

Cisco Ip Phone 7960 Series User Guide

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8 1 Web Interface Cisco IP Phone 7960

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Cisco Ip Phone 7960 Series

Cisco 7960 is a customizable IP phone system that shares information with other network devices such as your computer and cellular phones. The Cisco 7960 is perfect for larger organizations and companies.

Cisco 7960 Manual User Guide for Cisco 7960 IP Phone Users ...

Phones in the Cisco IP Phone 7960 series have six line or speed dial buttons and phones in the 7940 series have two. Page 9 Related Topics Using the Feature Buttons, page 1-10 • Navigating on Your Phone, page 1-11 • Cisco IP Phone 7960 and 7940 Series User Guide 78-10182-08...

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The Cisco Unified IP Phone 7960G is designed to meet the communication needs of professional workers in enclosed office environments--employees who experience a high amount of phone traffic in the course of a business day.

Amazon.com : Cisco CP-7960 IP Telephone : Voip Telephones ...

Cisco 7960 7960G IP 7900 Series Phone Face Plate Bezel NEW! \$7.25. shipping: + \$3.75 shipping . Batch of 18 Cisco 7960 and 7940 IP Phones. \$120.00. shipping: + \$18.75 shipping . Tested Cisco 7960 CP-7960G VoIP Business IP Phone 7960 Unified PoE. \$23.57. shipping: + \$23.57 shipping .

Cisco IP Phone 7960 Series | eBay

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Cisco IP Phone 7960 Series | eBay

Cisco Unified IP Phone 7960G and 7940G Phone Guide for Cisco Unified Communications Manager 7.0 (SCCP) 3 Caution In European Union countries, use only external speakers, microphones, and headsets that are fully compliant with the EMC Directive [89/336/EC]. Phone Behavior During Times of Network Congestion

Cisco Unified IP Phone 7960G and 7940G for Cisco Unified ...

Cisco Headset RJ9 for Office Call Centers Telephone with Noise Canceling Microphone & Volume Mute Control for Cisco IP Phones 6941, 7841, 7942, 7945, 7960, 7961, 7962, 7965, 7970, 8841, 8845, 8945 4.1 out of 5 stars 15

Amazon.com: cisco ip phone 7960 headset

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Cisco IP Phone Models 7960 and 7940 User Guide

Cisco SIP IP Phone 7960 Administrator Guide 78-10497-02 What is the Cisco SIP IP Phone 7960? 1-5 Supported Features 1-7 Supported Protocols 1-10 Prerequisites 1-12 Cisco SIP IP Phone Connections 1-13 Connecting to the Network 1-13 Connecting to Power 1-14 Using a Headset 1-15 The Cisco SIP IP Phone with a Catalyst Switch 1-16

Cisco SIP IP Phone 7960 Administrator Guide

Cisco Unified IP Phone 7900 Series. Enhance your telephony experience. Take full advantage of high-quality VoIP while retaining the convenience and ease of use you expect from a desktop phone. Improve productivity by meeting voice-communications needs throughout your organization.

Cisco Unified IP Phone 7900 Series - Cisco

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Cisco IP Phone 7960 and 7940 Series User Guide

DailyHeadset RJ9 Cisco Phone Headset for Cisco IP Phone 6941 6945 6961 7940 7942 7945 7960 7962 7970 7971 More with Noise Cancelling Microphone,Plus 3.5 mm Connector 4.0 out of 5 stars 16 \$26.86 \$ 26 . 86

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Cisco has adopted circular design principles in its products by pioneering the use of post-consumer recycled (PCR) plastic. Select models within Cisco's iP Phone 8800 Series now reduce the use of natural resources while closing the loop with its own electronic waste.

IP Phones - Cisco

The Cisco Unified IP Phone 7960G is now obsolete (past End-of-Life and End-of-Support status).

Cisco Unified IP Phone 7960G - Retirement Notification - Cisco

IP Phone; 7960 Series; Cisco 7960 Series Manuals Manuals and User Guides for Cisco 7960 Series. We have 45 Cisco 7960 Series manuals available for free PDF download: Administrator's Manual, User Manual, Administration Manual, Getting Started, Manual, Reference Manual, ...

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To change any configuration options on a Cisco 7940 or 7960 series voip handset itself (not over tftp), you need to enter in a unlock config password. This may be needed if you want to set up the handset with a static IP address. To get to the unlock config screen, press the "Settings" button, ...

Default config unlock password on a Cisco 7940 / 7960 ...

Hello, I have a cisco 7960 phone that sometimes the calls get disconnected in the middle of the call. This the only phone having this issue and the funny part its worked fine in the last years. The phone also has the same firmware as the other 7960 phones that are work fine which 9.3 version. We...

Configure an end-to-end Cisco AVVID IP Telephony solution with an authorized self-study guide Cisco IP Telephony is based on the successful CIPT training class taught by the author and other Cisco-certified training partners. This book provides networking professionals with the fundamentals to implement a Cisco AVVID IP Telephony solution that can be run over a data network, therefore reducing costs associated with running separate data and telephone networks. Cisco IP Telephony focuses on using Cisco CallManager and other IP telephony components connected in LANs and WANs. This book provides you with a foundation for working with Cisco IP Telephony products, specifically Cisco CallManager. If your task is to install, configure, support, and maintain a CIPT network, this is the book for you. Part I of Cisco IP Telephony introduces IP telephony components in the Cisco AVVID environment. Part II covers basic CIPT installation, configuration, and administration tasks, including building CallManager clusters; configuring route plans, route groups, route lists, route patterns, partitions, and calling search spaces; configuring and managing shared media resources such as transcoders, conference bridges, and music on hold; configuring and managing Cisco IP Phone features and users; configuring IP telephony component hardware and software; automating database moves, adds, and changes using the Bulk Administration Tool (BAT); and installing, upgrading, and creating backups for Cisco CallManager components. Part III deals with advanced CIPT configuration tasks for call preservation and shared media resources; covers distributed and centralized call processing model design in WAN environments; explains how to deploy Survivable Remote Site Telephony (SRST) to provide local call processing redundancy at remote branch sites; and provides tips, guidelines, and rules for deploying a Cisco IP Telephony solution, culled from seasoned practitioners in the field. Part IV focuses on three of the primary Cisco applications designed for integration in a Cisco CallManager environment-Cisco WebAttendant, Cisco IP SoftPhone, and Cisco Unity. All this detailed information makes Cisco IP Telephony an ideal resource for the configuration and management of a Cisco IP Telephony solution. Cisco IP Telephony offers indispensable information on how to Configure and implement an end-to-end IP telephony solution using Cisco CallManager and CIPT devices to converge your voice and data networks Create, configure, and manage Cisco CallManager clusters to support small user environments as well as larger user environments with up to 10,000 users Optimize routing flexibility into your CIPT network design using route plans Ensure telephony class of service with partitions and calling search spaces Effect moves, adds, and changes on a large number of users and devices quickly and efficiently Perform proper installation, upgrade, and backup of Cisco CallManager clusters Monitor and perform troubleshooting tasks for a CIPT solution David Lovell is an educational specialist at Cisco Systems(r), Inc., where he designs, develops, and delivers training on CIPT networks. David is experienced in design and implementation of IP telephony systems and has been instructing students for six years, two of which have been focused solely on IP

Create applications that deliver interactive content to Cisco IP Phones Learn information and techniques vital to building and integrating third-party services for Cisco IP Phones Understand the development process using XML and HTTP client and server applications to successfully build a service Discover advanced services information about objects, advanced runtime generation, and other XML development tools Utilize the provided CallManager Simulator to support an IP phone for development purposes Get the most out of your IP phone systems with strategies and solutions direct from the Cisco team Services on Cisco IP Phones help you enhance productivity, gain the competitive advantage, and even help generate

revenue. Services are simply applications that run on the phone rather than on a PC or a web browser. By developing services tailored to your particular needs, you can achieve unlimited goals. Cisco AVVID IP Telephony provides an end-to-end voice-over-IP solution for enterprises. Part of that solution are Cisco IP Phones, a family of IP-based phones. Cisco IP Phones feature a large display, an XML micro browser capable of retrieving content from web servers, and the ability to deploy custom services tailored to your organization's or enterprise's needs. Developing Cisco IP Phone Services uses detailed code samples to explain the tools and processes used to develop custom phone services. You'll learn about XML, CallManager, Cisco IP Phones, and the history behind why Cisco chose XML to deploy phone services. You'll find detailed information to help you learn how to build a service, how to build a directory, and how to integrate your service with Cisco CallManager. This book complements and expands on the information provided in the Cisco IP Phone Services Software Developer's Kit (SDK). With the information in this book, you can maximize your productivity using the tools provided in the SDK and the custom tools provided on the companion CD-ROM. Beginner and advanced service developers alike benefit from the information in this book. Developing Cisco IP Phone Services represents the most comprehensive resource available for developing services for Cisco IP Phones. Companion CD-ROM The CD-ROM contains the sample services that are covered in the book, development utilities from the Cisco IP Phone Services SDK, and new tools written specifically for this book such as XML Validator. One of the most useful applications on the CD-ROM is the CallManager Simulator (CM-Sim). CM-Sim significantly lowers the requirements for service development. You only need a Windows-based PC with CM-Sim and a web server running, and one Cisco IP Phone 7940 or 7960. This book is part of the Cisco Press Networking Technologies Series, which offers networking professionals valuable information for constructing efficient networks, understanding new technologies, and building successful careers.

Authorized Self-Study Guide Designing for Cisco Internetwork Solutions (DESGN) Second Edition Foundation learning for CCDA exam 640-863 Designing for Cisco Internetwork Solutions (DESGN), Second Edition, is a Cisco®-authorized, self-paced learning tool for CCDA® foundation learning. This book provides you with the knowledge needed to design enterprise networks. By reading this book, you will gain a thorough understanding of designing routed and switched network infrastructures and services within a modular architecture. In Designing for Cisco Internetwork Solutions (DESGN), Second Edition, you will study a broad range of network design principles and guidelines. You will learn about network design in the context of the Cisco Service-Oriented Network Architecture (SONA) framework and the Cisco Enterprise Architecture. Specific topics include campus and data center infrastructure, remote connectivity, IP addressing design, routing protocol selection, voice network design, wireless network design, and including security in your designs. An ongoing case study plus chapter-ending review questions illustrate and help solidify the concepts presented in the book. Whether you are preparing for CCDA certification or simply want to gain a better understanding of network design principles, you will benefit from the foundation information presented in this book. Designing for Cisco Internetwork Solutions (DESGN), Second Edition, is part of a recommended learning path from Cisco that includes simulation and hands-on training from authorized Cisco Learning Partners and self-study products from Cisco Press. To find out more about instructor-led training, e-learning, and hands-on instruction offered by authorized Cisco Learning Partners worldwide, please visit www.cisco.com/go/authorizedtraining. Diane Teare is a professional in the networking, training, and e-learning fields. She has more than 20 years of experience in designing, implementing, and troubleshooting network hardware and software and has also been involved in teaching, course design, and project management. She has extensive knowledge of network design and routing technologies and is an instructor with one of the largest authorized Cisco Learning Partners. Understand the Cisco vision of intelligent networks and the SONA framework Learn how to structure and modularize network designs within the Cisco Enterprise Architecture Design basic campus and data center networks Build designs for remote connectivity with WAN technologies Create IPv4 addressing schemes Understand IPv6 design Select the appropriate routing protocol for various modules in the Cisco Enterprise Architecture Design basic VoIP and IP telephony networks Understand wireless design principles Build security into your network designs This volume is in the Certification Self-Study Series offered by Cisco Press®. Books in this series provide officially developed self-study solutions to help networking professionals understand technology implementations and prepare for the Cisco Career Certifications examinations. Category: Cisco Press—Network Design Covers: CCDA Exam 640-863

Now fully updated for Cisco's new CIPTV2 300-075 exam, Implementing Cisco IP Telephony and Video, Part 2 (CIPTV2) Foundation Learning Guide is your Cisco® authorized learning tool for CCNP® Collaboration preparation. Part of the Cisco Press Foundation Learning Series, it teaches advanced skills for implementing a Cisco Unified Collaboration solution in a multisite environment. The authors show how to implement Uniform Resource Identifier (URI) dialing, globalized call routing, Intercluster Lookup Service and Global Dial Plan Replication, Cisco Service Advertisement Framework and Call Control Discovery, tail-end hop-off, Cisco Unified Survivable Remote Site Telephony, Enhanced Location Call Admission Control (CAC) and Automated Alternate Routing (AAR), and important mobility features. They introduce each key challenge associated with Cisco Unified Communications (UC) multisite deployments, and present solutions-focused coverage of Cisco Video Communication Server (VCS) Control, the Cisco Expressway Series, and their interactions with Cisco Unified Communications Manager. Each chapter opens with a topic list that clearly identifies its focus, ends with a quick-study summary of key concepts, and presents review questions to assess and reinforce your understanding. The authors present best practices based on Cisco Solutions Reference Network Designs and Cisco Validated Designs, and illustrate operation and troubleshooting via configuration examples and sample verification outputs. This guide is ideal for all certification candidates who want to master all the topics covered on the CIPTV2 300-075 exam. Shows how to craft a multisite dial plan that scales, allocates bandwidth appropriately, and supports QoS Identifies common problems and proven solutions in multisite UC deployments Introduces best practice media architectures, including remote conferencing and centralized transcoding Thoroughly reviews PSTN and intersite connectivity options Shows how to provide remote site telephony and branch redundancy Covers bandwidth reservation at UC application level with CAC Explains how to plan and deploy Cisco Device Mobility, Extension Mobility, and Unified Mobility Walks through deployment of Cisco Video Communication Server and Expressway series, including user and endpoint provisioning Covers Cisco UCM and Cisco VCS interconnections Shows how to use Cisco UC Mobile and Remote Access Covers fallback methods for overcoming IP WAN failure Demonstrates NAT traversal for video and IM devices via VCS Expressway Introduces dynamic dial plan learning via GDPR, SAD, or CCD

Delivers the proven solutions that make a difference in your Cisco IP Telephony deployment Learn dial plan best practices that help you configure features such as intercom, group speed dials, music on hold, extension mobility, and more

Understand how to manage and monitor your system proactively for maximum uptime Use dial plan components to reduce your exposure to toll fraud Take advantage of call detail records for call tracing and accounting, as well as troubleshooting Utilize the many Cisco IP Telephony features to enable branch site deployments Discover the best ways to install, upgrade, patch, and back up CallManager Learn how backing up to remote media provides both configuration recovery and failure survivability IP telephony represents the future of telecommunications: a converged data and voice infrastructure boasting greater flexibility and more cost-effective scalability than traditional telephony. Having access to proven best practices, developed in the field by Cisco® IP Telephony experts, helps you ensure a solid, successful deployment. Cisco CallManager Best Practices offers best practice solutions for CallManager and related IP telephony components such as IP phones, gateways, and applications. Written in short, to-the-point sections, this book lets you explore the tips, tricks, and lessons learned that will help you plan, install, configure, back up, restore, upgrade, patch, and secure Cisco CallManager, the core call processing component in a Cisco IP Telephony deployment. You'll also discover the best ways to use services and parameters, directory integration, call detail records, management and monitoring applications, and more. Customers inspired this book by asking the same questions time after time: How do I configure intercom? What's the best way to use partitions and calling search spaces? How do I deploy CallManager regionally on my WAN? What do all those services really do? How do I know how many calls are active? How do I integrate CallManager with Active Directory? Years of expert experiences condensed for you in this book enable you to run a top-notch system while enhancing the performance and functionality of your IP telephony deployment.

In *The Implosion of Capitalism* world-renowned political economist Samir Amin connects the key events of our times - financial crisis, Eurozone implosion, the emerging BRIC nations and the rise of political Islam - identifying them as symptoms of a profound systemic crisis. In light of these major crises and tensions, Amin updates and modifies the classical definitions of social classes, political parties, social movements and ideology. In doing so he exposes the reality of monopoly capitalism in its contemporary global form. In a bravura conclusion, Amin argues that the current capitalist system is not viable and that implosion is unavoidable. *The Implosion of Capitalism* makes clear the stark choices facing humanity - and the urgent need for a more humane global order.

Implementing Cisco Unified Communications Manager, Part 2 (CIPT2), Second Edition is a Cisco®-authorized, self-paced learning tool for CCNP Voice® foundation learning. This book provides you with the knowledge needed to install and configure a Cisco Unified Communications Manager solution in a multisite environment. By reading this book, you will gain a thorough understanding of how to apply a dial plan for a multisite environment, configure survivability for remote sites during WAN failure, and implement solutions to reduce bandwidth requirements in the IP WAN. This book focuses on Cisco Unified Communications Manager (CUCM) Release 8.x, the call routing and signaling component for the Cisco Unified Communications solution. The book has been fully updated and includes new coverage of topics such as Service Advertisement Framework (SAF), and Call Control Discovery (CCD). Whether you are preparing for CCNP Voice certification or simply want to gain a better understanding of deploying Cisco Unified Communications Manager in a multisite environment, you will benefit from the foundation information presented in this book. *Implementing Cisco Unified Communications Manager, Part 2 (CIPT2), Second Edition*, is part of a recommended learning path from Cisco that includes simulation and hands-on training from authorized Cisco Learning Partners and self-study products from Cisco Press. To find out more about instructor-led training, e-learning, and hands-on instruction offered by authorized Cisco Learning Partners worldwide, please visit www.cisco.com/go/authorizedtraining. Chris Olsen, CCVP, and CCNP, along with numerous other Cisco voice specializations, Microsoft, VMware, and Novell certifications, has been an independent IT and telephony consultant, author, and technical editor for more than 15 years. He has been a technical trainer for more than 19 years and has taught more than 60 different courses in Cisco, Microsoft, VMware, and Novell. For the last seven years he has specialized in Cisco, and recently Microsoft Unified Communications along with VMware virtualization and Cisco data center technologies. He has done a wide array of IT and telephony consulting for many different companies.

- Identify multisite issues and deployment solutions
- Implement multisite connections
- Apply dial plans for multisite deployments
- Examine remote site redundancy options
- Implement Survivable Remote Site Telephony (SRST) and Media Gateway Control Protocol (MGCP) Fallback
- Implement CUCM Express in SRST mode
- Implement bandwidth management and call admission control (CAC)
- Configure device and extension mobility
- Apply Service Advertisement Framework (SAF) and Call Control Discovery (CCD)

This volume is in the Foundation Learning Guide Series offered by Cisco Press®. These guides are developed together with Cisco as the only authorized, self-paced learning tools that help networking professionals build their understanding of networking concepts and prepare for Cisco certification exams.

Authorized self-study guide for voice over data network foundation learning This book will help you to: Configure Voice over Frame Relay, ATM, or IP using Cisco IOS(r) software Analyze existing voice hardware/software, and select the Cisco multiservice access devices that best serve your needs Analyze existing branch and regional office voice networks and services, and choose the optimum transmission method for voice traffic: Frame Relay, ATM, or IP Learn the fundamentals of VoFR, VoATM, and VoIP standards, protocols, and the Cisco hardware that supports these services Learn the basics of the Architecture for Voice, Video, and Integrated Data (AVVID) including CallManager, Cisco IP Phones, and related voice gateway equipment Design, configure, integrate, and optimize an enterprise network in remote branch and regional offices by using integrated access technology that combines voice and data transmission over Frame Relay, ATM, and IP connections, access devices, and CIPT client hardware Learn the fundamentals of PBXs, and apply the principles and concepts to develop a process for integrating Cisco equipment with PBXs and for replacing PBXs Cisco Voice over Frame Relay, ATM, and IP teaches you the Cisco solutions for voice technology (VoIP, VoFR, VoATM). This complete solutions guide helps you analyze existing voice hardware and software and select the Cisco multiservice access devices that best serve the needs of your network environment. In addition to learning how to design, configure, integrate, and optimize networks in remote branch and regional offices, this book also provides you with a fundamental understanding of PBXs, enabling you to develop a process for integrating Cisco equipment with or replacing PBXs. Cisco Voice over Frame Relay, ATM, and IP prepares you for voice and data integration by teaching you how to install and configure Cisco voice and data network routers; how to configure Cisco voice-enabled equipment for Voice over Frame Relay, ATM, and IP; how to configure voice ports, dial peers, and special commands to enable voice transmission over a data network; and how to perform voice traffic analysis to determine how to improve the quality of service (QoS) for delay-sensitive voice traffic. This book features actual router output and configuration examples to aid in the discussion of the configuration of these technologies. At the end of

each chapter your comprehension is tested by review questions. Cisco Voice over Frame Relay, ATM, and IP has all of the tools you need to vastly improve your understanding of the Cisco solution to voice networking needs. Cisco Voice over Frame Relay, ATM, and IP is part of a recommended self-study program from Cisco Systems(r) that includes simulation and hands-on training from authorized Cisco Learning Partners, and self-study products from Cisco Press. To find out more about instructor-led, e-learning, and hands-on instruction offered by authorized Cisco Learning Partners, please visit www.cisco.com/go/authorizedtraining. This volume is in the Certification Self-Study Series offered by Cisco Press(r). Books in this series provide officially developed self-study solutions to help networking professionals understand technology implementations and prepare for the Cisco Career Certifications examinations.

Annotation Strategies for configuring, monitoring, and troubleshooting new Cisco telephony software! First book with specific coverage of Cisco CallManager written by its key developers. Includes specific configuration examples, configuration guidelines, troubleshooting tips, and case studies. Provides detailed information about such complex issues as Cisco CallManager routing and diagnostics. Cisco CallManager Fundamentals provides reference information about Cisco CallManager. This book fully details the innerworkings of Cisco CallManager, which will empower those responsible for designing and maintaining the system with the availability to make intelligent decisions about what, when, and how features within Cisco CallManager can be used. John Alexander is a software development manager for Cisco Systems. John managed the development of the call processing softwares as well as software development tasks. Chris Pearce has been a software engineer in telecommunications for the past nine years. In 1994 he was one of the first four engineers that designed and implemented what would eventually become the Cisco CallManager. Anne Smith is a senior technical writer at Cisco Systems, author of over two-dozen user guides, online help files, and Web-based documentation for various software and telephony companies. Delon Whetten is the technical lead of the Cisco CallManager software group at Cisco Systems. He has been involved in the design and development of message switching, voice messaging, video teleconferencing, and Voice over IP call management systems for the last 24 years.

A complete IP Telephony migration planning guide Includes Steps to Success Poster It's everyone's "must have." This is a reference book for the entire project team who works on the deployment of an IP Telephony solution. Take advantage of best practices. Includes more than 200 best practices, lessons learned, and tips for getting you through your IP Telephony deployment successfully. Minimize risk and learn from the mistakes of others. Read the list of the top 10 things that can go wrong during an IP Telephony deployment. Ask the right questions. Get the project team thinking and collaborating together with Stephanie's "Checklist of Questions to Ask the Project Team." Use proven planning tools. Work from sample checklists, templates, project plans, and workflow documents to guide your planning process. Keep the Steps to Success on the minds of your project team. Use the enclosed poster, which illustrates every major step associated with an IP Telephony deployment. There is no better path to the successful implementation of a new technology than to follow in the experienced footsteps of an organization that has already been there. The Road to IP Telephony tells you how Cisco Systems successfully moved its own organization to a converged, enterprise-wide network. You will learn the implementation and operational processes, what worked, what didn't work, and how to develop your own successful methodology. After presenting this topic to hundreds of Cisco customers, including Fortune 500 companies, Stephanie Carhee consistently encountered the same question, "If I decide to move to IP Telephony, where do I begin and what can I do to ensure that I do it right the first time?" Although the needs of every enterprise are different, some things are universal; planning, communication, teamwork, and understanding your user's requirements are as important as technical expertise. The Road to IP Telephony shares with you everything you need to know about managing your deployment. It starts with where to begin, including what needs to be addressed before you even begin the planning process, to building your project team. Key best practices are also offered to help you set the project's pace and schedule, get your users on board, identify a migration strategy, develop a services and support strategy, and work toward the final PBX decommission. "Cisco IT wants to share its implementation experience with Cisco customers and partners to aide in the deployment practices of new Cisco technologies. While conducting our own company-wide cutover, we learned a great deal about what to do and what not to do. This book shares our experiences." -Brad Boston, Senior Vice President and Chief Information Officer, Cisco Systems, Inc. This volume is in the Network Business Series offered by Cisco Press. Books in this series provide IT executives, decision makers, and networking professionals with pertinent information on today's most important technologies and business strategies.

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