

Cisco CCNP Advanced Call Control and Mobility Services (300-815 CLACCM) v1.2

This Cisco Advanced Call Control and Mobility Services (300-815 CLACCM v1.2) training covers how to provide collaboration solutions for a Cisco-enabled network. Use this CLACCM course to prepare for Cisco's 300-815 exam, which is one of four concentration exams you can take on the path to earning your CCNP Collaboration certification.

[CBT Nuggets course material](#) →

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