A New Step on Transforming VoIP into the Ubiquitous Service Era

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ABSTRACT

Over the coming years, operators will be using IP networks to replace their current PSTN. VoIP opens doors of opportunities such as cost saving and increased flexibility as well as for the uninvited users, viruses, worms and other unexpected threats from the network. In this paper, I have described benefits and threats that VoIP should face to get the leading place in providing the public phone service. By deploying a mobile wireless networking technology, we can step into the ubiquitous VoIP era. However, only talking about the service model and high-technology challenges is not enough. More facets should be tackled like regulatory requirements.

Keyword: VoIP, PSTN, service model
INTRODUCTION

During the past 10 years, the Internet has been widely adopted by both vendors and consumers. That led to the creation of the electronic marketplace, which swept out business barriers such as time, geography, language, currency, and culture. Consumer goods on the Web can be reviewed by millions of customers worldwide and be purchased at a low cost.

Except services that usually accompany the brick-and-mortar product purchasing context, the online stores offer a great number of Web-based services such as product reviews, product comparison tools, personalized shipping, personal offering, and other services that are tailored to individual customers (Chellappa and Kumar, 2005). Products on the Web are not just generic products but they are actually augmented products (Levitt, 1980), and this augmentation is the result of services that envelop a product to create a consumer’s product purchasing experience. In a telecommunication industry, consumer selects a company that offers other services, except phone call service, he or she would like to have at the lowest price. With the rapid emergence of a new voice-over-IP (VoIP) service, the range of services has been expanded and many of them are “free”. This exerts more pressure on telecommunication industry and impacts on the revenue growth (Shin, 2006).

The objective of this study is to describe the VoIP conceptual and service models. There are benefits and threats that VoIP should face to get the leading place in providing the public phone service in ubiquitous era. The VoIP service model is described by the Levitt's model (Levitt, 1980).

LITERATURE BACKGROUND OF VoIP

*How VoIP works*

IP technology treats voice as an application, destroying the old distinctions between “voice” and “data” that are a standard part of Public Switched Telephone Network (PSTN). With VoIP,
telephony signals are digitized and transmitted as packets to their destination, just as an email, streaming video or any other kind of IP transaction. Figure 1 depicts a simplified VoIP network layout.

A VoIP transmission can be performed among terminals or endpoints, a means of setting up the call, through the gateways that connect different networks. The endpoint is the phone - a traditional phone with an adapter, a dedicated IP-phone, or computer with a microphone. Call signaling is carried out by a call processing manager or “IP-PBX” (Private Branch eXchange), which sets up the call, handles routing, and provides configurations to endpoints. The two major signaling protocols are H.323 and Session Initiation Protocol (SIP). Both signaling protocols use known ports or ranges of ports for call setup, but the actual conversation takes place over high UDP ports negotiated on the fly (Edelson, 2005).

A gateway compresses and packetizes voice data and sends it to the IP network. It must translate VoIP signaling protocols to SS7,
the signaling protocol used in the PSTN. To make a call – or access voice mail, or have a telephone number – a VoIP client registers with an IP-PBX. The client sends a request to a gateway which resolves the telephone number to a network address. When a connection is established using one of the signaling protocols, the caller’s voice is digitized, compressed, possibly encrypted, and packetized using Real Time Protocol (RTP). RTP packets are then wrapped in UDP datagrams. These travel directly between the participants and are re-assembled by a voice processing application, based on sequence numbers and timestamps in their headers (Edelson, 2005). There are four different types of connections providing VoIP (Kim and Parameswaran, 2004):

- IP to PSTN trough Internet
- IP to PSTN without Internet
- IP to IP
- PSTN to PSTN.

**Security issues**

As well as introducing opportunities for the wide use of VoIP, increased flexibility and cost savings, IP networks also introduce a number of security concerns. These include traditional threats such as hacking and viruses, as well as those associated with the voice environment, including privacy, service theft, denial of service attacks, and service reliability (Edelson, 2005; Rowe, 2005; Whitworth, 2006).

Even though a company is not able to implement 100% military level security, it can choose the appropriate security measures to mitigate the risk that is related with VoIP. The main requirements for VoIP security (Edelson, 2005):

- VoIP network components should be dedicated, both for security and performance.
- Voice and data traffic should be separated.
- Firewalls are needed where traffic might legitimately flow between voice and data networks.
Simple packet filtering should be fully done where no cross-traffic is allowed.

Wireless network presents more challenges for VoIP security.

**Benefits**

The main reason of why people and organization choose to VoIP is cost. Cost benefits associated with a move to VoIP typically reach savings of around 30% (Whitworth, 2006). Along with the traditional phone service, VoIP providers introduce new innovation such as area code mobility, real-time billing and service provisioning, and easy conference calling. Personal digital assistant (PDA) allows consumers to send voice over a WiFi to the Internet. Broadcasting content can be viewed over cellular phones or other dedicated terminal by implementing digital multimedia broadcasting. Each of this embeds IP voice and can be regarded as VoIP (Shin, 2006).

**VoIP service model**

A product on the Web is more than the core value it provides or its price; rather, it is the bundle of attributes, services and perceived benefits that accompanies its purchase (Chellappa and Kumar, 2005). This view has been adopted from Levitt (1980) from his observation of physical markets: “In the actual world of markets, nothing is exempt from other considerations, even when price competition rages” In this paper I adapted Levitt’s total product concept to describe the VoIP service in Figure 2:

1) the generic service sought by customers – telecommunication;

2) the expected service, which represents the customer minimal purchase conditions – “Free” PC to PC communication, including calls, conferences;

3) the augmented service, which is the addition of extra or unprompted augmentations or benefits to the expected
product – “Free” PC to Phone, Phone to PC and Phone to Phone communication over IP;
4) the potential service, which refers to everything potentially feasible to attract and keep customers – Ubiquitous “Free” VoIP by mobile IP

On the market of VoIP providers, a lot of companies have emerged that offer PC to PC communication for free. Skype, owned by online auctioneer eBay, is undoubtedly the leader. Competitors such as Yahoo, AOL, MSN, NateOn are integrating more call features into their instant-messaging clients in order to catch up with the leader (Marguerite, 2006). We can regard this as an expected service. By offering the service for free, companies try to involve as many potential customers as it is possible, who can be charged for the extended features. More vivid combat among vendors for the customers is going on in the augment service area, since for this services customer has to pay.

The following list shows more popular vendors that offer augment services:
- Skype
• Vonage
• Yahoo + Dialpad
• AOL
• WOWCALL
• Internetcalls

The list is far more incomplete, since more and more companies spring up on the horizon. They try to attract more customers by offering competitive conditions such as low-rate per minute plans or monthly plans for unlimited calls. Graph 1 shows the prices that different VoIP vendors offer today (Dacom 2006; Skype 2006; Wowcall 2006).

<table>
<thead>
<tr>
<th>Country</th>
<th>Skype</th>
<th>WOWCALL</th>
<th>Dacom 002</th>
</tr>
</thead>
<tbody>
<tr>
<td>U.S.</td>
<td>2.4</td>
<td>8</td>
<td>15</td>
</tr>
<tr>
<td>Japan</td>
<td>2.6</td>
<td>14</td>
<td>30</td>
</tr>
<tr>
<td>China</td>
<td>2.4</td>
<td>19</td>
<td>40</td>
</tr>
<tr>
<td>Australia</td>
<td>2.4</td>
<td>12</td>
<td>38</td>
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</tbody>
</table>

Graph 1 VoIP price comparison

Recently Skype started offering a free calling to phones in U.S. and Canada till the end of the year (Marguerite, 2006). In above cases customers cannot use services if they are on the move. Future generation wireless networks can enable mobile users to switch network access and experience uninterrupted service continuity anywhere, anytime. If this happens,
telecommunication companies will face tremendous competition, since VoIP offerings will be at significant lower cost.

Presently, telecommunication companies offer a broadband access to the internet over general packet radio service (GPRS), which is still pretty expensive. On the other hand, mobile wireless technology has gained tremendous popularity due to its ability to provide ubiquitous information access to users on the move (Siddiqui and Zeadally, 2006). Since wireless network technologies are not homogeneous, the next generation should face challenges of constituting different types of access technologies (Deshpande, 2002). The heterogeneity that will characterize future wireless systems instigates the development of intelligent and efficient handoff management mechanisms that can provide seamless roaming capability to end-users moving between several different access networks. And it can be called a ubiquitous VoIP. Ubiquitous VoIP is a potential service that can bring a competitive advantage to a company on the telecommunication arena.

CONCLUSION

In this article, we have presented overall overview of VoIP, benefits and risks that are accompanied while using VoIP. Over the coming years, operators will be using IP networks to replace their current PSTN. VoIP eventually can become the dominant provider of public phone service. By deploying a mobile wireless network technology, we can step into the ubiquitous VoIP era. But talking about the service model and high-technology challenges only is not enough. More facets should be tackled like regulatory requirements.

Some countries prohibit VoIP service where governments have monopolies in telephone and telecommunication service. Other restricts it to communication from computer to computer. Another countries permit without putting any technological standards or quality of service requirements (Shin, 2006). In the U.S., FCC has imposed the e911 requirement on VoIP providers that obliges
them to support emergency calls (Federal Communication Commission, 2004).

REFERENCES


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