

## <<VOIP技术与应用>>

### 图书基本信息

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### 内容概要

本书主要内容包括：VoIP的基础知识，话音在IP网的传输，高速编码技术，H.323协议，SIP协议，VoIP与SS7，相关的业务质量(QoS)，接入网技术，在IP网中发送传真的技术等。

本书主要读者对象为从事通信的工程技术人员及大专院校相关专业师生。

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