

SIP-to-SIP Call Diversion Service for Digital Link Backup



AddPac

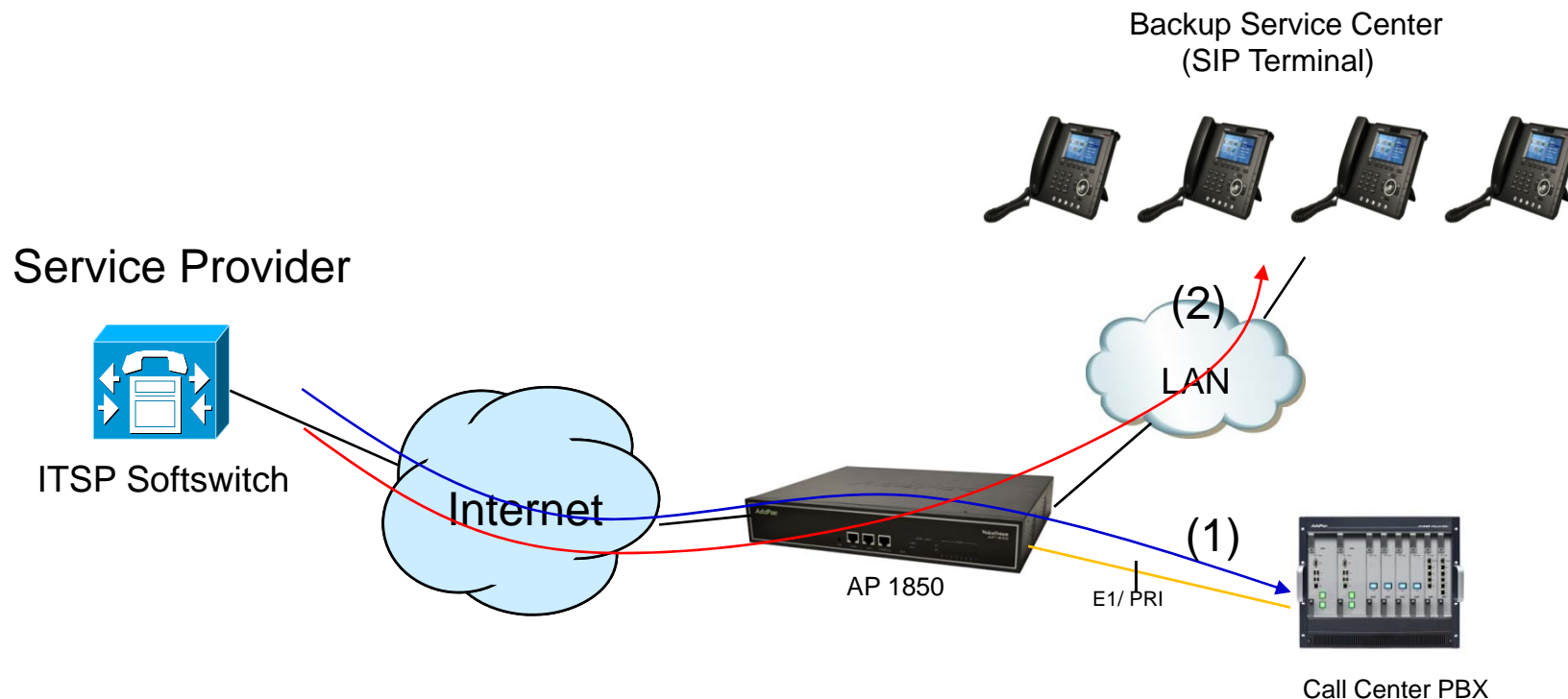
AddPac Technology

2014, Sales and Marketing

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SIP-to-SIP Call Diversion Service for Digital Link Backup

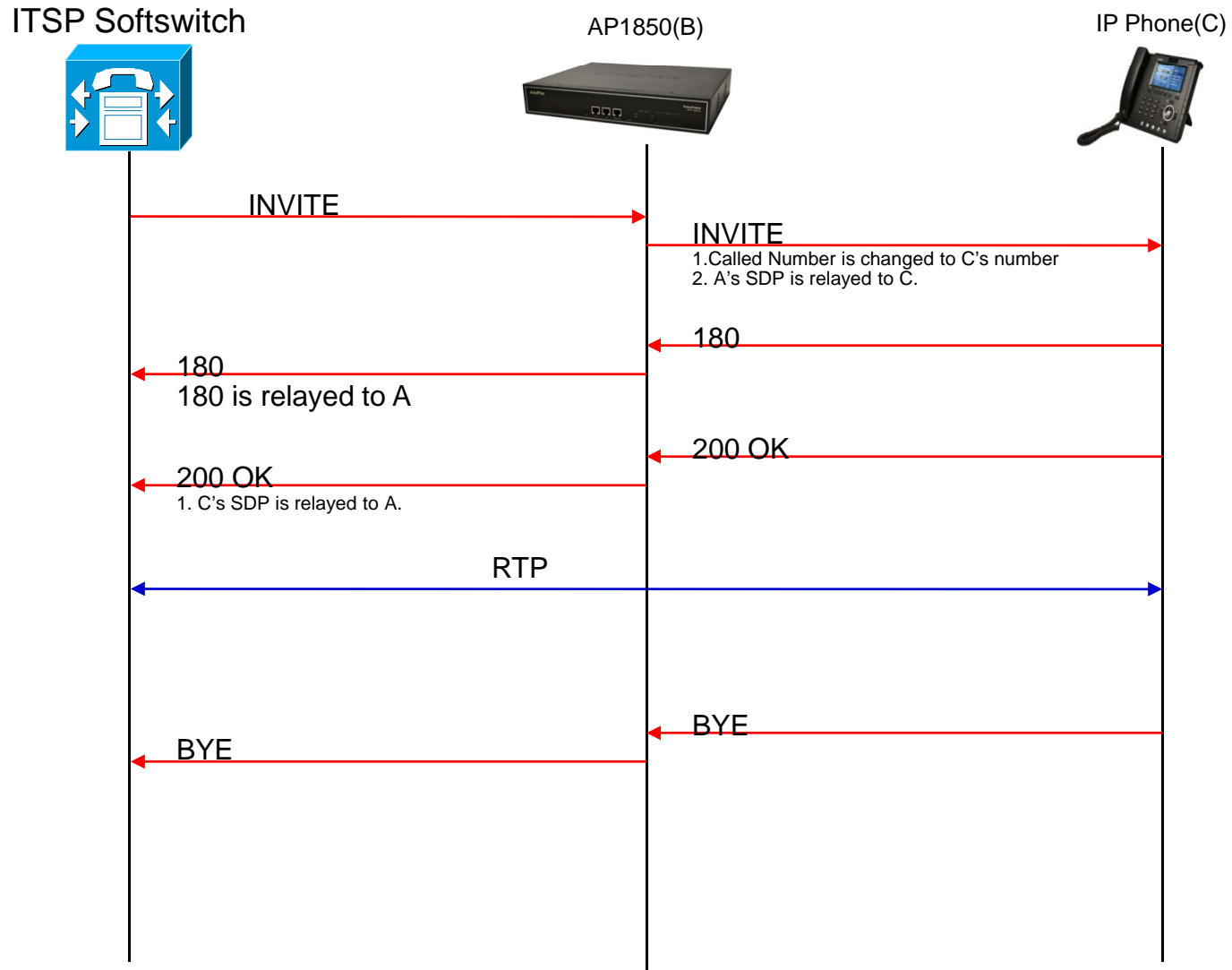


- (1) SIP Calls are routed to Call Center PBX via digital line when E1/PRI link is normal state.
- (2) SIP Calls are routed to SIP terminal based Backup Service Center when E1/PRI link is down

* All Phones have same numbers and different IP address in Backup service Center.

* Calls for Backup Service Center are routed by the algorithm(least selected number).

Signal Call Flow



VoIP(SIP) Call Diversion Configuration(1/2)

System

Basic

- Protocol
- Server SIP
- Server H.323
- SIP Registration
- H.323 Registration
- E1/T1 Trunk
- FXS Extension
- FXO/EnM Extension
- **E1/T1 Extension**
- DTMF/CODEC
- VoIP Dial Plan
- FXO DialPlan
- Static Route
- Hot Line
- SS7

Advanced

- Port Control
- Fax
- Service
- Filtering
- Security
- SNMP

E1/T1 Extension

Port Information

Port	P0	P1	P2	P3	P4	P5	P6	P7
SLOT 0	FXS	FXS	FXS	FXS	FXS	FXS	FXS	FXS
SLOT 1	E1	E1						

- Enable E1 Fault Call Diversion .
- Diversion Tag Identifier.

E1/T1 Extension Configuration

Pots Num	Port	Group	Numbers	HuntStop	E1 Fault Diversion	Forward Digits(0~99)	Control
2568	1/0	0	T	0	0		<input type="checkbox"/>

P1:0 0 None from last

-Configure IP Phone's number

VoIP(SIP) Call Diversion

ID	Called Number	Control
0	1800	<input type="checkbox"/>

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Thank you!

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