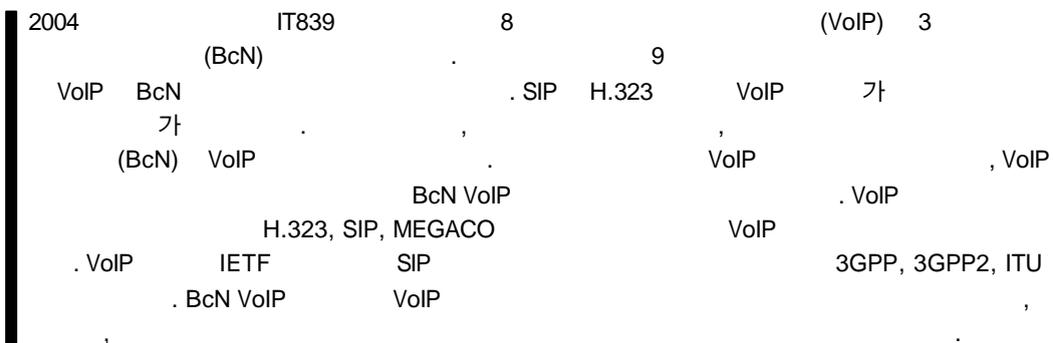




BcN (VoIP)

The Trend of Internet Telephony(VoIP) Technology for BcN

(T.G. Kang) VoIP
 (D.Y. Kim) VoIP
 (Y.S. Kim)



I. 가 가 , BcN
 IT839 8 . VoIP , ,
 (VoIP) 3 (Broad-
 band convergence Network: BcN) 2004 BcN .
 . 9 BcN , ,
 , (Voice over
 Internet Protocol: VoIP) .
 , 9 가 , VoIP
 VoIP BcN SIP/SDP ,
 . BcN
 PSTN VoIP IMS(IP Multimedia Sub-
 SIP/SDP [1]- system)
 [3]. PSTN(Public Switched Telephone Net- , 3 Open API
 work), PCS(Personal Communications .
 Service), 가 2,000 VoIP II
 , 가 VoIP ,
 가 , III VoIP

. VoIP

SIP/SDP
 IETF, ITU, 3GPP, 3GPP3
 . IV BcN VoIP
 . BcN VoIP

(1)
 H.323, SIP
 (Session Initiation Protocol), MGCP(Media Gate-
 way Control Protocol), H.248/MEGACO

[4].

II. VoIP

VoIP VoIP
 , VoIP , VoIP
 . VoIP
 . VoIP

ETSI NGN -SG(Starter Group)
 NGN , - - QoS,
 , NGN
 6가 가 가

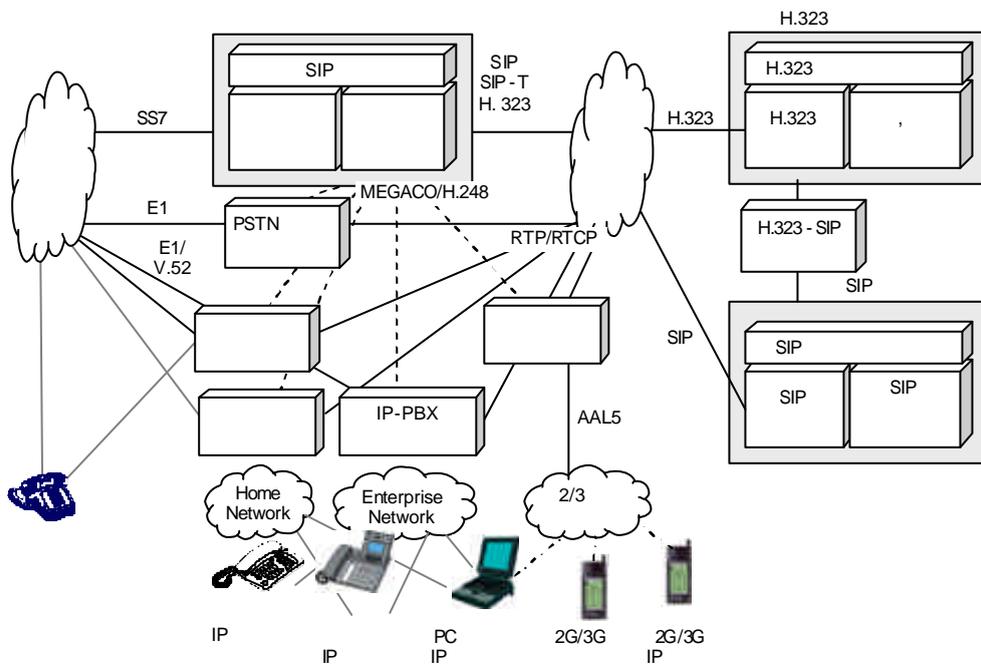
H.323 SIP
 . VoIP

가

1. VoIP

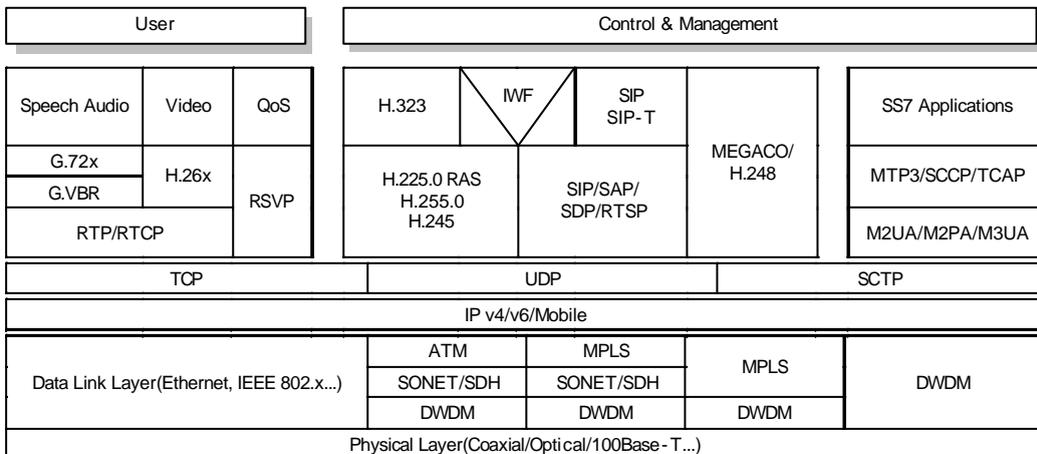
2. VoIP

VoIP PSTN



(1) VoIP

IETF SIP ITU-T/IETF H.225 H.245
 VoIP (protocol identifier) . H.323
 MEGACO/H.248, H.225.0 , H.245 , RTP
 H.323 H.225.0 RAS H.245
 MEGACO/H.248 H.323, SIP, H.225.0 H.245
 IETF SIP SMTP(Simple Mail Transfer Protocol), e-mail, HTTP(Hypertext Transfer Protocol)
 가 SIP 가 request 가 re-
 2) VoIP BcN IP SIP HTTP
 , ATM, MPLS, DWDM IP 가
 MEGACO/H.248 H.323, SIP
 H.323 ITU-T SG(Study Group)
 16 , 1996
 2000 4가
 2002
 5
 . IETF MGCP가
 . H.323 5 , MEGACO/H.248



(2) VoIP

< 1 >

	G.711	G.723.1		G.729A	AMR-NB	SMV	AMR-WB (G.722.2)
Freq. Range(Hz)	300~3,400	300~3,400	300~3,400	300~3,400	300~3,400	300~3,400	50~7,000
Sampling Rate(kHz)	8	8	8	8	8	8	16
Bit Rate (kbps)	64	5.3	6.3	8	4.75~12.2	3.95~8.12	6.60~23.85
Delay(ms)	0.125/0	30/7.5	30/7.5	10/5	20/0(12.2), 20/5	20/10(3)	20/5/1
Excitation	-	ACELP	MP-MLQ	CS-ACELP	CS-ACELP	eX-CELP (RCELP)	ACELP
Fractional Pitch	-	×	×	○	○	○	○
LPC Order	-	10	10	10	10	10	16
LPC Window(ms)	-	22.5	22.5	30	30	30	30

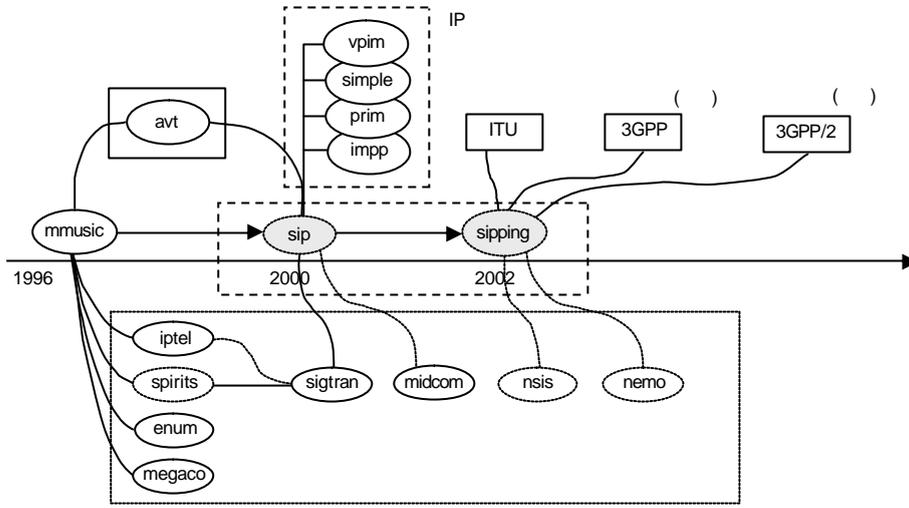
operability

bitrate scalability

2. VoIP SIP

bandwidth scalability

ITU-T Q.9, 2004, 1, Telecom, Q.10, Selection, G.729, G.723.1, interoperability, ITU-T, France, EV, ETRI, [5], 가, bitstream, PSTN, iptel, sprits, emum, megaco, sigtran, VoIP, IETF(Intenet Engineering Task Force), VoIP SIP, IETF, ITU, 3GPP, 3GPP2, (4), 1996, IETF, MMUSIC, SIP, avt, vpim, simple, prim, impp, 2000, SIP, [7], midcom, nsis, nemo, SIP, VoIP, ITU, 3GPP, 3GPP2, SIP, SIP, ITU-T, TTA, ITU-T, sipping 2002, [6].



(4) VoIP

IETF, ITU, 3GPP, 3GPP2

BcN

2. BcN VoIP

BcN

IV. BcN VoIP

VoIP

BcN

[8],[9].

. BcN VoIP

ATM, E1/T1 , DWDM,

OSI

ADSL, POTS ,

가

1. BcN VoIP

VoIP

PSTN,

BcN VoIP

VoIP

3.4kHz

7kHz

(5)

가

. BcN VoIP

Open API,

. VoIP

, 가

BcN

VoIP 20kHz

VoIP

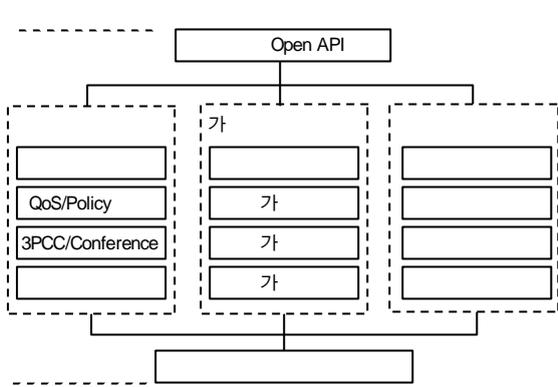
BcN

CCS No.7

Open API BcN

BcN VoIP

가



(5) BcN VoIP

, QoS/Policy,
3PCC/Conference,
, , ,
, QoS/Policy
. 3PCC/Conference

가 , 가 ,
가 , 가 .

, 가 가
. 가 가
, 가 가

가

, ,
, ,
, ,

BcN VoIP

, 가 ,
. VoIP

, BcN VoIP

, 가
, 3GPP 3GPP2
IMS CSCF(Call Session Control Function)
가 ,
가 Open API
가 .

3. BcN VoIP

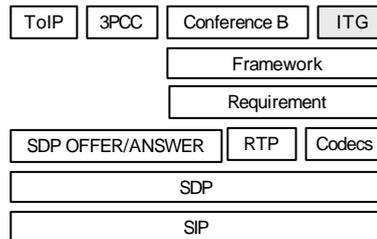
BcN VoIP IETF, 3GPP,
3GPP2 3
(conference bridge model),
ToIP(Text over IP)
3 (The third Party Call
Control: 3PCC), , ToIP
SIP, SDP OFFER/ANSWER,
(6)
3

3
[10].

가

[11]. ToIP

(Intelligent Trans-
coding Gateway: ITG) IMS



(6) BcN VoIP

. BcN VoIP , 9

V.

IT839 8 (VoIP)
 3 (BcN)
 . BcN VoIP 9
 VoIP
 , IETF SIP/SDP
 ITU, 3GPP, 3GPP2 SIP
 . VoIP
 BcN PSTN,
 , 3
 BcN VoIP가 , ,
 3 ,
 . BcN VoIP 3.4kHz
 7kHz , 7kHz 20kHz
 가 . BcN
 VoIP , , ,
 . BcN VoIP
 IMS -
 , / -
 , 3
 , Open API
 BcN VoIP , , ,
 3

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