[MS-SRTP]:

Secure Real-time Transport Protocol (SRTP) Extensions

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Revision Summary

Date	Revision History	Revision Class	Comments
4/4/2008	0.1		Initial version
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6/30/2015	5.0	No Change	No changes to the meaning, language, or formatting of the technical content.

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1 Introduction

The Secure Real-time Transport Protocol (SRTP) Extensions Protocol specifies a set of proprietary extensions to the Secure Real-time Transport Protocol (SRTP). This protocol provides the same functional capabilities as SRTP, which include providing confidentiality, message authentication, and replay protection to the RTP traffic and to the control traffic for RTP.

This protocol is a strict subset of SRTP and differs from it in two key aspects:

- The first key difference is that this protocol supports a strict subset of the SRTP default cryptographic transform algorithms and requires that some parameters of the encryption and authentication algorithms described in [RFC3711] be of specific values. These requirements are specified in section 3.

- The second key difference is that there is a set of "MAY, SHOULD, MUST, SHOULD NOT, MUST NOT" protocol behaviors that differ between this protocol and [RFC3711]. Section 3 enumerates these behavioral differences.

Unless explicitly noted in this document, this protocol follows standard SRTP.

Sections 1.8, 2, and 3 of this specification are normative and can contain the terms MAY, SHOULD, MUST, MUST NOT, and SHOULD NOT as defined in [RFC2119]. Sections 1.5 and 1.9 are also normative but do not contain those terms. All other sections and examples in this specification are informative.

1.1 Glossary

The following terms are specific to this document:

Advanced Encryption Standard (AES): A block cipher that supersedes the Data Encryption Standard (DES). AES can be used to protect electronic data. The AES algorithm can be used to encrypt (encipher) and decrypt (decipher) information. Encryption converts data to an unintelligible form called ciphertext; decrypting the ciphertext converts the data back into its original form, called plaintext. AES is used in symmetric-key cryptography, meaning that the same key is used for the encryption and decryption operations. It is also a block cipher, meaning that it operates on fixed-size blocks of plaintext and ciphertext, and requires the size of the plaintext as well as the ciphertext to be an exact multiple of this block size. AES is also known as the Rijndael symmetric encryption algorithm [FIPS197].

AES Counter Mode: A type of counter-mode encryption that generates encryption key streams by using **Advanced Encryption Standard (AES)** cipher and successive integers.

cryptographic context: A set of cryptographic state information that is maintained in a Secure Real-Time Transport Protocol (SRTP) stream.

dual-tone multi-frequency (DTMF): In telephony systems, a signaling system in which each digit is associated with two specific frequencies. This system typically is associated with touch-tone keypads for telephones.

endpoint: A device that is connected to a computer network.

Hash-based Message Authentication Code (HMAC): A mechanism for message authentication (2) using cryptographic hash functions. HMAC can be used with any iterative cryptographic hash function (for example, MD5 and **SHA-1**) in combination with a secret shared key. The cryptographic strength of HMAC depends on the properties of the underlying hash function.

master key: A key that provides information for packet encryption and authentication (2) in **Secure Real-Time Transport Protocol (SRTP)** and Scale Secure Real-Time Transport Protocol (SSRTP) transactions.

NULL cipher: A cipher that does not modify a **Real-Time Transport Protocol (RTP)** payload and is defined in the **Secure Real-Time Transport Protocol (SRTP)** protocol. It is used when **RTP packet** encryption is not necessary, but packet authentication (1) is necessary.

Real-Time Transport Control Protocol (RTCP): A network transport protocol that enables monitoring of Real-Time Transport Protocol (RTP) data delivery and provides minimal control and identification functionality, as described in [RFC3550].

Real-Time Transport Protocol (RTP): A network transport protocol that provides end-to-end transport functions that are suitable for applications that transmit real-time data, such as audio and video, as described in [RFC3550].

RTCP packet: A control packet consisting of a fixed header part similar to that of **RTP packets**, followed by structured elements that vary depending upon the RTCP packet type. Typically, multiple RTCP packets are sent together as a compound RTCP packet in a single packet of the underlying protocol; this is enabled by the length field in the fixed header of each RTCP packet. See [RFC3550] section 3.

RTP packet: A data packet consisting of the fixed RTP header, a possibly empty list of contributing sources, and the payload data. Some underlying protocols may require an encapsulation of the RTP packet to be defined. Typically one packet of the underlying protocol contains a single RTP packet, but several RTP packets can be contained if permitted by the encapsulation method. See [RFC3550] section 3.

RTP profile: A collection that contains payload type codes and mappings to payload formats, such as media encodings. It can also define extensions or modifications to the **Real-Time Transport Protocol (RTP)** that are specific to a particular class of applications. Typically, an application operates under only one profile.

salt: An additional random quantity, specified as input to an encryption function that is used to increase the strength of the encryption.

Secure Real-Time Transport Protocol (SRTP): A profile of **Real-Time Transport Protocol (RTP)** that provides encryption, message authentication (2), and replay protection to the RTP data, as described in [RFC3711].

Session Description Protocol (SDP): A protocol that is used for session announcement, session invitation, and other forms of multimedia session initiation. For more information see [MS-SDP] and [RFC3264].

session key: A symmetric key that is derived from a master key and is used to encrypt or authenticate a specific media stream by using the **Secure Real-Time Transport Protocol (SRTP)** and Scale Secure Real-Time Transport Protocol (SSRTP).

SHA-1: An algorithm that generates a 160-bit hash value from an arbitrary amount of input data, as described in [RFC3174]. SHA-1 is used with the Digital Signature Algorithm (DSA) in the Digital Signature Standard (DSS), in addition to other algorithms and standards.

SHA-1 hash: A hashing algorithm as specified in [FIPS180-2] that was developed by the National Institute of Standards and Technology (NIST) and the National Security Agency (NSA).

Synchronization Source (SSRC): A 32-bit identifier that uniquely identifies a media stream (2) in a **Real-Time Transport Protocol (RTP)** session. An SSRC value is part of an **RTP packet** header, as described in [RFC3550].

MAY, SHOULD, MUST, SHOULD NOT, MUST NOT: These terms (in all caps) are used as defined in [RFC2119]. All statements of optional behavior use either MAY, SHOULD, or SHOULD NOT.

1.2 References

Links to a document in the Microsoft Open Specifications library point to the correct section in the most recently published version of the referenced document. However, because individual documents in the library are not updated at the same time, the section numbers in the documents may not match. You can confirm the correct section numbering by checking the Errata.

1.2.1 Normative References

We conduct frequent surveys of the normative references to assure their continued availability. If you have any issue with finding a normative reference, please contact dochelp@microsoft.com. We will assist you in finding the relevant information.

[MS-RTP] Microsoft Corporation, "Real-time Transport Protocol (RTP) Extensions".

[RFC2104] Krawczyk, H., Bellare, M., and Canetti, R., "HMAC: Keyed-Hashing for Message Authentication", RFC 2104, February 1997, http://www.ietf.org/rfc/rfc2104.txt

[RFC2119] Bradner, S., "Key words for use in RFCs to Indicate Requirement Levels", BCP 14, RFC 2119, March 1997, http://www.rfc-editor.org/rfc/rfc2119.txt

[RFC3711] Baugher, M., McGrew, D., Naslund, M., et al., "The Secure Real-time Transport Protocol (SRTP)", RFC 3711, March 2004, http://www.rfc-editor.org/rfc/rfc3711.txt

1.2.2 Informative References

[MS-DTMF] Microsoft Corporation, "RTP Payload for DTMF Digits, Telephony Tones, and Telephony Signals Extensions".

[MS-SDPEXT] Microsoft Corporation, "Session Description Protocol (SDP) Version 2.0 Extensions".

1.3 Overview

This protocol provides the same functionality as the **Secure Real-Time Transport Protocol (SRTP)** by providing confidentiality, message authentication, and replay protection to **Real-Time Transport Protocol (RTP)** traffic and to the control traffic for RTP, the **Real-Time Transport Control Protocol (RTCP)**.

This protocol is a strict subset of SRTP and differs from it in the following two key aspects. In all other cases, this protocol follows standard SRTP.

- The first key difference is that this protocol supports a subset of the SRTP default cryptographic transform algorithms, and it requires certain encryption and authentication algorithm parameters to be fixed values. For example, the **NULL cipher** transform is not supported.

- The second key difference is that there is a set of "MAY, SHOULD, MUST, SHOULD NOT, MUST NOT" protocol behaviors where this protocol differs in behavior from [RFC3711]. Section 3 enumerates these behavioral differences.

1.4 Relationship to Other Protocols

This protocol relies on **Session Description Protocol (SDP)** to exchange **master keys** and key parameters. Refer to [MS-SDPEXT] for SDP information pertinent to this protocol.

This protocol works with other **RTP profiles**; for example, **dual-tone multi-frequency (DTMF)**, as described in [MS-DTMF]. This protocol treats all other RTP profile outputs the same as audio or video

data. It encrypts and authenticates after processing is performed on the sending side and authenticates and decrypts before passing **RTP packets** and **RTCP packets** on the receiving side.

The Secure Real-time Transport Control Protocol (SRTCP) is considered a sub-protocol to SRTP, and they are described together in [RFC3711]. The proprietary implementation of SRTCP is specified in this document in a similar way.

1.5   Prerequisites/Preconditions

This protocol has the following prerequisites:

▪ This protocol requires that encryption and authentication algorithms are negotiated using SDP, as described in [MS-SDPEXT] section 3.1.5.8.

▪ This protocol requires that the master keys are exchanged using SDP, as described in [MS-SDPEXT] section 3.1.5.8, and the keys are configured properly.

▪ This protocol only provides message confidentiality, authentication, and replay protection for RTP packets and RTCP packets.

1.6   Applicability Statement

This protocol is used where users require secure RTP traffic. This protocol is required to be used with the SDP extension described in [MS-SDPEXT] section 3.1.5.8 to set up the shared master key securely.

1.7   Versioning and Capability Negotiation

None.

1.8   Vendor-Extensible Fields

None.

1.9   Standards Assignments

None.

2 Messages

2.1 Transport

This protocol transforms RTP/RTCP packets only. Refer to [MS-RTP] section 2.1 for transports that the RTP protocol uses.

2.2 Message Syntax

This protocol uses the message syntax specified in [RFC3711].

- For the SRTP message syntax, see [RFC3711] section 3.1.

- For the SRTCP message syntax, see [RFC3711] section 3.4.

3   Protocol Details

3.1   Endpoint Details

This protocol can be used to secure any RTP traffic. All behavior described here applies to both protocol client and server roles.

The following sections specify the differences between this protocol and SRTP, as specified in [RFC3711].

3.1.1   Abstract Data Model

This section describes a conceptual model of possible data organization that an implementation maintains to participate in this protocol. The described organization is provided to facilitate the explanation of how the protocol behaves. This document does not mandate that implementations adhere to this model as long as their external behavior is consistent with that described in this document.

This protocol requires that each **endpoint** in an SRTP session maintains **cryptographic contexts**. A cryptographic context has two categories of parameters:

- Transform independent parameters

- Transform dependent parameters

3.1.1.1   Transform Independent Parameters

Transform independent parameters are parameters independent of what encryption and authentication algorithms are used. For example, regardless of which authentication algorithm is used, the replay checklist size is fixed to 64 entries in this protocol. For details, see [RFC3711] section 3.2.1.

This protocol does not introduce new states, but does require some states to be specific values. For details, see section 3.1.3.2.

3.1.1.2   Transform Dependent Parameters

Transform dependent parameters are parameters for specific encryption or authentication algorithms. This protocol implements the default cryptographic transform specified in [RFC3711] section 4, with exceptions specified in section 3.1.3.3. No new states are introduced.

3.1.2   Timers

None.

3.1.3   Initialization

3.1.3.1   Cryptographic Contexts

SRTP requires that each endpoint in an SRTP session maintain cryptographic contexts. For more information, see [RFC3711] section 3.2.3. This protocol maintains cryptographic contexts differently from SRTP [RFC3711].

This protocol maintains two cryptographic contexts per SRTP session:

- One for all media streams on the send direction.

- One for all media streams on the receive direction.

This protocol supports multiple media streams sharing the same SRTP session. Each media stream MUST be uniquely identified by one **Synchronization Source (SSRC)**. This protocol maintains per SSRC transform independent parameters in cryptographic contexts, as specified in section 3.1.3.2.

When sending or receiving an SRTP packet, this protocol first uses the SRTP session and direction to identify the cryptographic context, then uses the SSRC in the packet to decide the per SSRC transform independent parameters in the cryptographic context.

3.1.3.2 SRTP Parameter Settings

For information regarding SRTP transform independent parameters and transform dependent parameters, see [RFC3711] sections 3.2.1 and 3.2.2.

This protocol requires the following parameter settings for transform independent parameters:

- The encryption algorithm MUST be **AES Counter Mode**, and encryption MUST be used.

- The authentication algorithm MUST be **Hash-based Message Authentication Code (HMAC)**-**SHA-1 hash**, and authentication MUST be used.

- The replay list size MUST be 64 entries.

- The master key indicator MUST be used.

- The master key indicator length MUST be one byte.

- The key derivation rate MUST be zero.

- The master key length MUST be 128-bit.

- The master **salt** key length MUST be 112-bit.

- The encryption **session key** length MUST be 128-bit.

- The encryption session salt length MUST be 112-bit.

- The authentication session key length MUST be 160-bit.

- The master key lifetime MUST be 2^{48} -1 packets for RTP and 2^{31} -1 for RTCP.

- SRTCP and SRTP MUST have the same parameter settings with the exceptions specified in [RFC3711] section 3.2.1.

This protocol maintains the following transform independent parameters per SSRC.

- The rollover counter

- The highest received RTP sequence number

- The replay list

For information regarding transform dependent parameters, see sections 3.1.3.3.1 and 3.1.3.3.2.

Unless explicitly noted, this protocol follows SRTP, as specified in [RFC3711], to set other mandatory parameters.

3.1.3.3 SRTP Default Cryptographic Transform

This protocol implements a subset of the default SRTP algorithms.

3.1.3.3.1 Message Encryption

The SRTP default encryption algorithms are specified in [RFC3711] section 4.1.

This protocol MUST use AES Counter Mode. **AES** in f8 mode or NULL cipher mode MUST NOT be used.

This protocol requires that the encryption algorithm MUST be AES Counter Mode with the following parameters. For parameter details, see [RFC3711] section 4.1.

- n_b (block cipher size) MUST be 128-bit (AES algorithm's fixed cipher block size).

- n_e (encryption key size) MUST be 128-bit.

- The Session salt key MUST be used and n_s MUST be 112-bit.

- SRTP_PREFIX_LENGTH MUST be 0.

3.1.3.3.2 Message Authentication and Integrity

The SRTP default authentication algorithm is Hash-based Message Authentication Code (HMAC)-**SHA-1** [RFC2104], as specified in [RFC3711] section 4.2. This protocol implements HMAC-SHA-1 and requires the following parameters:

- n_a (authentication key size) MUST be 160-bit.

- n_tag (authentication tag size) MUST be 80-bit.

3.1.3.4 Session Key Derivation

This protocol implements the session key derivation algorithm specified in [RFC3711] section 4.3.

3.1.4   Higher-Layer Triggered Events

None.

3.1.5   Message Processing Events and Sequencing Rules

3.1.5.1 SRTP Packet Processing

3.1.5.1.1 Sending an SRTP Packet

This protocol implements the steps specified in [RFC3711] section 3.3, with the exception of the method used to identify the appropriate cryptographic context and the per SSRC transform independent parameters. This protocol uses the method specified in section 3.1.3.1.

This protocol requires that RTP packets MUST be encrypted and authenticated.

3.1.5.1.2 Receiving an SRTP Packet

This protocol implements the steps specified in [RFC3711] section 3.3, with the following exceptions:

- This protocol uses the method specified in section 3.1.3.1 to identify the cryptographic context and the SSRC to identify the transform independent parameters in the cryptographic context.

- The replay checklist size MUST be 64 entries.

- This protocol logs the number of SRTP failures. Individual replay check failures or authentication failures are not logged.

3.1.5.2 SRTCP Packet Processing

3.1.5.2.1 Sending an SRTCP Packet

This protocol implements the steps specified in [RFC3711] section 3.4. RTCP packets MUST be encrypted and authenticated.

This protocol can adjust the avg_rtcp_size or packet_size variables, as specified in [RFC3711] section 3.4.

The SRTCP index counter is shared by all media streams on the same direction in the SRTP session.

3.1.5.2.2 Receiving an SRTCP Packet

This protocol implements the steps specified in [RFC3711] section 3.4, with the following exceptions:

- This protocol does not honor the e-bit. All incoming RTCP packets MUST be encrypted regardless of the e-bit setting.

- This protocol uses the method specified in section 3.1.3.1 to identify the cryptographic context to use.

- The SRTCP index counter is shared by all media streams.

- The replay checklist size MUST be 64 entries.

- This protocol logs the number of SRTCP failures. Individual replay check failures or authentication failures are not logged.

3.1.6 Timer Events

None.

3.1.7 Other Local Events

None.

4   Protocol Examples

This protocol does not introduce new protocol behaviors. The test vectors in [RFC3711] apply to this protocol. For more information, see [RFC3711] Appendix B.

5   Security

5.1   Security Considerations for Implementers

- Master keys are randomly generated. The send and receive directions in the same SRTP session do not use the same master key.

- Master key exchange is done through external mechanisms in SDP. SDP is transferred on a secure transport, for instance Transport Layer Security TLS.

- The Initial RTP sequence number is randomly generated. But it cannot use a value close to 65535, because this could cause a rollover counter mismatch if there is packet loss at the beginning of session startup. For example, the server products supported by this protocol use a random value between 0 and 32767.

- SRTP cannot terminate the connection when a replay attack is detected. Some RTP profiles intentionally send the same packet multiple times, and the duplicated packets fail replay check. For example, DTMF as described in [MS-DTMF].

5.2   Index of Security Parameters

Security parameter	Section
Encryption algorithm	3.1.3.2
Authentication algorithm	3.1.3.2
Replay list size	3.1.3.2
Master key indicator length	3.1.3.2
Session key derivation rate	3.1.3.2
Master key length	3.1.3.2
Master salt length	3.1.3.2
Encryption session key length	3.1.3.2
Encryption session salt length	3.1.3.2
Authentication session key length	3.1.3.2
Master key lifetime	3.1.3.2
Advanced Encryption Standard (AES) cipher block size	3.1.3.3.1
SRTP cipher prefix size	3.1.3.3.1
Authentication tag size	3.1.3.3.2

6   Appendix A: Product Behavior

The information in this specification is applicable to the following Microsoft products or supplemental software. References to product versions include released service packs.

- Microsoft Office Communications Server 2007

- Microsoft Office Communications Server 2007 R2

- Microsoft Office Communicator 2007

- Microsoft Office Communicator 2007 R2

- Microsoft Lync Server 2010

- Microsoft Lync 2010

- Microsoft Lync Server 2013

- Microsoft Skype for Business (formerly Lync 2013)

- Skype for Business

- Skype for Business Server

Exceptions, if any, are noted below. If a service pack or Quick Fix Engineering (QFE) number appears with the product version, behavior changed in that service pack or QFE. The new behavior also applies to subsequent service packs of the product unless otherwise specified. If a product edition appears with the product version, behavior is different in that product edition.

Unless otherwise specified, any statement of optional behavior in this specification that is prescribed using the terms SHOULD or SHOULD NOT implies product behavior in accordance with the SHOULD or SHOULD NOT prescription. Unless otherwise specified, the term MAY implies that the product does not follow the prescription.

No table of changes is available. The document is either new or has had no changes since its last release.

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