Internet Telephony

Mahbub Hassan et al.

Presenter: Yi Sun

Outline

- 1. Introduction
- 2. Applications and Services
- 3. Technical Challenges

1. Introduction

- Circuit switching and packet switching are the two main technologies for computer and telecommunication networks.
- Because Internet supports data communications, a range of other services can be bundled together with Internet telephony.
- However, Internet is initially designed for non-real-time data communications, so it poses several technical challenges that must be overcome.

2. Applications and Services

Video Telephony

It's easier to support video telephony over IP telephony.

Sound Grading

IP telephony can support high-quality stereo and surround sound.

Unified Messaging

Users can get all the messages sent to one location from which he can access it at his convenience.

Web-based Call Centers

It captures potential customers' attention by making direct calls when they are browsing the Internet.

Packet Loss:

Network Upgrade

High speed transmission technologies include ATM (asynchronous transfer mode) for mb/s, SONET (synchronous optical network) for gb/s, WDM (wavelength-division multiplexing) for tb/s.

Silence Substitution

Substitute silence in place of a missing packet.

Noise Substitution

Substitute white (background) noise for lost packets.

Packet Repetition

Replay the last correctly received packet for lost packet.

Packet Interpolation

Use the characteristics of the packets in the neighborhood of the lost one to produce a replacement.

Frame Interleaving

The effect of packet loss can be reduced by interleaving voice frames across different packets.

The loss of a single packet will only result in multiple short gaps in different streams of the received data.

Packet Delay:

Codec Delay

Time to convert analog voice to digital data and vice versa.

Serialization Delay

Time to place a packet on the transmission line.

Queuing Delay

Occurs at various switching and transmission points of the network, such as routers and gateways.

Propagation Delay

Time required by signals to travel from one point to another.

Network Jitter:

- Jitter is the variance in the interframe arrival times at the receiver.
- To allow for variable packet arrival times and still achieve steady stream of packets, the receiver holds the first several packets in a jitter buffer for a while before playing them out.
- Selection of jitter buffer size is crucial to IP telephony systems.

Thanks!

References

 Mahbub Hassan and Alfandika Nayandoro, et al. Internet Telephony: Services, Technical Challenges, and Products, IEEE Communications Magazine, April 2000.