

Dial plans and routing

One page pulled from the Canon rooms named in the source column. Open the room before you quote a number in a design. This is not a copy of a Cisco document.

Check	What the room says	Source
What is the default extension length?	4 digits, and the allowed range is 2-10. [cite:1]	T2-R03-dial-plans-pstn-routing.html
What three PSTN choices can a location have?	Cisco PSTN, Cloud Connected PSTN, or premises-based PSTN. [cite:2]	T2-R03-dial-plans-pstn-routing.html
Who can place a call through cloud PSTN?	Webex Calling users. Premises-originated calls cannot. [cite:2]	T2-R03-dial-plans-pstn-routing.html
What is an ESN?	Location routing prefix plus extension, used when inter-location extension dialing is disabled. [cite:1]	T2-R03-dial-plans-pstn-routing.html
How many trunks can a route group hold?	1 to 10. [cite:2]	T2-R03-dial-plans-pstn-routing.html
What does priority level 1 on two trunks mean?	Calls are distributed randomly across them. Trunk 3 at level 2 is the fallback when a level-1 trunk is unavailable. [cite:2]	T2-R03-dial-plans-pstn-routing.html
Where may ! appear?	Once, at the end, and only on an E.164 pattern. [cite:2]	T2-R03-dial-plans-pstn-routing.html

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Sources and go deeper

Checked links were opened this month. A confirm row has no address because that page was not opened.

- **Wikipedia** [Webex \(Wikipedia\)](#) last checked 2026-10
- **Vendor doc** [Webex Help Center](#) last checked 2026-10
- **DevDocs** [Webex developer documentation](#) last checked 2026-10
- **DevDocs** [Webex developer getting started](#) last checked 2026-10
- **Community** [Webex community](#) last checked 2026-10
- **SRND** [Cisco Collaboration design zone](#) last checked 2026-10
- **RFC** [RFC 3261 SIP: Session Initiation Protocol](#) last checked 2026-10
- **Port table** [IANA service name and port number registry](#) last checked 2026-10
- **Port table** [Port reference for Webex Calling](#) last checked 2026-10
- **RFC** [RFC 3263 Locating SIP Servers](#) last checked 2026-10

- **RFC** [RFC 3264 An Offer/Answer Model with SDP](#) last checked 2026-10
- **Wikipedia** [Session Initiation Protocol \(Wikipedia\)](#) last checked 2026-10
- **RFC** [RFC 3550 RTP](#) last checked 2026-10
- **RFC** [RFC 3711 SRTP](#) last checked 2026-10
- **RFC** [RFC 4566 SDP](#) last checked 2026-10
- **RFC** [RFC 4733 RTP payload for DTMF](#) last checked 2026-10
- **Wikipedia** [Real-time Transport Protocol \(Wikipedia\)](#) last checked 2026-10
- **Release notes** [What's new in Webex Calling](#) last checked 2026-10
- **Vendor doc** [help.webex.com article njvdjf2](https://help.webex.com/article/njvdjf2) last checked 2026-10
- **Vendor doc** [help.webex.com article n0xb944](https://help.webex.com/article/n0xb944) last checked 2026-10

[sources.json](#) for this page.



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