INTERNET-DRAFT Internet Engineering Task Force Issued: April 2001 Expires: October 2001

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Stream Control Transmission Protocol Applicability Statement
<draft-ietf-sigtran-sctp-applicability-06.txt>

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# Abstract

This document describes the applicability of the Stream Control Transmission Protocol (SCTP)[<u>RFC2960</u>] for general usage in the Internet. This document describes the key features of SCTP and how they are used for general purpose data transport.

Informational

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## **1** Introduction

# **<u>1.1</u>** Terminology

The following terms are commonly identified in related work:

Association: SCTP connection between two endpoints.

Transport address: A combination of IP address and SCTP port number.

Upper layer: The user of the SCTP protocol, which may be an adaptation layer, a session layer protocol, or the user application directly.

TLA: Top level aggregation, part of a aggregatable unicast address

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# **1.2** Protocol Overview

The Stream Control Transmission Protocol (SCTP) provides a reliable transport between two endpoints.

The following functions are provided by SCTP:

- Reliable Data Transfer
- Multiple streams to help avoid head-of-line blocking
- Ordered and unordered data delivery on a per-stream basis
- Bundling and fragmentation of user data
- TCP friendly Congestion and flow control
- Support continuous monitoring of reachability
- Graceful termination of association
- Support of multi-homing for added reliability
- Some protection against blind denial-of-service attacks
- Some protection against blind masquerade attacks

# **2** Applicability of SCTP

When making a choice between reliable transport protocols, namely UDP, TCP and SCTP, various factors will enter in to deciding which one to choose. Certain applications will find an extreme advantage to using SCTP, while others may find little advantage.

## 2.1 Features provided by SCTP

SCTP provides acknowledged, error-free, non-duplicated transfer of user data, with framing to preserve user protocol data unit boundaries, and transport-level data fragmentation to conform to discovered path MTU size. It provides sequenced delivery within streams, with the option for designating messages for order-of-arrival delivery instead.

SCTP also provides mechanisms for network-level fault tolerance through

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the use of multi-homing at either or both ends of an SCTP association. SCTP automatically detects failure to reach a destination address and compensates through the use of available alternate addresses.

#### **<u>2.2</u>** Applications that would benefit from using SCTP.

Applications using SCTP should have sufficient traffic levels to justify the overhead and benefit from SCTP association establishment and congestion and flow control procedures.

Applications that require framing of reliable data streams can get that feature from SCTP.

Applications which require ordered transport of messages, but transfer multiple independent message sequences that are unrelated (sometimes called transactions) will benefit from the partial ordering provided by SCTP streams.

Applications that need to transfer messages that hold no particular sequence or relationship to one another or can be correlated and sequenced at the application level can benefit from the unordered delivery service and transport-level fragmentation provided by SCTP.

Application which depend on fast retransmit of data will have a reduced dependence on timeouts(thus easing the load on the Operating System).

Applications requiring network layer redundancy can use SCTP's multihoming feature to support this, provided that the host supports multiple addresses and routing is configured appropriately (see <u>Section 3</u>).

Transport of PSTN signaling protocols such as Signaling System No. 7 over IP networks is an example application fitting these requirements.

#### **<u>2.3</u>** Applications that may receive little benefit.

Applications requiring strict ordering of all data sent between communicating endpoints would not benefit from the multi-stream capability provided by SCTP. In addition, applications oriented towards byte stream transfer would not benefit from SCTP framing.

An example application that would derives no benefit from framing and multi-stream capabilities of SCTP is file transfer.

Applications which do not require network-level redundancy or with

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physical limitations, e.g., only one network interface card is present, would derive no benefit from SCTP's multi-homing feature.

Applications which generate small amounts of unrelated transactions towards a destination do not gain a great benefit from using TCP friendly congestion control and may experience a conservative retransmission policy.

#### **3** Issues affecting deployment of SCTP

#### **<u>3.1</u>** SCTP multihoming and interaction with routing

For fault resilient communication between two SCTP endpoints, the multihoming feature needs more than one IP network interface for each endpoint. The number of paths used is the minimum of network interfaces used by any of the endpoints. It is recommended to bind the association to all the IP source addresses of the endpoint. Note that in IPv6, each network interface will have more than one IP address.

Under the assumption that every IP address will have a different, seperate paths towards the remote endpoint, (this is the responsibility of the routing protocols or of manual configuration), if the transport to one of the IP address (= 1 particular path) fails then the traffic can migrate to the other remaining IP address (= other paths) within the SCTP association.

+	-+ *~~~~~*	++
Endpoint A	* Cloud *	Endpoint B
1.2	++ 1.1<>3.1 +	+ 3.2
2.2	++ 2.1<>4.1 +	+ 4.2
	* *	
+	-+ *~~~~~*	++

Figure 3.1.1: Two hosts with redundant networks.

Consider figure 3.1.1, if the host routing tables look as follows the endpoint will achieve maximum use of the multi-homing feature:

Endpoint A		Endpoint B	
Destination	Gateway	Destination	Gateway
3.0	1.1	1.0	3.1
4.0	2.1	2.0	4.1

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Now if you consider figure 3.1.1, if the host routing table looks as follows, the association is subject to a single point of failure in that if any interface breaks, the whole association will break(See figure 3.1.2).

Host A Host B Destination Gateway Destination Gateway ----------3.0 1.11.0 4.1 4.0 2.1 2.0 3.1 Example: link 4.2-4.1 fails Primary path: link 1.2-1.1 - link 3.1-3.2 Second Path : Link 2.2-2.1 - link 4.1-4.2 Endpoint A +----+ |S= 1.2 | D= 3.2 | DATA | -----> Arrives at Endpoint B +----+ Endpoint B answers with SACK +----+ |S= 4.2 | D= 1.2 | SACK | Gets lost, because send out on the failed +----+ 4.1-4.2 link After X time, retransmit on the other path by endpoint A Endpoint A +----+ S= 2.2 | D= 4.2 | DATA | Is send out on link 2.2-2.1, but gets lost, +----+ as msg has to pass via failed 4.1-4.2 link The same scenario will play out for failures on the other links Note : S = Source address D = Destination address Figure 3.1.2: Single point of failure case in redundant network due to routing table in host B

When an endpoint selects its source address, careful consideration must be taken. If the same source address is always used, then it is possible that the endpoint will be subject to the same single point of failure illustrated above. If possible the endpoint should always select the source address of the packet to correspond to the IP address of the Network interface where the packet will be emitted.

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+	·+ *~~~	~~~~*	+	+
Endpoint A	* C	Loud *	Endpoir	nt B
1.2	++ 1.1<	+		
1		->3.1	+ 3.2	
2.2	++ 2.1<	+		
	*	*		
+	+ *~~~	~~~~*	+	+

Figure 3.1.3: Two hosts with asymmetric networks.

In Figure 3.1.3 consider the following host routing table:

In this case the fault tolerance becomes limited by two seperate issues. If the path between 3.1 and 3.2 breaks in both directions any association will break between endpoint A and endpoint B. The second failure will occur for the whole the association as well due to a breakage between 1.2 and 1.1 in both directions, since no alternative route exists to 3.2 and all traffic is being routed through one interface.

Now one of these issues can be remedied by the following modification even when only one interface exists on endpoint B.

+	- +	*~	~~~~~~	*	+	+
Endpoint A		*	Cloud	*	Endp	oint B
1.2	+	+ 1.	1<+	I		
			+->3	.1+	-+ 3.2	& 4.2
2.2	+	+ 2.	1<+		1	
		*		*		
+	-+	*~	~~~~~~	*	+	+

Figure 3.1.4: Two hosts with asymmetric networks, but symmetric addresses.

In Figure 3.1.4 consider the following host routing table:

Endpoint A		Endpoint B	
Destination	Gateway	Destination	Gateway
3.0	1.1	1.0	3.1
4.0	2.1	2.0	3.1

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Now with the duplicate IP addresses assigned to the same interface and the above routing tables, even if the interface between 1.1 and 1.2 breaks, an association will still survive this failure.

As a practical matter, it is recommended that IP addresses in a multihomed endpoint be assigned IP endpoints from different TLA's to ensure against network failure.

In IP implementations the outgoing interface of multihomed hosts is often determined by the destination IP address. The mapping is done by a lookup in a routing table maintained by the operating system. Therefore the outgoing interface is not determined by SCTP. Using such implementations, it should be noted that a multihomed host cannot make use of the multiple local IP addresses if the peer is singlehomed. The multihomed host has only one path and will normally use only one of its interfaces to send the SCTP datagrams to the peer. If this physical path fails, the IP routing table in the multihome host has to be changed. This problem is out of scope for SCTP.

SCTP will always send its traffic to a certain transport address (= destination address + port number combination) for as long as the transmission is uninterrupted (= primary). The other transport addresses (secondary paths) will act as a backup in case the primary path goes out of service. The changeover between primary and backup will occur without packet loss and is completely transparent to the application. The secondary path can also be used for retransmissions(per <u>section 6.4 of</u> [RFC2960]).

The port number is the same for all transport addresses of that specific association.

Applications directly using SCTP may choose to control the multihoming service themselves. The applications have then to supply the specific IP address to SCTP for each outbound user message. This might be done for reasons of load-sharing and load-balancing across the different paths. This might not be advisable as the throughput of any of the paths is not known in advance and constantly changes due to the actions of other associations and transport protocols along that particular path, would require very tight feedback of each of the paths to the loadsharing functions of the user.

By sending a keep alive message on all the multiple paths that are not used for active transmission of messages across the association, it is possible for SCTP to detect whether one or more paths have failed. SCTP will not use these failed paths when a changeover is required.

The transmission rate of sending keep alive message should be modifiable and the possible loss of keep alive message could be used for the

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monitoring and measurements of the concerned paths.

#### 3.2 Use of SCTP in Network Address Translators (NAT) Networks [RFC2663]

When a NAT is present between two endpoints, the endpoint that is behind the NAT, i.e., one that does not have a publicly available network address, shall take one of the following options:

(1) When single homed sessions are to be used, no transport addresses should be sent in the INIT or INIT ACK chunk(Refer to <u>section 3.3</u> of RFC2960 for chunk definitions). This will force the endpoint that receives this initiation message to use the source address in the IP header as the only destination address for this association. This method can be used for a NAT, but any multi-homing configuration at the endpoint that is behind the NAT will not be visible to its peer, and thus not be taken advantage of. See figure 3.2.1.

+	- +	+		- +	*~	~~~~~~	~~*	++
Host A	Ι	Ι	NAT		*	Cloud	*	Host B
10.2	+-	- 10	.1 2.1		-			+ 1.2
					*		*	
+	- +	+		- +	*~	~~~~~	~~*	++

#### Fig 3.2.1: SCTP through NAT without multihoming

For multihoming the NAT must have a public IP address for each represented internal IP address. The host can preconfigure IP address that the NAT can substitute. Or the NAT can have internal Application Layer Gateway (ALG) which will intelligently translate the IP addresses in the INIT and INIT ACK chunks. See Figure 3.2.2.

If Network Address Port Translation is used with a multihomed SCTP endpoint, then any port translation must be applied on a perassociation basis such that an SCTP endpoint continues to receive the same port number for all messages within a given association.

+	-+ +		- +	*~~	-~~~~~	~~*	+	+
Host A		NAT		*	Cloud	*	Host	: B
10.2	++	10.1 5.2	+	+ 1.	1<+->3	.1+	+ 1.2	2
11.2	++	11.1 6.2			+->4	. 2 +	+ 2.2	2
				*		*		
+	-+ +		- +	*_		~~*	+	+

Fig 3.2.2: SCTP through NAT with multihoming

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(2) Another alternative is to use the hostname feature and DNS to resolve the addresses. The hostname is included in the INIT of the association or in the INIT ACK. The hostname must be resolved by DNS before the association is completely set up. There are special issues regarding NAT and DNS, refer to <u>RFC2694</u> for details.

# **<u>4</u>** Security considerations

SCTP only tries to increase the availability of a network. SCTP does not contain any protocol mechanisms which are directly related to user message authentication, integrity and confidentiality functions. For such features, it depends on the IPSEC protocols and architecture and/or on security features of its user protocols.

Mechanisms for reducing the risk of blind denial-of-service attacks and masquerade attacks are built into SCTP protocol. See <u>RFC2960</u>, <u>section 11</u> for detailed information.

Currently the IPSEC working group is investigating the support of multihoming by IPSEC protocols. At the present time to use IPSEC, one must use 2 \* N \* M security associations if one endpoint uses N addresses and the other M addresses.

## **<u>5</u>** References and related work

- [RFC2960] Stewart, R. R., Xie, Q., Morneault, K., Sharp, C. , , Schwarzbauer, H. J., Taylor, T., Rytina, I., Kalla, M., Zhang, L. and Paxson, V."Stream Control Transmission Protocol", <u>RFC2960</u>, October 2000.
- [RFC2401] Kent, S., and Atkinson, R., "Security Architecture for the Internet Protocol", <u>RFC 2401</u>, November 1998.
- [RFC2663] Srisuresh, P. and Holdrege, M., "IP Network Address Translator (NAT) Terminology and Considerations", <u>RFC2663</u>, August 1999

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[RFC2694] Srisuresh, P., Tsirtsis, G., Akkiraju, P. and Heffernan, A., "DNS extensions to Network Address Translators (DNS\_ALG)", <u>RFC2694</u>, September 1999

## **<u>6</u>** Acknowledgments

The authors wish to thank Renee Revis, I. Rytina, H.J. Schwarzbauer, J.P. Martin-Flatin, T. Taylor, G. Sidebottom, K. Morneault, T. George, <u>M. Stillman, N. Makinae, S. Bradner, A. Mankin, G. Camarillo, H.</u> Schulzrinne, R. Kantola, J. Rosenberg and many others for their invaluable comments.

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