

Iperf version 1.7.0

Iperf User Docs

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Compiling

Once you have the distribution, on UNIX, unpack it using gzip and tar. That will create a new directory '**iperf**-<version#>' with the source files and documentation.

Iperf compiles cleanly on many systems including Linux, SGI IRIX, HP-UX, Solaris, AIX, and Cray UNICOS. Use 'make' to configure for your OS and compile the source code.

```
gunzip -c iperf-<version>.tar.gz | tar -xvf -  
cd iperf-<version>  
make
```

To install **iperf** and **jperf**, use 'make install', which will ask you where to install it. To recompile, the easiest way is to start over. Do 'make distclean' then 'make'. See the Makefile for more options.

If you have problems, please [contact us and let us know](#) and we will try to fix them quickly.

Features

- TCP
 - Measure bandwidth
 - Report MSS/MTU size and observed read sizes.
 - Support for TCP window size via socket buffers.
 - Multi-threaded if pthreads or Win32 threads are available. Client and server can have multiple simultaneous connections.
- UDP
 - Client can create UDP streams of specified bandwidth.
 - Measure packet loss
 - Measure delay jitter
 - Multicast capable
 - Multi-threaded if pthreads are available. Client and server can have multiple simultaneous connections. (This doesn't work in Windows.)

- Where appropriate, options can be specified with K (kilo-) and M (mega-) suffices. So 128K instead of 131072 bytes.
- Can run for specified time, rather than a set amount of data to transfer.
- Picks the best units for the size of data being reported.
- Server handles multiple connections, rather than quitting after a single test.
- Print periodic, intermediate bandwidth, jitter, and loss reports at specified intervals.
- Run the server as a daemon (Check out [Nettest](#) for running it as a secure daemon).
- Run the server as a Windows NT Service
- Use representative streams to test out how link layer compression affects your achievable bandwidth.
- A library of [useful functions and C++ classes](#).

Command line option	Environment variable option	Description										
Client and Server options												
-f, --format [bkmaBKMA]	\$IPERF_FORMAT	A letter specifying the format to print bandwidth numbers in. Supported formats are <table style="width: 100%; border: none;"> <tr> <td>'b' = bits/sec</td> <td>'B' = Bytes/sec</td> </tr> <tr> <td>'k' = Kbits/sec</td> <td>'K' = KBytes/sec</td> </tr> <tr> <td>'m' = Mbits/sec</td> <td>'M' = MBytes/sec</td> </tr> <tr> <td>'g' = Gbits/sec</td> <td>'G' = GBytes/sec</td> </tr> <tr> <td>'a' = adaptive bits/sec</td> <td>'A' = adaptive Bytes/sec</td> </tr> </table> <p>The adaptive formats choose between kilo- and mega- as appropriate. Fields other than bandwidth always print bytes, but otherwise follow the requested format. Default is 'a'. <i>NOTE:</i> here Kilo = 1024, Mega = 1024² and Giga = 1024³ when dealing with bytes. Commonly in networking, Kilo = 1000, Mega = 1000², and Giga = 1000³ so we use this when dealing with bits. If this really bothers you, use -f b and do the math.</p>	'b' = bits/sec	'B' = Bytes/sec	'k' = Kbits/sec	'K' = KBytes/sec	'm' = Mbits/sec	'M' = MBytes/sec	'g' = Gbits/sec	'G' = GBytes/sec	'a' = adaptive bits/sec	'A' = adaptive Bytes/sec
'b' = bits/sec	'B' = Bytes/sec											
'k' = Kbits/sec	'K' = KBytes/sec											
'm' = Mbits/sec	'M' = MBytes/sec											
'g' = Gbits/sec	'G' = GBytes/sec											
'a' = adaptive bits/sec	'A' = adaptive Bytes/sec											
-i, --interval #	\$IPERF_INTERVAL	Sets the interval time in seconds between periodic bandwidth, jitter, and loss reports. If non-zero, a report is made every <i>interval</i> seconds of the bandwidth since the last report. If zero, no periodic reports are printed. Default is zero.										
-l, --len # [KM]	\$IPERF_LEN	The length of buffers to read or write. Iperf works by writing an array of <i>len</i> bytes a number of times. Default is 8 KB for TCP, 1470 bytes for UDP. Note for UDP, this is the datagram size and needs to be lowered when using IPv6 addressing to 1450 or less to avoid fragmentation. See also the -n and -t options.										
-m, --print_mss	\$IPERF_PRINT_MSS	Print the reported TCP MSS size (via the TCP_MAXSEG option) and the observed read sizes which often correlate with the MSS. The MSS is usually the MTU - 40 bytes for the TCP/IP header. Often a slightly smaller MSS is reported because of extra header space from IP options. The interface type corresponding to the MTU is also										

		printed (ethernet, FDDI, etc.). This option is not implemented on many OSES, but the read sizes may still indicate the MSS.
-p, --port #	\$IPERF_PORT	The server port for the server to listen on and the client to connect to. This should be the same in both client and server. Default is 5001, the same as <code>ttcp</code> .
-u, --udp	\$IPERF_UDP	Use UDP rather than TCP. See also the <code>-b</code> option.
-w, --window # [KM]	\$TCP_WINDOW_SIZE	Sets the socket buffer sizes to the specified value. For TCP, this sets the TCP window size. For UDP it is just the buffer which datagrams are received in, and so limits the largest receivable datagram size.
-B, --bind host	\$IPERF_BIND	Bind to <i>host</i> , one of this machine's addresses. For the client this sets the outbound interface. For a server this sets the incoming interface. This is only useful on multihomed hosts, which have multiple network interfaces. For Iperf in UDP server mode, this is also used to bind and join to a multicast group. Use addresses in the range 224.0.0.0 to 239.255.255.255 for multicast. See also the <code>-I</code> option.
-C, --compatibility	\$IPERF_COMPAT	Compatibility mode allows for use with older version of iperf . This mode is not required for interoperability but it is highly recommended. In some cases when using representative streaming you could cause a 1.7 server to crash or cause undesired connection attempts.
-M, --mss # [KM]	\$IPERF_MSS	Attempt to set the TCP maximum segment size (MSS) via the <code>TCP_MAXSEG</code> option. The MSS is usually the MTU - 40 bytes for the TCP/IP header. For ethernet, the MSS is 1460 bytes (1500 byte MTU). This option is not implemented on many OSES.
-N, --nodelay	\$IPERF_NODELAY	Set the TCP no delay option, disabling Nagle's algorithm. Normally this is only disabled for interactive applications like telnet.
-v (from v1.6 or higher)	.	Bind to an IPv6 address Server side: \$ iperf -s -V Client side: \$ iperf -c <Server IPv6 Address> -V Note: On version 1.6.3 and later a specific IPv6 Address does not need to be bound with the <code>-B</code> option, previous 1.6 versions do. Also on most OSES using this option will also respond to IPv4 clients using IPv4 mapped addresses.
Server specific options		
-s, --server	\$IPERF_SERVER	Run Iperf in server mode.
-D (from	.	Run the server as a daemon (Unix platforms)

v1.2 or higher)		On Win32 platforms where services are available, Iperf will start running as a service.
-R (only for Windows, from v1.2 or higher)		Remove the Iperf service (if it's running).
-o (only for Windows, from v1.2 or higher)		Redirect output to given file.
-c, --client host	\$IPERF_CLIENT	If Iperf is in server mode, then specifying a host with -c will limit the connections that Iperf will accept to the <i>host</i> specified. Does not work well for UDP.
-P, --parallel #	\$IPERF_PARALLEL	The number of connections to handle by the server before closing. Default is 0 (which means to accept connections forever).
Client specific options		
-b, --bandwidth # [KM]	\$IPERF_BANDWIDTH	The UDP bandwidth to send at, in bits/sec. This implies the -u option. Default is 1 Mbit/sec.
-c, --client host	\$IPERF_CLIENT	Run Iperf in client mode, connecting to an Iperf server running on <i>host</i> .
-d, --dualtest	\$IPERF_DUALTEST	Run Iperf in dual testing mode. This will cause the server to connect back to the client on the port specified in the -L option (or defaults to the port the client connected to the server on). This is done immediately therefore running the tests simultaneously. If you want an alternating test try -r.
-n, --num # [KM]	\$IPERF_NUM	The number of buffers to transmit. Normally, Iperf sends for 10 seconds. The -n option overrides this and sends an array of <i>len</i> bytes <i>num</i> times, no matter how long that takes. See also the -l and -t options.
-r, --tradeoff	\$IPERF_TRADEOFF	Run Iperf in tradeoff testing mode. This will cause the server to connect back to the client on the port specified in the -L option (or defaults to the port the client connected to the server on). This is done following the client connection termination, therefore running the tests alternating. If you want an simultaneous test try -d.
-t, --time #	\$IPERF_TIME	The time in seconds to transmit for. Iperf normally works by repeatedly sending an array of <i>len</i> bytes for <i>time</i> seconds. Default is 10 seconds. See also the -l and -n options.
-L, --listenport #	\$IPERF_LISTENPORT	This specifies the port that the server will connect back to the client on. It defaults to the port used to connect to the server from the client.
-P, --parallel	\$IPERF_PARALLEL	The number of simultaneous connections to make to the server. Default is 1. Requires thread support on both the client and server.

l #		
-S, --tos #	\$IPERF_TOS	The type-of-service for outgoing packets. (Many routers ignore the TOS field.) You may specify the value in hex with a '0x' prefix, in octal with a '0' prefix, or in decimal. For example, '0x10' hex = '020' octal = '16' decimal. The TOS numbers specified in RFC 1349 are: <pre> IPTOS_LOWDELAY minimize delay 0x10 IPTOS_THROUGHPUT maximize throughput 0x08 IPTOS_RELIABILITY maximize reliability 0x04 IPTOS_LOWCOST minimize cost 0x02 </pre>
-T, --ttl #	\$IPERF_TTL	The time-to-live for outgoing multicast packets. This is essentially the number of router hops to go through, and is also used for scoping. Default is 1, link-local.
-F (from v1.2 or higher)	.	Use a representative stream to measure bandwidth, e.g. :- \$ iperf -c <server address> -F <file-name>
-I (from v1.2 or higher)	.	Same as -F, input from stdin.
Miscellaneous options		
-h, --help		Print out a summary of commands and quit.
-v, --version		Print version information and quit. Prints 'pthreads' if compiled with POSIX threads, 'win32 threads' if compiled with Microsoft Win32 threads, or 'single threaded' if compiled without threads.

Tuning a TCP connection

The primary goal of **iperf** is to help in tuning TCP connections over a particular path. The most fundamental tuning issue for TCP is the TCP window size, which controls how much data can be in the network at any one point. If it is too small, the sender will be idle at times and get poor performance. The theoretical value to use for the TCP window size is the *bandwidth delay product*,

$$\text{bottleneck bandwidth} * \text{round trip time}$$

In the below modi4/cyclops example, the bottleneck link is a 45 Mbit/sec DS3 link and the round trip time measured with ping is 42 ms. The bandwidth delay product is

$$\begin{aligned}
& 45 \text{ Mbit/sec} * 42 \text{ ms} \\
& = (45\text{e}6) * (42\text{e-}3) \\
& = 1890000 \text{ bits} \\
& = 230 \text{ KByte}
\end{aligned}$$

That is a starting point for figuring the best window size; setting it higher or lower may produce better results. In our example, buffer sizes over 130K did not improve the

performance, despite the bandwidth delay product of 230K.

Note that many OSES and hosts have upper limits on the TCP window size. These may be as low as 64 KB, or as high as several MB. **iperf** tries to detect when these occur and give a warning that the actual and requested window sizes are not equal (as below, though that is due to rounding in IRIX). PSC has a [list detailing](#) how to change the default and maximum window sizes for various OSES. For more information on TCP window sizes, see the [User's Guide to TCP Windows](#).

Here is an example session, between node1 in Illinois and node2 in North Carolina. These are connected via the vBNS backbone and a 45 Mbit/sec DS3 link. Notice we improve bandwidth performance by a factor of 3 using proper TCP window sizes. Use the adaptive window sizes feature on platforms which allow setting window sizes in the granularity of bytes.

```
node2> iperf -s
```

```
-----  
Server listening on TCP port 5001  
TCP window size: 60.0 KByte (default)  
-----
```

```
[ 4] local <IP Addr node2> port 5001 connected with <IP Addr node1> port  
2357  
[ ID] Interval    Transfer    Bandwidth  
[ 4] 0.0-10.1 sec  6.5 MBytes  5.2 Mbits/sec
```

```
node1> iperf -c node2
```

```
-----  
Client connecting to node1, TCP port 5001  
TCP window size: 59.9 KByte (default)  
-----
```

```
[ 3] local <IP Addr node1> port 2357 connected with <IP Addr node2> port  
5001  
[ ID] Interval    Transfer    Bandwidth  
[ 3] 0.0-10.0 sec  6.5 MBytes  5.2 Mbits/sec
```

```
node2> iperf -s -w 130k
```

```
-----  
Server listening on TCP port 5001  
TCP window size: 130 KByte  
-----
```

```
[ 4] local <IP Addr node 2> port 5001 connected with <IP Addr node 1> port  
2530  
[ ID] Interval    Transfer    Bandwidth  
[ 4] 0.0-10.1 sec  19.7 MBytes 15.7 Mbits/sec
```

```
node1> iperf -c node2 -w 130k
```

```
-----  
Client connecting to node2, TCP port 5001  
TCP window size: 129 KByte (WARNING: requested 130 KByte)  
-----
```

```
[ 3] local <IP Addr node1> port 2530 connected with <IP Addr node2> port  
5001  
[ ID] Interval    Transfer    Bandwidth
```

```
[ 3] 0.0-10.0 sec 19.7 MBytes 15.8 Mbits/sec
```

Another test to do is run parallel TCP streams. If the total aggregate bandwidth is more than what an individual stream gets, something is wrong. Either the TCP window size is too small, or the OS's TCP implementation has bugs, or the network itself has deficiencies. See above for TCP window sizes; otherwise diagnosing which is somewhat difficult. If **iperf** is compiled with pthreads, a single client and server can test this, otherwise setup multiple clients and servers on different ports. Here's an example where a single stream gets 16.5 Mbit/sec, but two parallel streams together get 16.7 + 9.4 = 26.1 Mbit/sec, even when using large TCP window sizes:

```
node2> iperf -s -w 300k
```

```
-----  
Server listening on TCP port 5001  
TCP window size: 300 KByte
```

```
-----  
[ 4] local <IP Addr node2> port 5001 connected with <IP Addr node1> port  
6902
```

```
[ ID] Interval    Transfer    Bandwidth  
[ 4] 0.0-10.2 sec 20.9 MBytes 16.5 Mbits/sec
```

```
[ 4] local <IP Addr node2> port 5001 connected with <IP Addr node1> port 6911
```

```
[ 5] local <IP Addr node2> port 5001 connected with <IP Addr node2> port  
6912
```

```
[ ID] Interval    Transfer    Bandwidth  
[ 5] 0.0-10.1 sec 21.0 MBytes 16.7 Mbits/sec  
[ 4] 0.0-10.3 sec 12.0 MBytes  9.4 Mbits/sec
```

```
node1> ./iperf -c node2 -w 300k
```

```
-----  
Client connecting to node2, TCP port 5001  
TCP window size: 299 KByte (WARNING: requested 300 KByte)
```

```
-----  
[ 3] local <IP Addr node2> port 6902 connected with <IP Addr node1> port  
5001
```

```
[ ID] Interval    Transfer    Bandwidth  
[ 3] 0.0-10.2 sec 20.9 MBytes 16.4 Mbits/sec
```

```
node1> iperf -c node2 -w 300k -P 2
```

```
-----  
Client connecting to node2, TCP port 5001  
TCP window size: 299 KByte (WARNING: requested 300 KByte)
```

```
-----  
[ 4] local <IP Addr node2> port 6912 connected with <IP Addr node1> port  
5001
```

```
[ 3] local <IP Addr node2> port 6911 connected with <IP Addr node1> port 5001  
[ ID] Interval    Transfer    Bandwidth  
[ 4] 0.0-10.1 sec 21.0 MBytes 16.6 Mbits/sec  
[ 3] 0.0-10.2 sec 12.0 MBytes  9.4 Mbits/sec
```

A secondary tuning issue for TCP is the maximum transmission unit (MTU). To be most effective, both hosts should support Path MTU Discovery. PSC has a [list detailing](#) what OSes support Path MTU Discovery. Hosts without Path MTU Discovery often use 536 as

the MSS, which wastes bandwidth and processing time. Use the -m option to display what MSS is being used, and see if this matches what you expect. Often it is around 1460 bytes for ethernet.

```
node3> iperf -s -m
```

```
-----  
Server listening on TCP port 5001  
TCP window size: 60.0 KByte (default)  
-----  
[ 4] local <IP Addr node3> port 5001 connected with <IP Addr node4> port  
1096  
[ ID] Interval    Transfer    Bandwidth  
[ 4] 0.0- 2.0 sec  1.8 MBytes  6.9 Mbits/sec  
[ 4] MSS size 1448 bytes (MTU 1500 bytes, ethernet)  
[ 4] Read lengths occurring in more than 5% of reads:  
[ 4] 952 bytes read  219 times (16.2%)  
[ 4] 1448 bytes read 1128 times (83.6%)
```

Here is a host that doesn't support Path MTU Discovery. It will only send and receive small 576 byte packets.

```
node4> iperf -s -m
```

```
-----  
Server listening on TCP port 5001  
TCP window size: 32.0 KByte (default)  
-----  
[ 4] local <IP Addr node4> port 5001 connected with <IP Addr node3> port  
13914  
[ ID] Interval    Transfer    Bandwidth  
[ 4] 0.0- 2.3 sec  632 KBytes  2.1 Mbits/sec  
WARNING: Path MTU Discovery may not be enabled.  
[ 4] MSS size 536 bytes (MTU 576 bytes, minimum)  
[ 4] Read lengths occurring in more than 5% of reads:  
[ 4] 536 bytes read  308 times (58.4%)  
[ 4] 1072 bytes read  91 times (17.3%)  
[ 4] 1608 bytes read  29 times (5.5%)
```

iperf supports other tuning options, which were added for exceptional network situations like HIPPI-to-HIPPI over ATM.

Tuning a UDP connection

iperf creates a constant bit rate UDP stream. This is a very artificial stream, similar to voice communication but not much else.

You will want to adjust the datagram size (-l) to the size your application uses.

The server detects UDP datagram loss by ID numbers in the datagrams. Usually a UDP datagram becomes several IP packets. Losing a single IP packet will lose the entire datagram. To measure packet loss instead of datagram loss, make the datagrams small enough to fit into a single packet, using the -l option. The default size of 1470 bytes works for ethernet. Out-of-order packets are also detected. (Out-of-order packets cause some ambiguity in the lost packet count; **iperf** assumes they are not duplicate packets, so they are excluded from the lost packet count.) Since TCP does not report loss to the user, I find

UDP tests helpful to see packet loss along a path.

Jitter calculations are continuously computed by the server, as specified by RTP in RFC 1889. The client records a 64 bit second/microsecond timestamp in the packet. The server computes the relative transit time as (server's receive time - client's send time). The client's and server's clocks do not need to be synchronized; any difference is subtracted out in the jitter calculation. Jitter is the smoothed mean of differences between consecutive transit times.

```
node2> iperf -s -u -i 1
```

```
-----  
Server listening on UDP port 5001  
Receiving 1470 byte datagrams  
UDP buffer size: 60.0 KByte (default)  
-----
```

```
[ 4] local <IP Addr node2> port 5001 connected with <IP Addr node1> port 9726
```

```
[ ID] Interval    Transfer    Bandwidth    Jitter    Lost/Total Datagrams  
[ 4] 0.0- 1.0 sec  1.3 MBytes  10.0 Mb/s    0.209 ms  1/ 894 (0.11%)  
[ 4] 1.0- 2.0 sec  1.3 MBytes  10.0 Mb/s    0.221 ms  0/ 892 (0%)  
[ 4] 2.0- 3.0 sec  1.3 MBytes  10.0 Mb/s    0.277 ms  0/ 892 (0%)  
[ 4] 3.0- 4.0 sec  1.3 MBytes  10.0 Mb/s    0.359 ms  0/ 893 (0%)  
[ 4] 4.0- 5.0 sec  1.3 MBytes  10.0 Mb/s    0.251 ms  0/ 892 (0%)  
[ 4] 5.0- 6.0 sec  1.3 MBytes  10.0 Mb/s    0.215 ms  0/ 892 (0%)  
[ 4] 6.0- 7.0 sec  1.3 MBytes  10.0 Mb/s    0.325 ms  0/ 892 (0%)  
[ 4] 7.0- 8.0 sec  1.3 MBytes  10.0 Mb/s    0.254 ms  0/ 892 (0%)  
[ 4] 8.0- 9.0 sec  1.3 MBytes  10.0 Mb/s    0.282 ms  0/ 892 (0%)  
[ 4] 0.0-10.0 sec 12.5 MBytes  10.0 Mb/s    0.243 ms  1/ 8922 (0.011%)
```

```
node1> iperf -c node2 -u -b 10m
```

```
-----  
Client connecting to node2, UDP port 5001  
Sending 1470 byte datagrams  
UDP buffer size: 60.0 KByte (default)  
-----
```

```
[ 3] local <IP Addr node1> port 9726 connected with <IP Addr node2> port 5001
```

```
[ ID] Interval    Transfer    Bandwidth  
[ 3] 0.0-10.0 sec 12.5 MBytes  10.0 Mb/s  
[ 3] Sent 8922 datagrams
```

Notice the higher jitter due to datagram reassembly when using larger 32 KB datagrams, each split into 23 packets of 1500 bytes. The higher datagram loss seen here may be due to the burstiness of the traffic, which is 23 back-to-back packets and then a long pause, rather than evenly spaced individual packets.

```
node2> iperf -s -u -l 32k -w 128k -i 1
```

```
-----  
Server listening on UDP port 5001  
Receiving 32768 byte datagrams  
UDP buffer size: 128 KByte  
-----
```

```
[ 3] local <IP Addr node2> port 5001 connected with <IP Addr node1> port 11303
```

[ID]	Interval	Transfer	Bandwidth	Jitter	Lost/Total	Datagrams
[3]	0.0- 1.0 sec	1.3 MBytes	10.0 Mbites/sec	0.430 ms	0/ 41	(0%)
[3]	1.0- 2.0 sec	1.1 MBytes	8.5 Mbites/sec	5.996 ms	6/ 40	(15%)
[3]	2.0- 3.0 sec	1.2 MBytes	9.7 Mbites/sec	0.796 ms	1/ 40	(2.5%)
[3]	3.0- 4.0 sec	1.2 MBytes	10.0 Mbites/sec	0.403 ms	0/ 40	(0%)
[3]	4.0- 5.0 sec	1.2 MBytes	10.0 Mbites/sec	0.448 ms	0/ 40	(0%)
[3]	5.0- 6.0 sec	1.2 MBytes	10.0 Mbites/sec	0.464 ms	0/ 40	(0%)
[3]	6.0- 7.0 sec	1.2 MBytes	10.0 Mbites/sec	0.442 ms	0/ 40	(0%)
[3]	7.0- 8.0 sec	1.2 MBytes	10.0 Mbites/sec	0.342 ms	0/ 40	(0%)
[3]	8.0- 9.0 sec	1.2 MBytes	10.0 Mbites/sec	0.431 ms	0/ 40	(0%)
[3]	9.0-10.0 sec	1.2 MBytes	10.0 Mbites/sec	0.407 ms	0/ 40	(0%)
[3]	0.0-10.0 sec	12.3 MBytes	9.8 Mbites/sec	0.407 ms	7/ 401	(1.7%)

```
node1> iperf -c node2 -b 10m -l 32k -w 128k
```

```
-----
Client connecting to node2, UDP port 5001
Sending 32768 byte datagrams
UDP buffer size: 128 KByte
```

```
-----
[ 3] local <IP Addr node2> port 11303 connected with <IP Addr node1> port
5001
```

[ID]	Interval	Transfer	Bandwidth
[3]	0.0-10.0 sec	12.5 MBytes	10.0 Mbites/sec
[3]	Sent 401 datagrams		

Multicast

To test multicast, run several servers with the bind option (-B, --bind) set to the multicast group address. Run the client, connecting to the multicast group address and setting the TTL (-T, --ttl) as needed. Unlike normal TCP and UDP tests, multicast servers may be started after the client. In that case, datagrams sent before the server started show up as losses in the first periodic report (61 datagrams on arno below).

```
node5> iperf -c 224.0.67.67 -u --ttl 5 -t 5
```

```
-----
Client connecting to 224.0.67.67, UDP port 5001
Sending 1470 byte datagrams
Setting multicast TTL to 5
UDP buffer size: 32.0 KByte (default)
```

```
-----
[ 3] local <IP Addr node5> port 1025 connected with 224.0.67.67 port 5001
```

[ID]	Interval	Transfer	Bandwidth
[3]	0.0- 5.0 sec	642 KBytes	1.0 Mbites/sec
[3]	Sent 447 datagrams		

```
node5> iperf -s -u -B 224.0.67.67 -i 1
```

```
-----
Server listening on UDP port 5001
Binding to local address 224.0.67.67
Joining multicast group 224.0.67.67
Receiving 1470 byte datagrams
UDP buffer size: 32.0 KByte (default)
```

```
-----
[ 3] local 224.0.67.67 port 5001 connected with <IP Addr node5> port 1025
```

[ID]	Interval	Transfer	Bandwidth	Jitter	Lost/Total	Datagrams
[3]	0.0- 1.0 sec	131 KBytes	1.0 Mbits/sec	0.007 ms	0/ 91	(0%)
[3]	1.0- 2.0 sec	128 KBytes	1.0 Mbits/sec	0.008 ms	0/ 89	(0%)
[3]	2.0- 3.0 sec	128 KBytes	1.0 Mbits/sec	0.010 ms	0/ 89	(0%)
[3]	3.0- 4.0 sec	128 KBytes	1.0 Mbits/sec	0.013 ms	0/ 89	(0%)
[3]	4.0- 5.0 sec	128 KBytes	1.0 Mbits/sec	0.008 ms	0/ 89	(0%)
[3]	0.0- 5.0 sec	642 KBytes	1.0 Mbits/sec	0.008 ms	0/ 447	(0%)

```
node6> iperf -s -u -B 224.0.67.67 -i 1
```

```
-----
Server listening on UDP port 5001
Binding to local address 224.0.67.67
Joining multicast group 224.0.67.67
Receiving 1470 byte datagrams
UDP buffer size: 60.0 KByte (default)
-----
```

```
[ 3] local 224.0.67.67 port 5001 connected with <IP Addr node5> port 1025
[ ID] Interval    Transfer    Bandwidth   Jitter    Lost/Total Datagrams
[ 3] 0.0- 1.0 sec  129 KBytes  1.0 Mbits/sec  0.778 ms  61/ 151 (40%)
[ 3] 1.0- 2.0 sec  128 KBytes  1.0 Mbits/sec  0.236 ms   0/  89 (0%)
[ 3] 2.0- 3.0 sec  128 KBytes  1.0 Mbits/sec  0.264 ms   0/  89 (0%)
[ 3] 3.0- 4.0 sec  128 KBytes  1.0 Mbits/sec  0.248 ms   0/  89 (0%)
[ 3] 0.0- 4.3 sec  554 KBytes  1.0 Mbits/sec  0.298 ms  61/ 447 (14%)
```

IPv6 Mode

Download the IPv6 version of this release.

Get the IPv6 address of the node using the 'ifconfig' command.

Use the `-V` option to indicate that you are using an IPv6 address. Please note that we need to explicitly bind the server address also.

Server side:

```
$ iperf -s -V
```

Client side:

```
$ iperf -c <Server IPv6 Address> -V
```

Note: **iperf** version 1.6.2 and earlier require a IPv6 address to be explicitly bound with the

`-B` option for the server.

Using Representative Streams to measure bandwidth

Use the `-F` or `-I` option. If you want to test how your network performs with compressed / uncompressed streams, just create representative streams and use the `-F` option to test it. This is usually due to the link layer compressing data.

The `-F` option is for file input.

The `-I` option is for input from stdin.

E.g.

Client: \$ **iperf** -c <server address> -F <file-name>

Client: \$ **iperf** -c <server address> -l

Running the server as a daemon

Use the **-D** command line option to run the server as a daemon. Redirect the output to a file.

E.g. **iperf -s -D > iperfLog**. This will have the **iperf** Server running as a daemon and the server messages will be logged in the file iperfLog.

Using Iperf as a Service under Win32

There are three options for Win32:

- o outputfilename
output the messages into the specified file
- s -D
install **iperf** as a service and run it
- s -R
uninstall the **iperf** service

Examples:

iperf -s -D -o iperflog.txt

will install the **iperf** service and run it. Messages will be reported into "%windir %\system32\iperflog.txt"

iperf -s -R

will uninstall the **iperf** service if it is installed.

Note: If you stop want to restart the **iperf** service after having killed it with the Microsoft Management Console or the Windows Task Manager, make sure to use the proper OPTION in the service properties dialog.

Running the multicast server and client

Use the **-B** option while starting the server to bind it to a multicast address.

E.g. **iperf -s -u -B 224.0.55.55**.

This will have the **iperf** server listening for datagrams (-u) for the address 224.0.55.55(-B 224.0.55.55).

Now, start a client sending packets to this multicast address.

E.g. : **iperf -c 224.0.55.55 -u**. This will have a UDP client (-u) sending to the multicast address 224.0.55.55(-c 224.0.55.55).

Start multiple clients or servers as explained above, sending data to the same multicast server. (If you have multiple servers listening on the multicast address, each of the servers will be getting the data)

Direction of Traffic Flow

By default, the **iperf** client sends traffic to the server. Additionally, if either the "dualtest" or "tradeoff" options are specified, the client will also receive, either simultaneously or sequentially, respectively.