} = \mathbf{var}$ and $B^{out} > B^{in}$.

Consider as an example, a streaming transfer of a file to a network connection. Such streams are the norm in web servers. In this stream we have $T^{in} = \mathbf{fix}$, $T^{out} = \mathbf{var}$, and assume that $B^{out} = 2B^{in}$. Under the case 1 algorithm, $N = M$ output transfers of size B^{out} would be performed, essentially wasting half the space in each output

```

int n;
for (n = 0; n < infilesize;) {
    bin = bget(inbufsize);
    read(in, bin);
    while (bin->pos < bin->len)
        if (bout == NULL) {
            if (bin->len - bin->pos >= outbufsize) {
                bout = balloc();
                bout->start = bin->start + bin->pos;
                bout->len = outbufsize;
                write(out, bout);
                n += outbufsize;
                bin->pos += outbufsize;
            } else {
                bout = bget(outbufsize);
                bcopy(bout->start,
                    bin->start + bin->pos,
                    bin->len - bin->pos);
                bout->len = outbufsize;
                bout->pos = bin->len - bin->pos;
                n += bin->len - bin->pos;
                break;
            }
        } else {
            if (bin->len - bin->pos >= bout->len - bout->pos) {
                bcopy(bout->start + bout->pos,
                    bin->start + bin->pos,
                    bout->len - bout->pos);
                n += bout->len - bout->pos;
                bin->pos += bout->len - bout->pos;
                write(out, bout);
            } else {
                bcopy(bout->start + bout->pos,
                    bin->start + bin->pos,
                    bin->len - bin->pos);
                bout->pos += bin->len - bin->pos;
                n += bin->len - bin->pos;
                break;
            }
        }
    }
    brel(bin);
}

```

Figure 8: The case 3 stream algorithm

```

int n;
for (n = 0; n < infilesize;) {
    if (bin == NULL) {
        bin = bget(inbufsize);
        read(in, bin);
    }
    if (bout == NULL) {
        bout = bget(outbufsize);
        bout->len = outbufsize;
    }
    if (bin->len - bin->pos >= bout->len - bout->pos) {
        bcopy(bout->start + bout->pos,
              bin->start + bin->pos,
              bout->len - bout->pos);
        n += bout->len - bout->pos;
        bin->pos += bout->len - bout->pos;
        write(out, bout);
    } else {
        bcopy(bout->start + bout->pos,
              bin->start + bin->pos,
              bin->len - bin->pos);
        n += bin->len - bin->pos;
        bout->pos += bin->len - bin->pos;
        brel(bin);
        bin = NULL;
    }
}

```

Figure 9: The case 4 stream algorithm

buffer.

If we assume the source file resides in the cache then input transfer times are trivial. Assume a value of $20 \mu s$ for all input transfers. If a constant output transfer time is assumed to be $100 \mu s$, the total time to execute this stream is given as $M(20 + 100) = 120M$. Now consider a simple step function associated with the output transfer times, t_o (in μs),

$$t_o = \begin{cases} 1 \dots \frac{N}{2} - 1 & : 100 \\ \frac{N}{2} \dots N & : 400 \end{cases}$$

Such a function might represent a change in route occurring because a router on the network path to the web client goes down. The execution time of the stream under this circumstance increases to $20M + \frac{M}{2}(100 + 400) = \frac{M}{2}(40 + 100 + 400) = 270M$.

Note however, it is possible to fill the output blocks when the transition to the higher output transfer times occurs. If a constant block copy

time of $200 \mu s$ is assumed, the resulting execution time is given as $20M + \frac{100M}{2} + \frac{400M}{4} = \frac{M}{4}(80 + 200 + 400) = 170M$. While this is still a degradation with respect to the original execution time, it is a significant reduction in the degradation compared to continuing with nominal case 1 processing.

A mechanism to perform the monitoring or transfer times and to effect case switching in the overlay algorithm is currently being designed and implemented. Performance results and optimizations based on other dynamic conditions are the subject of future research.

6 Global Buffer Cache Management

The *Roadrunner* system uses a global buffer cache. This differs from traditional systems where each class of I/O element maintains a local buffer cache, e.g. BSD systems where network and file

subsystems maintain distinct caches. This difference is fundamental. Not only does it force us to revisit various design decisions regarding the management of buffer pools, both mechanism and policy, it also has implications for potential utilization.

Consider a conceptual model of all the separate buffer pools in some current system as being partitions of a single “virtual global” cache. The current system can be said to partition its global cache based on subsystems, e.g. partitions for the network and file subsystems. We call this type of partitioning, *subsystem partitioning*. In *Roadrunner*, the partitioning of the global cache is done based on application requirements. When an application requires buffers, it draws from the global pool, based on priority (since the high priority thread making requests can dominate), until it is satisfied. No consideration is given to whether a particular subsystem might need to reserve some number of buffers to itself. We call this type of partitioning, *application-based partitioning*.

The potential advantage of this new approach is that the use of buffers can be better tailored to application demands. The disadvantage is the potential for an application to dominate the resource. Note however that embedded, real-time systems, the potential for domination may not be considered a disadvantage.

Of course, the *Roadrunner* prototype system is not quite fit either model exactly. The actual global pool consists of a set of descriptors and several pools of data blocks of various sizes. When the system is initialized, the buffer pool must allocate some data blocks. This is currently accomplished with a configuration file that specifies sets of data blocks with different sizes. An administrator must set up this file to reflect the block sizes of semantic abstractions (subsystems) that will be mounted on that system.²

²Initialization of the buffer pool is tricky since it must generally occur before any of the subsystems themselves are mounted and initialized. One current implementation effort is directed at looking for ways to do the allocation of data blocks with sizing automatically at startup. Difficulty allocating block sizes is further compounded by the potential for dynamic changes to mounted subsystems.

We have observed that more than one mounted subsystem may use data blocks of the same size. When this occurs, these subsystems display application-based partitioning within the pool of data blocks with the given size. However, subsystems that make requests from data block pools with different sizes display subsystem partitioning.

Experience with the *Roadrunner* prototype has shown that domination of the buffer pool does occur under exceptional conditions. The most common occurrence observed to date is an exceptional condition in the network protocol stack causing all the data blocks for the pool that fit the network packet size request to be consumed and placed in an interface queue. The reality is that some mechanism for protection must be put in place to disallow a single thread from draining the buffer pool of data blocks.

The hybrid partitioning of buffer pools by subsystem in has served to assist in limiting the domination by active entities.³ Reference counts are also used in various places to prevent the use of too many buffer resources.

The use of move semantics, where buffers are not tracked by the buffer management subsystem at all times, makes this problem difficult. The current *Roadrunner* system simply limits the length of buffer queues. When a queue insertion occurs, the length is checked and if it exceeds a threshold, the buffer is freed. Operationally, this works well since the domination problem has surfaced only with respect to network input and freeing network packets in buffers can be compensated for by the protocols themselves. This simple mechanism will not be sufficient for other I/O subsystems and consideration of more comprehensive mechanisms is a subject for future research.

³An active entity in this context is an operating system element capable of “owning” a buffer. Threads and device drivers are the active entities in *Roadrunner*. UNIX processes also qualify as active entities under this definition.

7 Performance

Performance measurements, especially apples to apples comparisons are difficult in this work for a number of reasons. Probably most important is that we are not aware of any other systems with an I/O subsystem designed to do generalized coupling of I/O elements within the kernel in the manner being described here. The approach that has been taken is to provide two sets of measurements for a subset of an orthogonal coupling of a representative set of I/O endpoints. The first set *simulates* the environment found in most current systems, i.e. it uses `read()/write()` system calls to perform streaming transfers. The second set uses the appropriate streaming algorithm.

All of the performance measurements presented were made on the same machine, an IBM PC/350, that includes a 75 MHz Pentium with 256 Kbyte second level cache, 16 Mbytes of memory, an 800 Mbyte EIDE hard disk, and an SMC EtherPower 10BaseT PCI Ethernet adapter. While this machine is somewhat dated compared to those currently available, it is important to note that the absolute figures in these measurements are not as important as the relative results, those using the streaming interface versus those using the traditional `read()/write()` interface.

A list of the file systems and devices used as stream endpoints is given in Figure 4. The `/dev/null` device is accessed through the `devfs` file system. It provides a variable block type with a maximum block size of 4096 bytes. The UDP and TCP file systems provides endpoints utilizing the DEC tulip based PCI Ethernet adapter. The UDP endpoint uses a variable block type with a maximum of 1446 bytes⁴. The TCP endpoint uses a variable block type with a maximum of 4096 bytes. Two DOS compatible file systems are used that reside on the same hard disk in different partitions. Both have a fixed block type with one having a block size of 2048 bytes and the other a block size of 4096 bytes.

⁴This is the maximum Ethernet frame size minus length of the various Internet protocol headers.

Table 5 shows how these endpoints can be combined for streaming. There are two types of cells in this matrix. Those cells that contain only a single number give the streaming case exercised when the corresponding endpoints are combined. For example, when the DOS file system with block size 2048 bytes is used a source and the UDP endpoint is used a sink, the case 2 streaming algorithm applies. Those cells that contain two numbers separated by a slash give the streaming case and the figure number of measurements presented for that endpoint combination. For example, when the DOS file system with block size 2048 bytes is used as a source and the TCP endpoint is used as a sink, the case 1 streaming algorithm applies and measurements taken for this case are presented in Figure 19.

Figure 10 provides a reference point intended to illustrate the bandwidth to main memory. The `bcopy()` used was hand coded assembly that makes use of the Intel `rep movs` instructions⁹ and the result shows the age of the machines in use. The result is linear in the number of bytes copied giving a bandwidth of approximately 66 Mbps.

Figure 11 illustrates the best conceptual speedups for streaming over the `read()/write()` interface. The `/dev/null` device simply returns the buffer provided to it in the `nullread()` routine and frees the buffer in the `nullwrite()` routine.⁵ These two routines are called directly by the `stream()` and the buffer is passed between them by reference. In contrast, when the the `fs` layer `read()` and `write()` routines are used, the buffers are copied out or into an application buffer, respectively. The use of streaming results in a throughput of 164 Mbps and the use of `read()/write()` yields a throughput of 31 Mbps. The streaming transfer provides a 430% improvement.

The graphs in Figures 12 and 13 provide a baseline for network performance using streams. Data is moved from `/dev/null` to a UDP protocol port or TCP connection, respectively.

⁵`nullread()` and `nullwrite()` are routines provided by the `nullfs` semantics abstraction to support the `fs` layer `read()` and `write()` routines, respectively.

Endpoint	File System/Device	Block Type	Max. Size
1	devfs/null	var	4096
2	udpfs/tulip ethernet (PCI)	var	1446
3	tcpfs/tulip ethernet (PCI)	var	4096
4	dosfs/wd EIDE hard disk	fix	2048
5	dosfs/wd EIDE hard disk	fix	4096

Table 4: Stream endpoints used in measurements

		Destination				
		1	2	3	4	5
Source	1	1/11	2/12	1/13	4/15	4
	2	1	4	1	4	4
	3	1	4	4	4	4
	4	1/14	2	1/19	1/16	4/18
	5	1	2	1	3/17	1

Table 5: Endpoint combinations for measurements. Cells containing a single number give the streaming case used for the endpoint combination. Cells containing two numbers separated by a slash give the streaming case and the figure number for measurements taken for the endpoint combination.

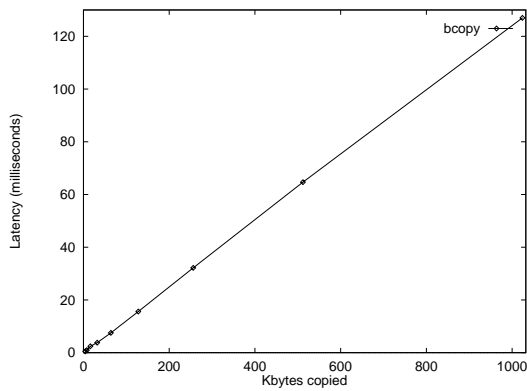


Figure 10: bcopy()

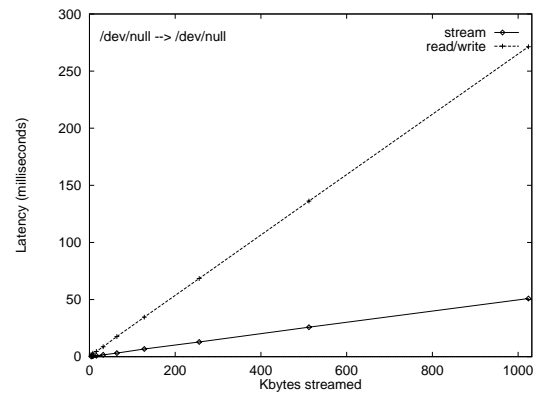


Figure 11: Null stream

The mechanism is similar, the `stream()` reads from the `nullread()` routine and writes to the `udpwrite()` or `tcpfswrite` routine passing buffers by reference. The `read()/write()` measurements are taken using the `fs` layer interface. The best case observed speedup for UDP observed in Figure 12 is 215%, and the best case observation for TCP in Figure 13 is 132%. Not surprisingly, the best case speedup for TCP comes when the amount of data streamed matches the send buffer size.

Figure 14 shows the baseline performance of the IDE disk in the test machine. Because disk I/O transfers take so long compared to memory movements, speedups are heavily dependent on the use of caching. The improvements are good when the cache is in play, i.e. the file, or portions of it, reside in the cache. The improvements are modest when disk I/O dominates the transfers. Consider the points for a 1 MB cache when a 512 KB file is read. The streaming transfers are accomplished at a rate of 30.5 Mbps while the `read()/write()` transfers yield a throughput of 22.6 Mbps. Streaming yields a 35% improvement in this case. However, if the same cache size is used to transfer a 1 MB file, the best throughput for a streaming transfer is 1.9 Mbps versus 1.7 Mbps for the `read()/write()` transfers. Streaming yields only a 6% improvement here.

Figure 15 illustrates the effect of the overall cache size for a given data block size during file writes. Since delayed writes are occurring, disk operations do not come into play until the file size exceeds the cache size. When the file fits into the cache, speedups are in the neighborhood of 79%. When the file size exceeds the cache size, `bget()` calls must flush buffers, i.e. write the dirty contents to disk, before filling them with new data. This results in the significantly reduced speedup of approximately 5%.

The remaining sets of measurements provide results for applications that make use of the streaming algorithms. Figures 16, 17, and 18 give speedups for execution of the file copy program, `cp` and Figure 19 gives the speedups associated with the *Roadrunner* web server, `httpd`.

In Figure 16, a copy from one place to another on the same file system is presented. Improvements here range from 33% for the case of a 1 MB file and 512 KB cache to 75% for a 512 KB file and 1 MB cache to 77% for a 1 MB file and 1 MB cache. In Figure 17, a file is copied from a file system with a buffer size of 4096 bytes to a file system with a buffer size of 2048 bytes. The performance improvements for the input buffer greater than the output buffer are the best of any of the algorithm cases. Consider the data points for the 256 KB cache size. The `read()/write()` performance is 1.25 Mbps for a 256 KB file, 608 Kbps for a 512 KB file, and 535 Kbps for a 1 MB file. The stream performance is 1.4 Mbps for a 256 KB file, 1.29 Mbps for a 512 KB file, and 1.2 Mbps for a 1 MB file. These throughputs yield 12.5%, 87%, and 124% speedups, respectively. In Figure 18, a file is copied from a file system with a buffer size of 2048 bytes to a file system with a buffer size of 4096 bytes. The speedups here, while uniform, are modest. This is to be expected since this stream uses case 4, where copies are always required.

There is an interesting dynamic here with the cached blocks in Figures 16 and 17. When `read()/write()` is used, a block is brought into the cache for the input channel and another block is entered in the cache for the output channel. When streaming is used, a single block is entered in the cache. The block is originally brought in for the input channel but is then passed by reference to the output channel. When this happens, the input channel reference goes away. This halves the cache utilization for the stream and explains why the performance of the 1 MB file and 1 MB cache hasn't yet dropped off as in the case of the 1 MB file and 512 KB cache. The tradeoff is that future references to the same input block will require another disk reference.

The speedups in Figure 19 vary depending on the retrieved file size. A best case speedup of 505% has been observed when a file that is less than or equal to the TCP send buffer size resides in the global cache. More modest improvements of 26% are observed for large files. It should be

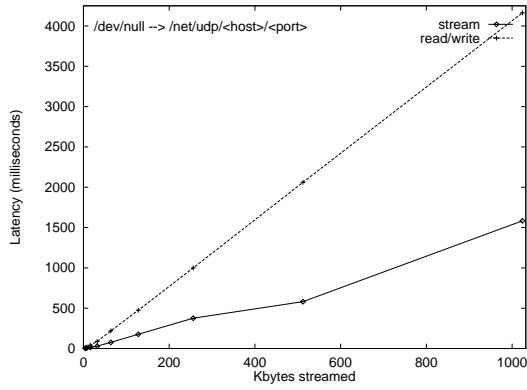


Figure 12: UDP stream

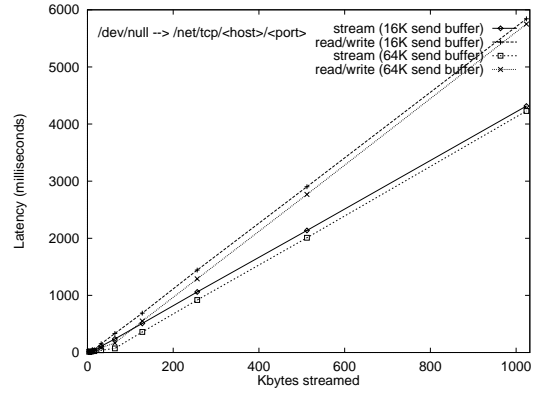


Figure 13: TCP stream

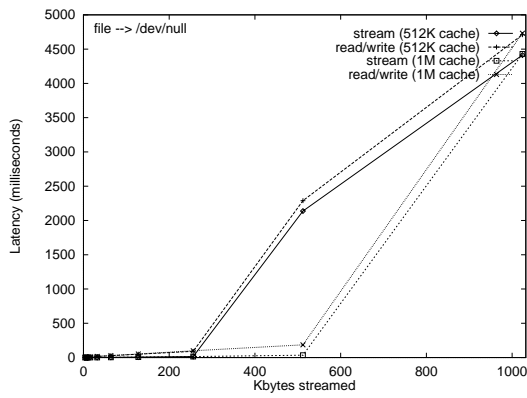


Figure 14: File read stream

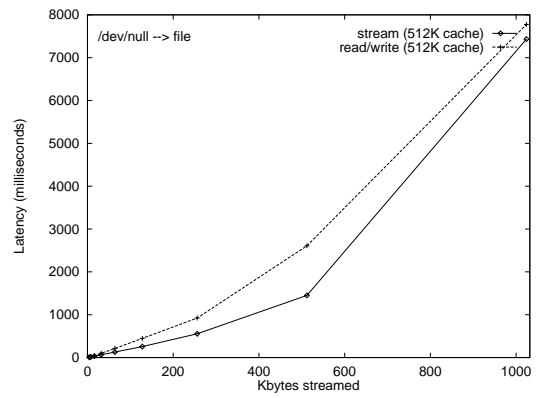


Figure 15: File write stream

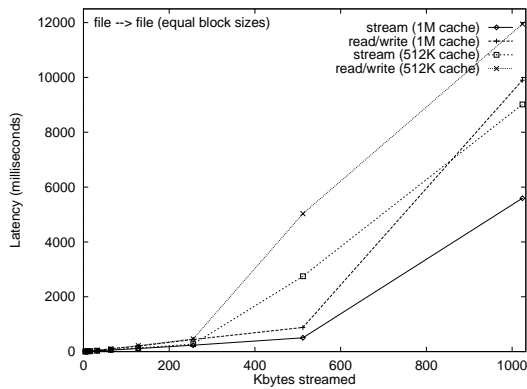


Figure 16: File copy stream, $B^{in} = B^{out}$

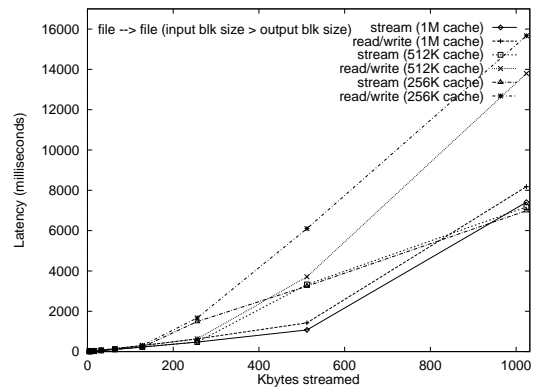


Figure 17: File copy stream, $B^{in} > B^{out}$

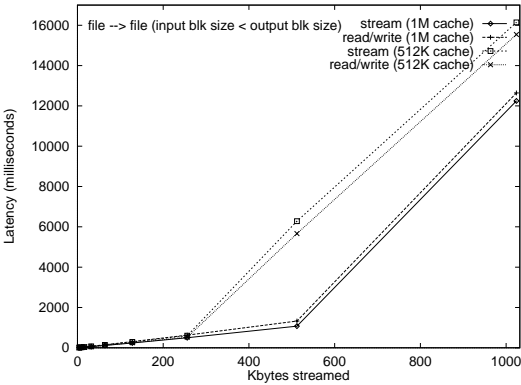


Figure 18: File copy stream, $B^{in} < B^{out}$

noted however that in both the web server and raw TCP write measurements, a modified XINU² implementation is being used. This implementation contains few of the optimizations present in BSD derived stacks, such as no-copy send and receive buffers, optimized checksums and header prediction.

8 Conclusion

We have presented a general framework that characterizes the need for copying in systems that perform data streaming. We have demonstrated analytically that true zero-copy data transfers can be used in many instances and described the performance of a prototype implementation in a real-time, embeddable, 32-bit operating system. We have shown that zero-copy transfers uniformly improve performance in our environment. It is important to note that these improvements are hardware independent. The I/O system design and streaming algorithms are specific only to our operating system architecture and should provide speedups on any hardware platform on which they are used.

Our approach does have potential drawbacks. First, the exclusive use of *move* semantics eliminates the possibility of concurrent access to a single buffer, which reduces the number of expensive I/O operations performed. Second, performance improvements are greater when I/O transfer times are close to data copy times, and this is

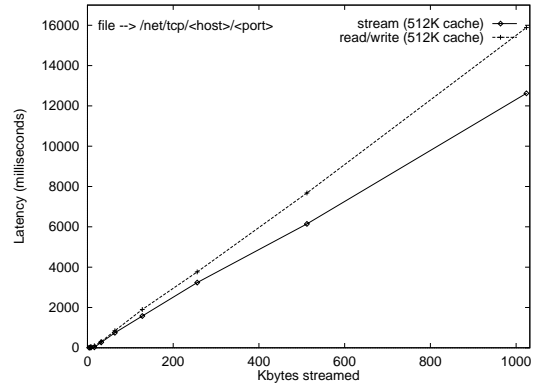


Figure 19: Web server stream

not generally the case for disk I/O.

Future work will concentrate on investigating the areas mentioned earlier, i.e. how application based partitioning compares to current systems use of subsystem partitioning and mechanisms and policies for managing the global buffer cache domination problem. In addition, we plan to investigate whether certain hardware features lend themselves to greater improvements with this type of I/O system architecture.

Finally, we have implemented an entirely new I/O system from scratch in order to prototype this design approach. The I/O system was made small, fast, and inherently extensible by using a stackable *vnode*-based design. While the *Roadrunner* system is targeted at embedded system development, we do have a set of tools that allow self-hosted development. The authors are prepared to make the complete sources for the *Roadrunner* system available for further research.

References

1. Brustolini, J. and Steenkiste, P., “Effects of Buffering Semantics on I/O Performance”, *Proc. of 2nd Symposium on Operating System Design and Implementation (OSDI)*, 1996.
2. Comer, D., *Internetworking with TCP/IP: Volume II*, Prentice-Hall, 1995.
3. Cornfed Systems, Inc., *Roadrunner Operating System Reference*, 1998.
4. Druschel, P., “Operating System Support for High-Speed Communication”, *CACM*, 39, 9, 1996.

5. Fall, K. and Pasquale, J., "Improving Continuous-Media Playback Performance with In-Kernel Data Paths", *Proc. of the Intl. Conference on Multimedia Computing and Systems (ICMCS)*, 1994.
6. Fall, K., *A Peer-to-Peer I/O System in Support of I/O Intensive Workloads*, Ph.D. Dissertation, University of California, San Diego, 1994.
7. Heidemann, J. and Popek, G., "File-System Development with Stackable Layers", *ACM Transactions on Computer Systems*, 12, 1, 1994.
8. I₂O Special Interest Group, *Intelligent I/O for Enterprise Computing*, Presentation by I₂O SIG Member Representatives at Fall '97 COMDEX, 1997.
9. Intel Corp., *80386 Programmer's Guide*, 1994.
10. Kleinman, S., "Vnodes: An Architecture for Multiple File System Types in Sun UNIX", *Proc. of the Summer 1986 Conference*, USENIX, 1986.
11. McKusick, M., Bostic, K., Karels, M., and Quarterman, J., *The Design and Implementation of the 4.4BSD Operating System*, Addison-Wesley, 1996.
12. Miller, F. W. and Tripathi, S. K., "An Integrated Input/Output System for Kernel Data Streaming", *Proc. of the SPIE/ACM Multimedia Computing and Networking (MMCN)*, 1998.
13. Pai V., Druschel P., and Zwaenopoel, W., *IO-lite: A Unified I/O Buffering and Caching System*, TR97-294, Rice University, 1997.
14. Pike, R., Presotto, D., Thompson, K., and Trickey H., "Plan 9 from Bell Labs", *Plan 9: The Documents - Volume 2*, AT&T Bell Laboratories, 1995.
15. Rosenthal, D., "Evolving the Vnode Interface", *Proc. of the Summer 1990 Conference*, USENIX, 1990.
16. Skinner, G. and Wong, T., "Stacking Vnodes: A Progress Report", *Proc. of the Summer 1993 Conference*, USENIX, 1993.