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**RTP Payload Format for the SCIP Codec**  
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## Abstract

This document describes the RTP payload format of the Secure Communication Interoperability Protocol (SCIP) as audio and video media subtypes. It provides [RFC 6838](#) compliant media subtype definitions. SCIP-214.2 and SCIP-210 describe the protocols that comprise the SCIP RTP packet payload. This document follows the registration for related media types called "audio/scip" and "video/scip" with IANA and formatted according to [RFC 4855](#).

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## **[1. Introduction](#)**

The IANA registration of media subtype types in the IETF tree created two similar media subtypes "scip" under the audio and video media types [[AUDIOSCIP](#)], [[VIDEOSCIP](#)]. This document, as the common top-level reference, provides information on their similarities and differences and the usage of those media subtypes.

This document details usage of the scip pseudo-codec as a secure session establishment protocol and transport protocol over RTP. It provides a reference for network security policymakers, network equipment OEMs, procurement personnel, and government agency and commercial industry representatives.



### **1.1. Conventions**

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [[RFC2119](#)].

Best current practices for writing an RTP payload format specification were followed [[RFC2736](#)] [[RFC8088](#)].

### **1.2. Abbreviations**

The following abbreviations are used in this document.

AVP:	Audio/Video Profile
DTX:	Discontinuous Transmission
FNBBDT:	Future Narrowband Digital Terminal
ICWG:	Interoperability Control Working Group
IICWG:	International Interoperability Control Working Group
NATO:	North Atlantic Treaty Organization
SCIP:	Secure Communication Interoperability Protocol
SDP:	Session Description Protocol

## **2. Background**

The Secure Communication Interoperability Protocol (SCIP) allows the negotiation of several voice, data, and video applications using various encryption suites. SCIP also provides several important characteristics that have led to its broad acceptance in the international user community. These features include end-to-end security at the application layer, authentication of user identity, the ability to apply different security levels for each secure session, and secure communication over any end-to-end data connection.

SCIP began in the U.S. as the Future Narrowband Digital Terminal (FNBBDT) Protocol. A combined Department of Defense and vendor consortium formed a governing organization named the Interoperability Control Working Group (ICWG) to manage the protocol. In time, the group expanded to include NATO, NATO partners and European vendors under the name International Interoperability Control Working Group (IICWG), which was later renamed the SCIP Working Group.

SCIP is presently implemented in U.S. and NATO secure voice, video, and data products operating on commercial, private, and tactical IP networks worldwide using the scip media subtype. First generation SCIP devices operated on circuit-switched networks. SCIP was then expanded to radio and IP networks. The scip media subtype transports SCIP secure session establishment signaling and secure application traffic. The built-in negotiation and flexibility provided by the SCIP standards make it a natural choice for many scenarios that require various secure applications and associated encryption suites. SCIP has been endorsed by many nations as the secure end-to-end solution for secure voice, video, and data devices. SCIP standards are currently available to participating government/military communities and select OEMs of equipment that support SCIP.

However, SCIP must operate over global networks (including private and commercial networks). Without access to necessary information to support SCIP, some networks may not support the SCIP media subtypes. Issues may occur simply because information is not as readily available to OEMs, network administrators, and network architects.

This RFC provides essential information about audio/scip and video/scip media subtypes that enables network equipment manufacturers to include scip as a known audio and video media subtype in their equipment and enables network administrators to define and implement a compatible security policy.

All current IP-based SCIP devices support "scip" as a media subtype. Registration of scip as a media subtype provides a common reference for network equipment manufacturers to recognize SCIP in a payload declaration.

### **3. Media Format Description**

The "scip" media subtype indicates support for and identifies SCIP traffic that is being transferred using RTP. Transcoding, lossy compression, or other data modifications SHALL NOT be performed on the SCIP RTP payload. The audio/scip and video/scip media subtype data streams within the network, including the VoIP network, MUST be a transparent relay and be treated as "clear-channel data", similar to the Clearmode media subtype defined by [RFC 4040](#). However, Clearmode is defined as a gateway protocol and limited to a sample rate of 8000 Hz and 64kbps bandwidth only [[RFC4040](#)]. Clearmode is not defined for



the higher sample and data rates required for some SCIP traffic.

#### **4. Payload Format**

The RTP Packet content of SCIP traffic is dependent upon the SCIP session state. SCIP secure session establishment uses protocols defined in SCIP-210 [[SCIP210](#)] to negotiate an application. SCIP secure traffic may consist of the encrypted output of codecs such as MELPe [[RFC8130](#)], G.729D [[RFC3551](#)], H.264 [[RFC6184](#)], or other media encodings, based on the application negotiated during SCIP secure session establishment. SCIP traffic is highly variable and may include other SCIP signaling information in the media stream. SCIP traffic may not always be a continuous stream at the bit rate specified in the SDP [[RFC8866](#)] since discontinuous transmission (DTX) or other mechanisms may be used. The SCIP payload size will vary, especially during SCIP secure session establishment.

##### **4.1. RTP Header Fields**

The SCIP RTP header fields SHALL conform to [RFC 3550](#).

SCIP traffic may be continuous or discontinuous. The Timestamp field increments based on the sampling clock for discontinuous transmission as described in [[RFC3550](#)], [Section 5.1](#). The Timestamp field for continuous transmission applications is dependent on the sampling rate of the media as specified in the media subtype's specification (e.g., MELPe [[RFC8130](#)]). Note that during a call, both discontinuous and continuous traffic are highly probable. Therefore, a jitter buffer MAY be implemented in endpoint devices only but SHOULD NOT be implemented in network devices. Additionally, network devices SHOULD NOT repacketize SCIP packets.

The Marker bit SHALL be set to zero for discontinuous traffic. The Marker bit for continuous traffic is based on the underlying media subtype specification. The underlying media is opaque within SCIP RTP packets.

#### **5. Payload Format Parameters**

The SCIP RTP payload format is identified using the scip media subtype, which is registered in accordance with [[RFC4855](#)] and per the media type registration template form [[RFC6838](#)]. A clock rate of 8000 Hz SHALL be used for "audio/scip". A clock rate of 90000 Hz SHALL be used for "video/scip".





### **5.1. Media Subtype "audio/scip"**

Media type name: audio

Media subtype name: scip

Required parameters: N/A

Optional parameters: N/A

Encoding considerations: Binary. This media subtype is only defined for transfer via RTP. There SHALL be no encoding/decoding (transcoding) of the audio stream as it traverses the network.

Security considerations: See [Section 6](#).

Interoperability considerations: N/A

Published specifications: [[SCIP214](#)], [[SCIP210](#)]

Applications which use this media: N/A

Fragment Identifier considerations: none

Restrictions on usage: N/A

Additional information:

1. Deprecated alias names for this type: N/A
2. Magic number(s): N/A
3. File extension(s): N/A
4. Macintosh file type code: N/A
5. Object Identifiers: N/A

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**5.2. Media Subtype "video/scip"**

Media type name: video

Media subtype name: scip

Required parameters: N/A

Optional parameters: N/A

Encoding considerations: Binary. This media subtype is only defined for transfer via RTP. There SHALL be no encoding/decoding (transcoding) of the video stream as it traverses the network.

Security considerations: See [Section 6](#).

Interoperability considerations: N/A

Published specifications: [[SCIP214](#)], [[SCIP210](#)]

Applications which use this media: N/A

Fragment Identifier considerations: none

Restrictions on usage: N/A

## Additional information:

1. Deprecated alias names for this type: N/A
2. Magic number(s): N/A
3. File extension(s): N/A
4. Macintosh file type code: N/A

## 5. Object Identifiers: N/A

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### **5.3. Mapping to SDP**

The mapping of the above defined payload format media subtype and its parameters SHALL be done according to [Section 3 of \[RFC4855\]](#).

An example mapping for audio/scip is:

```
m=audio 50000 RTP/AVP 96
a=rtpmap:96 scip/8000
```

An example mapping for video/scip is:

```
m=video 50002 RTP/AVP 97
a=rtpmap:97 scip/90000
```

An example mapping for both audio/scip and video/scip is:

```
m=audio 50000 RTP/AVP 96
a=rtpmap:96 scip/8000
m=video 50002 RTP/AVP 97
a=rtpmap:97 scip/90000
```

The application negotiation between endpoints will determine whether the audio and video streams are transported as separate

streams over the audio and video payload types or as a single media stream on the video payload type.

#### **5.4. SDP Offer/Answer Considerations**

In accordance with the SDP Offer/Answer model [[RFC3264](#)], the SCIP device SHALL list the SCIP payload type in order of preference in the "m" media line.

### **6. Security Considerations**

RTP packets using the payload format defined in this specification are subject to the security considerations discussed in the RTP specification [[RFC3550](#)], and in any applicable RTP profile such as RTP/AVP [[RFC3551](#)], RTP/AVPF [[RFC4585](#)], RTP/SAVP [[RFC3711](#)], or RTP/SAVPF [[RFC5124](#)]. However, as "Securing the RTP Protocol Framework: Why RTP Does Not Mandate a Single Media Security Solution" [[RFC7202](#)] discusses, it is not an RTP payload format's responsibility to discuss or mandate what solutions are used to meet the basic security goals like confidentiality, integrity, and source authenticity for RTP in general. This responsibility lays on anyone using RTP in an application. They can find guidance on available security mechanisms and important considerations in "Options for Securing RTP Sessions" [[RFC7201](#)]. Applications SHOULD use one or more appropriate strong security mechanisms. The rest of this Security Considerations section discusses the security impacting properties of the payload format itself.

This RTP payload format and its media decoder do not exhibit any significant non-uniformity in the receiver-side computational complexity for packet processing, and thus are unlikely to pose a denial-of-service threat due to the receipt of pathological data. Nor does the RTP payload format contain any active content.

### **7. IANA Considerations**

The audio/scip and video/scip media subtypes have been registered with IANA [[AUDIOSCIP](#)] [[VIDEOSCIP](#)].

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