

Signaling System 7

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Signaling System Number 7 (SS7) is a set of telephony signaling protocols which are used to set up most of the world's public switched telephone network telephone calls. The main purpose is to set up and tear down telephone calls. Other uses include number translation, prepaid billing mechanisms, short message service (SMS), and a variety of other mass market services.

It is usually abbreviated as **Signaling System No. 7**, **Signaling System #7**, or just **SS7**. In North America it is often referred to as **CCSS7**, an acronym for "Common Channel Signaling System 7". In some European countries, specifically the United Kingdom, it is sometimes called **C7** (CCITT number 7) and is also known as **number 7** and **CCIS7**. (ITU-T was formerly known as CCITT.)

There is only one international SS7 protocol defined by ITU-T in its Q.700-series recommendations.^[1] There are however, many national variants of the SS7 protocols. Most national variants are based on two widely deployed national variants as standardized by ANSI and ETSI, which are in turn based on the international protocol defined by ITU-T. Each national variant has its own unique characteristics. Some national variants with rather striking characteristics are the China (PRC) and Japan (TTC) national variants.

The Internet Engineering Task Force (IETF) has also defined level 2, 3, and 4 protocols that are compatible with SS7 MTP2 (M2UA and M2PA) MTP3 (M3UA) and SCCP (SUA), but use an SCTP transport mechanism. This suite of protocols is called SIGTRAN.

SS7 protocol suite	
OSI Layer	SS7 Protocols
Application	INAP, MAP, IS-41...
Network	TCAP, CAP, ISUP, ...
Data link	MTP Level 3 + SCCP
Physical	MTP Level 2
	MTP Level 1

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History

Common Channel Signaling protocols have been developed by major telephone companies and the ITU-T since 1975 and the first international Common Channel Signaling protocol was defined by the ITU-T as Signalling System No. 6 in 1977.^[2] Signalling System No. 7 was defined as an international standard by ITU-T in its 1980 (Yellow Book) Q.7XX-series recommendations.^[1] SS7 was designed to replace Signalling System No. 6, which had a restricted 28-bit signal unit that was both limited in function and not amenable to digital systems.^[3] SS7 has substantially replaced SS6, SS5, R1 and R2, with the exception that R1 and R2 variants are still used in numerous nations.

SS5 and earlier used in-band signaling, where the call-setup information was sent by playing special multi-frequency tones into the telephone lines (known as *bearer channels* in the parlance of the telecom industry). This led to security problems with *blue boxes*. Modern designs of telephone equipment that implement out-of-band signaling protocols explicitly keep the end-user's audio path—the so-called *speech path*—separate from the signaling phase to eliminate the possibility that end users may introduce tones that would be mistaken for those used for signaling. See *falsing*.

SS6 and SS7 moved to a system in which the signaling information was out-of-band, carried in a separate *signaling channel*.^[4] This avoided the security problems earlier systems had, as the end user had no connection to these channels. SS6 and SS7 are referred to as so-called *Common Channel Interoffice Signalling Systems* (CCIS) or

Common Channel Signaling (CCS) due to their hard separation of signaling and bearer channels. This required a separate channel dedicated solely to signaling, but the greater speed of signaling decreased the holding time of the bearer channels, and the number of available channels was rapidly increasing anyway at the time SS7 was implemented.

The common channel signaling paradigm was translated to IP via the SIGTRAN protocols as defined by the IETF. While running on a transport based upon IP, the SIGTRAN protocols are not an SS7 variant, but simply transport existing national and international variants of SS7.^[5]

Functionality

The term signalling, when used in telephony, refers to the exchange of control information associated with the establishment of a telephone call on a telecommunications circuit.^[6] An example of this control information is the digits dialed by the caller, the caller's billing number, and other call-related information.

When the signalling is performed on the same circuit that will ultimately carry the conversation of the call, it is termed Circuit-Associated Signalling (CAS). This is the case for earlier analogue trunks, MF and R2 digital trunks, and ISDN PRI or DSS1/DASS PBX trunks.

In stark contrast, SS7 signalling is termed Non-Circuit-Associated Signalling (NCAS) in that the path and facility used by the signalling is separate and distinct from the telecommunications channels that will ultimately carry the telephone conversation. With Non-Circuit-Associated Signalling, it becomes possible to exchange signalling without first seizing a facility, leading to significant savings and performance increases in both signalling and facility usage.

Because of the mechanisms used by signalling methods prior to SS7 (battery reversal, multi-frequency digit outpulsing, A- and B-bit signalling), these older methods could not communicate much signalling information. Usually only the dialed digits were signalled, and only during call setup. For charged calls, dialed digits and charge number digits were outpulsed. SS7, being a high-speed and high-performance packet-based communications protocol, can communicate significant amounts of information when setting up a call, during the call, and at the end of the call. This permits rich call-related services to be developed. Some of the first such services were call management related services that we take for granted today: call forwarding (busy and no answer), voice mail, call waiting, conference calling, calling name and number display, called name and number display, call screening, malicious caller identification, busy callback.^[7]

The earliest deployed upper layer protocol in the SS7 signalling suite were dedicated to the setup, maintenance, and release of telephone calls.^[8] The Telephone User Part (TUP) was adopted in Europe and the Integrated Services Digital Network (ISDN) User Part (ISUP) adapted for Public Switched Telephone Network (PSTN) calls was adopted in North America to process Plain Old Telephone System (POTS) telephone calls. ISUP was later used in Europe when the European networks upgraded to the ISDN. (North America never accomplished full upgrade to the ISDN and the predominant telephone service is still the older PSTN POTS service.) Due to its richness and the need for a completely separate signalling network for its operation, SS7 signalling is mostly used for signalling between telephone switches and not for signalling between local exchanges and Customer Premises Equipment (CPE).

Because SS7 signalling does not require seizure of a channel for a conversation prior to the exchange of control information, Non-Facility-Associated Signalling (NFAS) became possible. Non-Facility-Associated Signalling is signalling that is not directly associated with the path that a conversation will traverse and may concern other information located at a centralized database such as service subscription, feature activation, and service logic. This makes possible a set of network-based services that do not rely upon the call being routed to a particular subscription switch at which service logic would be executed, but permits service logic to be distributed throughout the telephone network and executed more expediently at originating switches far in advance of call routing. It also permits the subscriber increased mobility due to the decoupling of service logic from the subscription switch. Another characteristic of ISUP made possible by SS7 with NFAS is the exchange of signalling information during the middle of a call.^[9]

Also possible with SS7 is Non-Call-Associated Signalling. Non-Call-Associated Signalling is signalling that is not directly related to the establishment of a telephone call.^[10] An example of this is the exchange of the registration information used between a mobile telephone and a Home Location Register (HLR) database: a database that tracks the location of the mobile. Other examples include Intelligent Network and Local Number Portability databases.^[11]

Signalling Modes

As well as provided for signalling with these various degrees of association with call set up and the facilities used to carry calls, SS7 is designed to operate in two modes:^[12]

- Associated Mode
- Quasi-Associated Mode

When operating in the *Associated Mode*, SS7 signalling progresses from switch to switch through the PSTN following the same path as the *associated* facilities that carry the telephone call. This mode is more economical for small networks. The *Associated Mode* of signalling is not the predominant choice of modes in North America.^[13]

When operating in the *Quasi-Associated Mode*, SS7 signalling progresses from the originating switch to the terminating switch following a path through a separate SS7 signalling network composed of STPs. This mode is more economical for large networks with lightly loaded signalling links. The *Quasi-Associated Mode* of signalling is the predominant choice of modes in North America.^[14]

Physical network

SS7 clearly splits the signaling planes and voice circuits. An SS7 network has to be made up of SS7-capable equipment from end to end in order to provide its full functionality. The network is made up of several link types (A, B, C, D, E, and F) and three signaling nodes - Service switching point (SSPs), signal transfer point (STPs), and Service Control Point (SCPs). Each node is identified on the network by a number, a point code. Extended services are provided by a database interface at the SCP level using the SS7 network.

The links between nodes are full-duplex 56, 64, 1,536, or 1,984 kbit/s graded communications channels. In Europe they are usually one (64 kbit/s) or all (1,984 kbit/s) timeslots (DS0s) within an E1 facility; in North America one (56 or 64 kbit/s) or all (1,536 kbit/s) timeslots (DS0As or DS0s) within an T1 facility. One or more signaling links can be connected to the same two endpoints that together form a signaling link set. Signaling links are added to link sets to increase the signaling capacity of the link set.

In Europe, SS7 links normally are directly connected between switching exchanges using F-links. This direct connection is called *associated signalling*. In North America, SS7 links are normally indirectly connected between switching exchanges using an intervening network of STPs. This indirect connection is called *quasi-associated signalling*. *Quasi-associated signalling* reduces the number of SS7 links necessary to interconnect all switching exchanges and SCPs in an SS7 signaling network.^[15]

SS7 links at higher signaling capacity (1.536 and 1.984 Mbit/s, simply referred to as the 1.5 Mbit/s and 2.0 Mbit/s rates) are called *High Speed Links (HSL)* in contrast to the low speed (56 and 64 kbit/s) links. *High Speed Links (HSL)* are specified in ITU-T Recommendation Q.703 (<http://www.itu.int/T-REC-Q.703/en/>) for the 1.5 Mbit/s and 2.0 Mbit/s rates, and ANSI Standard T1.111.3 for the 1.536 Mbit/s rate. There are differences between the specifications for the 1.5 Mbit/s rate. *High Speed Links* utilize the entire bandwidth of a T1 (1.536 Mbit/s) or E1 (1.984 Mbit/s) transmission facility for the transport of SS7 signaling messages.^[16]

SIGTRAN provides signaling using SCTP associations over the Internet Protocol.^[17] The protocols for SIGTRAN are M2PA, M2UA, M3UA and SUA.

SS7 protocol suite

The SS7 protocol stack borrows partially from the OSI Model of a packetized digital protocol stack. OSI layers 1 to 3 are provided by the Message Transfer Part (MTP) and the Signalling Connection Control Part (SCCP) of the SS7 protocol (together referred to as the Network Service Part (NSP)); for circuit related signalling, such as the Telephone User Part (TUP) or the ISDN User Part (ISUP), the User

SS7 protocol suite

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Data link	MTP Level 3 + SCCP MTP Level 2

Part provides layer 7. Currently there are no protocol components that provide OSI layers 4 through 6.^[1]

The Transaction Capabilities Application Part

(TCAP) is the primary SCCP User in the Core Network, using SCCP in connectionless mode. SCCP in connection oriented mode provides the transport layer for air interface protocols such as BSSAP and RANAP. TCAP provides transaction capabilities to its Users (TC-Users), such as the Mobile Application Part, the Intelligent Network Application Part and the CAMEL Application Part.

The MTP covers a portion of the functions of the OSI network layer including: network interface, information transfer, message handling and routing to the higher levels. SCCP is at functional Level 4. Together with MTP Level 3 it is called the Network Service Part (NSP). SCCP completes the functions of the OSI network layer: end-to-end addressing and routing, connectionless messages (UDTs), and management services for users of the Network Service Part (NSP).^[18] TUP is a link-by-link signaling system used to connect calls. ISUP is the key user part, providing a circuit-based protocol to establish, maintain, and end the connections for calls. TCAP is used to create database queries and invoke advanced network functionality, or links to Intelligent Network Application Part (INAP) for intelligent networks, or Mobile Application Part (MAP) for mobile services.

Notes

1. ^ a b c ITU-T Recommendation Q.700 (<http://www.itu.int/rec/T-REC-Q.700/en>)
2. ^ (Ronayne 1986, p. 145).
3. ^ (Ronayne 1986, p. 145).
4. ^ (Ronayne 1986, p. 141).
5. ^ RFC 2719 (<http://www.ietf.org/rfc/rfc2719.txt>) - Framework Architecture for Signaling Transport
6. ^ (Russell 2002, p. 318).
7. ^ (Russell 2002, p. xx).
8. ^ ITU-T Recommendation Q.700 (<http://www.itu.int/rec/T-REC-Q.700-199303-I/en/>), Section 3.2.1, p. 7.
9. ^ (Russell 2002, p. 318).
10. ^ (Russell 2002, p. 319).
11. ^ (Russell 2002, p. 433).
12. ^ ITU-T Recommendation Q.700 (<http://www.itu.int/rec/T-REC-Q.700-199303-I/en/>), p. 4.
13. ^ (Dryburgh 2004, pp. 22-23).
14. ^ (Dryburgh 2004, p. 23).
15. ^ ITU-T Recommendation Q.700 (<http://www.itu.int/rec/T-REC-Q.700-199303-I/en/>), Section 2.2.3, "Signalling modes", pp. 4-5.
16. ^ ITU-T Recommendation Q.703 (<http://www.itu.int/rec/T-REC-Q.703-199607-I/en/>), Annex A, "Additions for a national option for high speed signalling links", pp 81-86.
17. ^ (Russell 2002, p. 456).
18. ^ ITU-T Recommendation Q.711 (<http://www.itu.int/rec/T-REC-Q.711-200103-I/en/>), Section 1, "Scope and field of application", pp 1-2.

References

- Dryburgh, Lee; Jeff Hewitt (2004). *Signalling System No. 7 (SS7/C7): Protocol, Architecture, and Services*. Indianapolis: Cisco Press. ISBN 1-587-05040-4. Link to online version of text below.
- Ronayne, John P. (1986). "The Digital Network". *Introduction to Digital Communications Switching* (1st edition ed.). Indianapolis: Howard W. Sams & Co., Inc.. ISBN 0-672-22498-4.
- Russell, Travis (2002). *Signaling System #7* (4th Edition ed.). New York: McGraw-Hill. ISBN 978-0071387729.

External links

- Signaling System No. 7 (SS7/C7): Protocol, Architecture, and Services (<http://www.ss7-training.net/>). Full HTML version of the 2004 edition of the Dryburgh/Hewitt book (above).
- IEC SS7 Tutorial (<http://www.iec.org/online/tutorials/ss7/>) - IEC SS7 Tutorial covering signalling, architecture, link types, call setup and database query examples, signal unit structures.
- SS7 Protocol layer architecture overview, tutorials and discussion forum (<http://www.telecomspace.com/ss7.html>)
- SS7 - A Brief Comparison with TCP/IP (<http://www.kenneyjacob.com/2007/06/05/ss7-backbone-of-mobile-networks>)

- SS7 insights - Typical SS7 Case Studies (<http://www.squire-technologies.co.uk/case-studies.php>)
- www.protocols.com: practical overview (<http://www.protocols.com/pbook/pdf/ss7.pdf>)
- SS7 open source project (<http://www.openss7.org/>)

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