

AARNet RTC infrastructure review

 Background
 Existing network
 Motivation for change
 Concept
 New network
 Realisation
 ITSPs, SBCs
 Connecting and rules of engagement
 Moving forward - Services
 Feedback

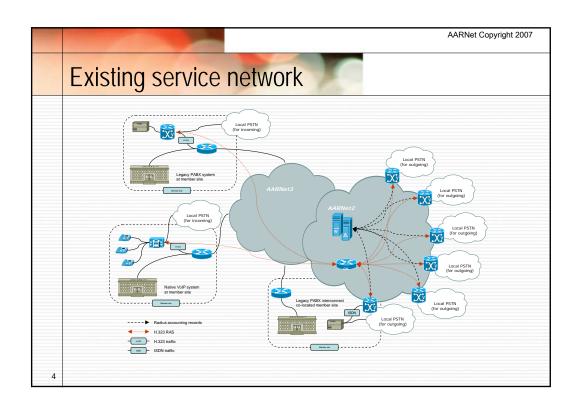
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Background

- · AARNet VoIP service
 - Established around 8 years ago following a CSIRO project for voice over IP services
 - Essentially a toll bypass service
 - VoIP access via member connection to the AARNet IP network;
 - PSTN access via H323 voice gateways distributed nationally;
 - · H323 gatekeepers used for routing and call admission control;
 - · Charging mechanisms based on radius;
 - Web based user portals for reporting, billing, and performance monitoring.
- · Ad-hoc growth for video, multipoint, others

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Motivation for change

• Existing network

- Single service only

- Runs on A2, aged equipment

- Difficult to manage, complicated charging mechanism

- Limited flexibility, lacking LI capabilities

• Desired network

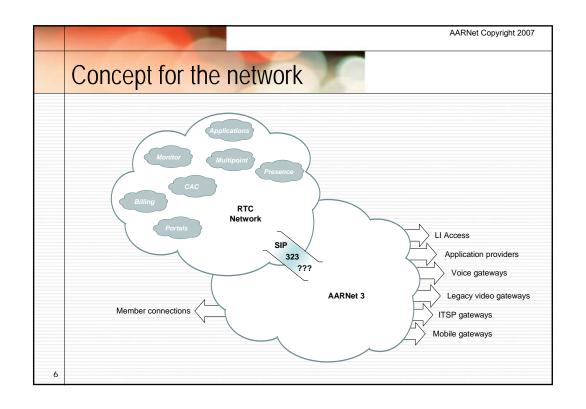
- Multiple services

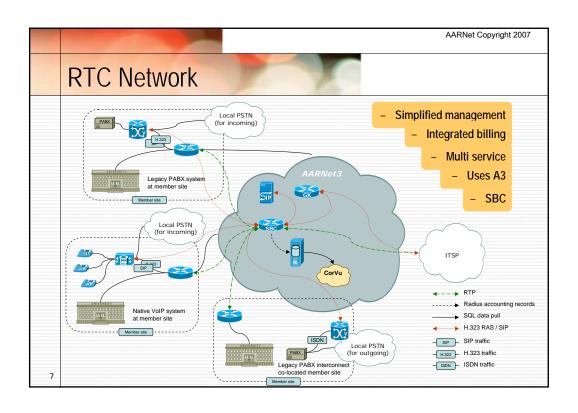
- Multiple protocols

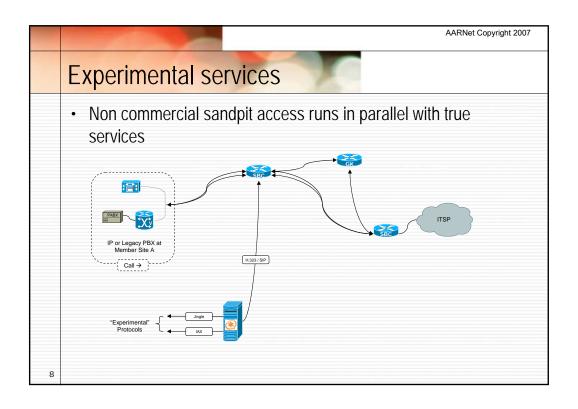
- Overlay on A3

- Cost effective, scalable, integrated

- Provides for LI







AARNet Copyright 2007 Realisation August 2007, ITSP interconnect trial SBC gateway for voice Route test traffic only (via internet) Connect SBC radius records with billing system - Reconcile carrier billing with AARnet records October 2007, direct connection to ITSP Evaluate service quality November 2007, migrate member traffic via ITSP Decommission ISDN voice gateways December 2007, gatekeeper consolidation, SBC for member access Video and AUroot gatekeepers merge - Introduce SBC on member edge 2008 - SIP on the member edge - New services

Sandpit

AARNet Copyright 2007 **ITSPs** IP Telephony Service Providers Call termination service · Hand off voice traffic as an IP stream ITSP terminates it in the PSTN - Call collection service • ITSP collects calls in local areas · Routes to AARNet - Interconnect methods • Back to back TDM gateways (some carriers want this) • SIP trunks, public internet routed or direct connection - Issues · Number range ownership Connecting Security

SBCs

• Session border controllers

- Interface one network to another

- Terminate session on network 1, establish session on network 2

- Deal with network addressing issues and firewalls

- Isolate one network from another

- Provide central point for call detail records

- Provide for legal intercept requirements

Basic SBC may just simulate back to back TDM

- High end SBC may programmatically change protocol events

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What it means to you

- · When complete
 - Support for emerging protocols, technologies
 - Better than competitive prices value for membership share
 - Accurate billing
 - Less technical problems
 - Quicker response to any that do occur
 - New services
- Right now
 - Some changes, rules of engagement

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Rules of engagement (1)

- Dialed numbers (B num)
 - Change from 11 digit numbers 61xxxxxxxxx to 10 digit 0xxxxxxxxx
 - National numbers should always be dialed as 10 digits (ie: area code followed by local number)
 - Dialed numbers with less than 10 digits will not be routed by AARNet unless they match a special service number
 - Dialed numbers starting with 00, 01, 04, 05 will route direct to the PSTN gateway
 - Dialed numbers starting with 02, 03, 07, 08 will route to PSTN unless they match another member
 - · Dialed numbers starting with 06, 09 will not be routed
- Calling numbers (A num)
 - Must always be full 10 digit national number
 - Internal extensions not acceptable

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Rules of engagement (2)

- Gateway access
 - Gatekeepers must always be queried for gateway addresses
 - Gateway addresses may change without notification
 - Gateway addresses will not be publicised
 - (During first stage, RTP may be to/from gateway and thus it is still apparent, but call setup may be prevented to this IP address)

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Rules of engagement (3)

Public access (free-love gatekeeper)

Access to be controlled?

Need to ensure all A numbers are valid?

Perhaps through registration of a number from the members indial range

AARNet Copyright 2007 **Moving forward - Services** First ... service criteria - Will the service add value to the member organisation? - Will the service grow AARNet on net traffic? - Can AARNet charge a commercially competitive tariff for the service? – Can AARNet operate the service without losing money? · Service ideas - Hosted voice, Managed voice - Voice and video mail PSTN outdial backup, indial backup - 3G video gateway What's desired? - Pre paid (calling card) Differentiated voice services - Homogonised presence

Feedback

Real time communications BOF session

- Wednesday, 5.10 to 6pm

- Mossman room

Please come along and discuss:

- What you would like to see from an RTC network

- What services might interest you

- Talk about rules of engagement

- Ask technical questions

- Answer technical questions

