June 3, 2008

Marlene H. Dortch, Secretary Federal Communications Commission Office of the Secretary 445 12th Street, SW Washington, DC 20554

Re: MM Docket No. 99-325

Dear Ms. Dortch:

On June 2, 2008, the National Association of Broadcasters and the Consumer Electronics Association filed in the above-referenced docket a letter submitting to the FCC the recently revised NRSC In-Band/On-Channel (IBOC) Digital Radio Broadcasting Standard, NRSC-5-B. The electronic version of NRSC-5-B was inadvertently omitted from that filing. The complete filing is herewith attached.

Respectfully submitted,

Valerie Schulte

Deputy General Counsel

National Association of Broadcasters

acure Schulte

Attachment

June 2, 2008

Marlene H. Dortch, Secretary Federal Communications Commission Office of the Secretary 445 12th Street, SW Washington, DC 20554

Dear Ms. Dortch:

In the *First Report and Order* in MM Docket No. 99-325, the Commission sought the assistance of the National Radio Systems Committee (NRSC)¹ in the development of an In-Band/On-Channel (IBOC) digital radio standard.² On May 18, 2005, the Consumer Electronics Association (CEA) and the National Association of Broadcasters (NAB), as the co-sponsors of the NRSC, submitted the NRSC In-Band/On-Channel (IBOC) Digital Radio Broadcasting Standard, NRSC-5, to the Commission for consideration.

NRSC-5 was approved by the NRSC's Digital Audio Broadcasting (DAB) Subcommittee on April 16, 2005. It is based on iBiquity Digital Corporation's IBOC digital radio technology. Subsequently, the standard was revised, as NRSC-5-A by the NRSC's Digital Radio Broadcasting (DRB) Subcommittee, adding sections concerning Advanced Application Services and a new reference document.

The latest revision of the IBOC standard, NRSC-5-B, was adopted by the NRSC DRB Subcommittee on April 12, 2008. NRSC-5-B updates the reference documents and makes minor changes to the FM mask³ and to the description of hybrid IBOC operation.⁴ NRSC-5-B also incorporates the revisions of NRSC-5-A.

¹ NRSC serves as an industry-wide standards-setting body for technical aspects of terrestrial over-the-air radio broadcasting systems in the United States.

² First Report and Order, MM Docket No. 99-325, released October 11, 2002, at para. 44.

³ The new FM mask was previously submitted to the Commission by iBiquity Digital in a letter to the FCC, dated July 5, 2006.

⁴ For convenience, we are attaching an unofficial summary prepared by NAB engineers of the recent revisions to the IBOC standard.

NRSC-5-B is hereby submitted to the Commission, for its further consideration and approval as the basis for the FCC technical rules on IBOC digital radio.

Respectfully submitted,

CONSUMER ELECTRONICS ASSOCIATION

NATIONAL ASSOCIATION OF BROADCASTERS

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AB Radio TechCheck



The Weekly NAB Newsletter for Radio Broadcast Engineers

June 2, 2008

Updated NRSC-5 Standard Published

The National Radio Systems Committee (NRSC) has now published the *NRSC-5-B In-band/on-channel Digital Radio Broadcasting Standard* which was adopted by the Digital Radio Broadcast (DRB) Subcommittee at the April 12, 2008 meeting of the group (held in Las Vegas in conjunction with the 2008 NAB Show). This document and the associated reference documents are available for download free of charge at

www.nrscstandards.org/SG/NRSC-5-B.asp.

First adopted in April 2005 then revised in September 2005 (as NRSC-5-A), this Standard sets forth the requirements for broadcasting digital audio and ancillary data signals in the AM and FM radio bands. It is structured as a main document (the Standard itself, cover shown in image at right) which provides an overview of the AM and FM IBOC systems designed by iBiquity Digital Corporation (and upon which the Standard is based), and a set of reference documents authored by iBiquity (with input from the NRSC) which provide the detailed information needed for those "skilled in the art" to construct compatible equipment. NRSC-5-A differed from the original document in that it specified a mechanism for transmission of advanced data services.

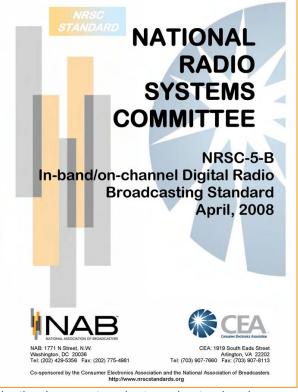
This latest version includes substantive (but relatively minor) changes as well as editorial revisions designed to make the document easier to understand and more complete. None of these changes are expected to cause any backwards compatibility issues with HD Radio receivers. Some of the more significant changes are highlighted here (section numbers refer to sections in NRSC-5-B):

Table of Contents – tables listing all of the Figures and Tables in the Standard are now included; also, Table 5 (description of information transmitted by SIS) has been augmented to include all types of SIS data;

Section 3.3.2 – Data Inputs – this section was edited to make clearer the different types of data supported by the system. While the previous version discussed only two types of data—program service data (PSD) and non-program service data—the updated version makes clear the fact that the system supports three distinct data services:

- *PSD*, which includes descriptive information associated with the transmitted audio programming such as song title and artist;
- Station Information Service (SIS) data, which contains information about the station and the signal that is not associated with an individual program stream; and,
- Advanced Data Services (ADS) data, which is data that is sent using the ADS portion of the system (examples would include navigation/traffic information, image files of CD covers, etc.).

Section 4.1.8 – Spectrum Emission Limits for AM IBOC – this is a new section which includes and augments information from Section 4.1.7 of the previous version (the AM IBOC mask itself is unchanged). A new sub-section, 4.1.8.1 – Measurement of mask compliance for AM IBOC systems, references an NRSC Guideline which is currently under development by the NRSC's IBOC Standards Development Working Group



(ISDWG). This Guideline will include detailed mask compliance measurement procedures applicable to different types of measurements (*e.g.*, factory test, in-service out-of-band emissions), as well as recommended locations for making measurements depending upon the specifics of a particular implementation. The ISDWG expects to complete this new Guideline later this year.

Section 4.2.8 – Spectrum Emission Limits for FM IBOC – the FM hybrid IBOC spectrum emission mask has been modified from the previous versions and is included in this section. This is a minor modification (a relaxation of the mask in the region from 200-250 kHz offset from the channel center frequency) which was previously submitted to the FCC by iBiquity Digital Corporation (developers of the HD Radio AM and FM IBOC systems) in July, 2006. The modified mask is shown in *Figure 19 - NRSC-5 FM hybrid waveform noise and emission limits*. Additional changes here are similar to those discussed for Section 4.1.8 above.

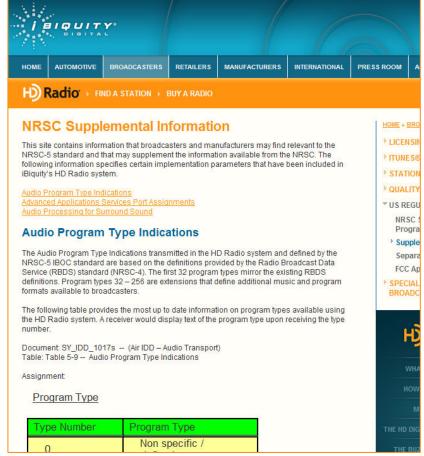
Section 5.3 – Advanced Data Services

– new text in this section clarifies the definition and function of the Advanced Applications Services Transport (AAT). AAT is the method chosen by the NRSC for transport of fixed and opportunistic advanced data services (ADS) data in the IBOC system (AAT was first included in the NRSC-5-A version of the Standard).

Reference document 1011s rev. F — the most important change here is the substitution of extended hybrid service mode MP4 with a new mode, MP11. The new service mode has the same approximate information rate as did the old (148 kbps) but the bits are allocated differently among the various logical channels, and a potential backwards compatibility issue (identified by iBiquity since adoption of NRSC-5-A) is resolved. In addition, this revision of 1011s removes an interleaver definition which was found to cause delay in audio and data, leaving a single interleaver

Coincident with the adoption of this most recent version of the Standard, iBiquity has launched an "NRSC Supplemental

definition instead of two.



Information" Web page (see image at right) which will allow iBiquity to provide HD Radio system information to broadcasters and manufacturers that has not yet been incorporated in the Standard. Included here at present is information about:

- Audio Program Type lists the codes for program formats (e.g., sports, talk, etc.) which have been added since adoption of the Standard (type numbers 32-54 shown here are new);
- AAT port assignments provides new information about the "port identifiers" required for each advanced data service;
- Surround sound codes iBiquity has assigned a unique code to various surround sound formats which
 can be used with the HD Radio system.

This Web page can be accessed directly at

<u>www.ibiquity.com/broadcasters/us_regulatory/nrsc_supplemental_information</u> or through links provided on the NRSC-5-B Web page.

NRSC STANDARD

NATIONAL RADIO SYSTEMS COMMITTEE

In-band/on-channel Digital Radio
Broadcasting Standard
April, 2008



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This NRSC Standard is considered to have International Standardization implication, but the International Electrotechnical Commission activity has not progressed to the point where a valid comparison between the NRSC Standard and the IEC document can be made.

This Standard does not purport to address all safety problems associated with its use or all applicable regulatory requirements. It is the responsibility of the user of this Standard to establish appropriate safety and health practices and to determine the applicability of regulatory limitations before its use.

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Arlington, VA 22202

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FOREWORD

This standard was originally developed by the National Radio Systems Committee's (NRSC's) Digital Audio Broadcasting (DAB) Subcommittee and subsequently revised by the NRSC's Digital Radio Broadcasting (DRB) Subcommittee. At the time of first adoption, the NRSC was chaired by Charles Morgan of Susquehanna Broadcasting, the DAB Subcommittee was co-chaired by Michael Bergman of Kenwood and Milford Smith of Greater Media, and the IBOC Standards Development Working Group (ISDWG) was co-chaired by Paul Feinberg of Sony Electronics Inc. and Dr. H. Donald Messer of the International Broadcasting Bureau. For Revision B, the NRSC was chaired by Milford Smith of Greater Media, the DRB Subcommittee was co-chaired by Michael Bergman of Kenwood and Andy Laird of Journal Broadcasting, and the IBOC Standards Development Working Group (ISDWG) was chaired by Dom Bordonaro of Cox Broadcasting.

The NRSC is jointly sponsored by the Consumer Electronics Association and the National Association of Broadcasters. It serves as an industry-wide standards-setting body for technical aspects of terrestrial over-the-air radio broadcasting systems in the United States.

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IN-BAND/ON-CHANNEL DIGITAL RADIO BROADCASTING STANDARD

1 SCOPE

This standard sets forth the requirements for a system for broadcasting digital audio and ancillary digital data signals over AM broadcast channels spaced 10 kHz apart that may contain analog amplitude modulated signals, and over FM broadcast channels spaced 200 kHz apart that may contain analog frequency modulated signals.

2 REFERENCES

2.1 Normative References

The following normative references are incorporated by reference herein. At the time of publication the edition indicated was valid. All standards are subject to revision, and parties to agreements based on this standard are encouraged to investigate the possibility of applying the most recent editions of this standard and/or the references listed below. In case of discrepancy normative references shall prevail.

For the purposes of compliance with this standard, the use of the term "HD Radio™" in the normative references shall be interpreted as the generic term "IBOC" for the NRSC-5 compliant system and shall not be construed as a requirement to adhere to undisclosed private specifications that are required to license the HD Radio name from its owner.

Note that information relating to the iBiquity Advanced Application Service (AAS) in the normative reference documents is considered non-normative. Only the Advanced Application Services Transport (AAT) is incorporated as normative in this Standard, and is described in normative reference [10].

- [1] Doc. No. SY_IDD_1011s rev. F, HD Radio Air Interface Design Description Layer 1 FM, iBiquity Digital Corporation, 8/7/07
- [2] Doc. No. SY_IDD_1012s rev. E, HD Radio Air Interface Design Description Layer 1 AM, iBiquity Digital Corporation, 3/22/05
- [3] Doc. No. SY_IDD_1014s rev. H, HD Radio Air Interface Design Description Layer 2 Channel Multiplex Protocol, iBiquity Digital Corporation, 12/5/07
- [4] Doc. No. SY_IDD_1017s rev. F, HD Radio Air Interface Design Description Audio Transport, iBiquity Digital Corporation, 12/6/07
- [5] Doc. No. SY_IDD_1020s rev. G, HD Radio Air Interface Design Description Station Information Service Protocol, iBiquity Digital Corporation, 2/28/08
- [6] Doc. No. SY_SSS_1026s rev. E, HD Radio FM Transmission System Specifications, iBiquity Digital Corporation, 1/30/08
- [7] Doc. No. SY_IDD_1028s rev. D, HD Radio Air Interface Design Description Main Program Service Data, iBiquity Digital Corporation, 11/7/07
- [8] Doc. No. SY_SSS_1082s rev. E, HD Radio AM Transmission System Specifications, iBiquity Digital Corporation, 1/30/08
- [9] Doc. No. SY_IDD_1085s rev. C, HD Radio Air Interface Design Description Program Service Data Transport, iBiquity Digital Corporation, 2/7/05

[10] Doc. No. SY_IDD_1019s rev. F, HD Radio Air Interface Design Description – Advanced Application Services Transport, iBiguity Digital Corporation, 12/6/07

2.2 Normative Reference Acquisition

iBiquity Digital Corporation reference documents:

National Radio Systems Committee (co-sponsored by the Consumer Electronics Association and the National Association of Broadcasters)

CEA: 1919 South Eads Street, Arlington VA 22202; Tel: 703-907-7421; Fax: 703-907-4190 NAB: 1771 N Street NW, Washington DC 20036; Tel: 202-429-5346; Fax: 202-775-4981 http://www.nrscstandards.org

2.3 Informative References

The following references contain information that may be useful to those implementing this standard. At the time of publication the edition indicated was valid. All standards are subject to revision, and parties to agreements based on this standard are encouraged to investigate the possibility of applying the most recent editions of the standards listed below.

- [1] NRSC-5-B RF Mask Compliance: Measurement Methods and Practice, NRSC-G-201, <date TBD>
- [2] The IBOC Handbook: Understanding HD Radio Technology, Focal Press, 2007
- [3] The NAB Engineering Handbook, 10th Edition, Focal Press, 2007

2.4 Informative Reference Acquisition

(Not applicable)

2.5 Definitions

In this standard the practice of the Institute of Electrical and Electronics Engineers (IEEE) as outlined in the Institute's published standards is used for definitions of terms. Definitions of terms used in this standard that are not covered by IEEE practice, or for which industry practice differs from IEEE practice, are as follows:

Advanced Application Services (AAS)	Advanced Application Services is the iBiquity implementation of
	the generic ADS described in this document and is not part of
	this Standard. AAS is transported on IBOC by AAT.

Advanced Application Services Transport (AAT)

Advanced Applications Services Transport is the transport mechanism incorporated in this Standard that enables the transmission of advanced data services, such as those supplied by the iBiquity AAS.

Advanced data services (ADS)	Advanced data services are any data services consisting of
	either text, audio, video, or other data carried on the IBOC
	transport other than SIS, MPSD, or SPSD.

All digital waveform	A transmitted waveform for modes composed entirely of digitally
	modulated subcarriers without an analog signal.

Channel encoding	The process used to add error protection to each of the logical
	channels to improve the reliability of the transmitted information.

Code rate Defines the increase in overhead on a coded channel resulting

from channel encoding. It is the ratio of information bits to the

total number of bits after coding.

Convolutional encoding A form of forward-error-correction channel encoding that inserts

coding bits into a continuous stream of information bits to form a predictable structure. Unlike a block encoder, a convolutional encoder has memory; its output is a function of current and

previous inputs.

Diversity delay Imposition of a fixed time delay in one of two channels carrying

the same information to defeat non-stationary channel impairments such as fading and impulsive noise.

Extended hybrid waveform A transmitted waveform for modes composed of the analog FM

signal plus digitally modulated primary main subcarriers and some or all primary extended subcarriers. This waveform will normally be used during an initial transitional phase preceding

conversion to the all digital waveform.

Frequency partition (For FM IBOC) a group of OFDM subcarriers containing data

subcarriers and a reference subcarrier.

Hybrid waveform A transmitted waveform for modes composed of the analog -

modulated signal, plus digitally modulated primary main subcarriers. This waveform will normally be used during an initial transitional phase preceding conversion to the all digital

waveform.

Interleaver partition A logical subdivision of the overall interleaver matrix.

Interleaving A reordering of the message bits to distribute them in time (over

different OFDM symbols) and frequency (over different OFDM

subcarriers) to mitigate the effects of signal fading and

interference.

Interleaving process A series of manipulations performed on one or more coded

transfer frames (vectors) to reorder their bits into one or more

interleaver matrices whose contents are destined for a

particular portion of the transmitted spectrum.

Logical channel A signal path that conducts transfer frames from Layer 2

through Layer 1 with a specified grade of service.

Main Program Service (MPS)

The audio programming and program service data that a radio

station broadcasts over its main channel for reception by the

general public.

Main Program Service Data (MPSD) One of two general classes of information sent through the MPS

(the other being Main Program Service Audio). Main Program Service Data is Program Service Data (defined below) that is

associated with the Main Program Service.

OFDM subcarrier A narrowband PSK or QAM-modulated carrier within the

allocated channel, which, taken together with all OFDM

subcarriers, constitute the frequency domain representation of one OFDM symbol.

OFDM symbol Time domain pulse of duration T_s, representing all the active

subcarriers and containing all the data in one row from the interleaver and system control data sequence matrices.

Program Service Data (PSD)Data that is transmitted along with the program audio and that is

intended to describe or complement the audio program heard

by the listener (e.g., song title, artist, etc.)

Protocol Data Unit (PDU) A Protocol Data Unit (PDU) is the structured data block in the

IBOC system that is produced by a specific layer (or process within a layer) of the transmitter protocol stack. The PDUs of a given layer may encapsulate PDUs from the next higher layer of the stack and/or include content data and protocol-control information originating in the layer (or process) itself. The PDUs generated by each layer (or process) in the transmitter protocol stack are inputs to a corresponding layer (or process)

in the receiver protocol stack.

Reference subcarrier Dedicated OFDM subcarrier modulated with the SCCH data.

There are up to 61 reference subcarriers depending on the

mode (for FM) and up to 4 for AM.

Service mode A specific configuration of operating parameters specifying

throughput, performance level, and selected logical channels.

Station Information Service (SIS) The Station Information Services provides the necessary radio

station control and identification information, such as station call sign identification, time and location reference information.

Supplemental Program Service (SPS) The Supplemental Program Service provides for the option of

multiplexing additional programs with the MPS. The SPS includes Supplemental Program Service Audio (SPSA) and may also include Supplemental Program Service Data (SPSD).

Supplemental Program Service Data (SPSD)

One of two general classes of information sent through the SPS

(the other being Supplemental Program Service Audio). Supplemental Program Service Data is Program Service Data (defined above) that is associated with the Supplemental

Program Service.

System control channel (SCCH) A channel consisting of control information from the

configuration administrator and status information from Layer 1.

System control data sequence A sequence of bits destined for each reference subcarrier

representing the various system control components relayed

between Layer 1 and Layer 2.

System protocol stack The ordered protocols associated with data processing in the

transmitter and receiver.

Transfer frameAn ordered, one-dimensional collection of data bits of specified

length originating in Layer 2, grouped for processing through a

logical channel.

Transfer frame multiplexerA device that combines two or more transfer frames into a

single vector.

2.6 Symbols and Abbreviations

Symbols and abbreviations used in this standard are as follows:

AAS Advanced Application Services

AAT Advanced Application Services Transport

ADS Advanced Data Services
AM Amplitude Modulation

API Application Programming Interface

BBM Block Boundary Marker
BPSK Binary Phase Shift Keying

FCC Federal Communications Commission

FEC Forward Error Correction
FM Frequency Modulation
GPS Global Positioning System

IBOC In-Band/On-Channel

ID Identification

IP Interleaving Processes

L1 Layer 1

MF Medium Frequency

MPA Main Program Service Audio

MPS Main Program ServiceMPSD Main Program Service Data

N/A Not Applicable

OFDM Orthogonal Frequency Division Multiplexing

P3IS P3 Interleaver Select (this function was rendered obsolete in rev. B)

PAD Program Associated Data
PCM Pulse Code Modulation
PDU Protocol Data Unit

PIDS Primary IBOC Data Service Logical Channel

PM Primary Main

PPP Point to Point Protocol
PSD Program Service Data
PX Primary Extended

QAM Quadrature Amplitude Modulation **QPSK** Quadrature Phase Shift Keying

RF Radio Frequency
SB Secondary Broadband
SCCH System Control Channel

SIDS Secondary IBOC Data Service Logical Channel

SIS Station Information Service

SM Secondary Main

SP Secondary Protected

SPA Supplemental Program Service Audio

SPS Supplemental Program Service

SPSD Supplemental Program Service Data

SX Secondary Extended
URL Uniform Resource Locator
VHF Very High Frequency

2.7 Compliance Notation

As used in this document, "shall" or "will" denotes a mandatory provision of the standard. "Should" denotes a provision that is recommended but not mandatory. "May" denotes a feature whose absence does not preclude compliance, and that may or may not be present at the option of the implementer.

3 SYSTEM OVERVIEW

The In-Band/On-Channel (IBOC) digital radio broadcasting system specified in this standard is designed to permit a smooth evolution from current analog radio broadcasting to fully digital radio broadcasting. This system delivers digital audio and data services to mobile, portable, and fixed receivers from terrestrial transmitters on existing Amplitude Modulation (AM) and Frequency Modulation (FM) radio broadcast channels. Broadcasters may continue to transmit AM and FM analog signals simultaneously with the IBOC digital signals, allowing themselves and their listeners to convert from analog to digital radio while maintaining their current frequency allocations.

The system accepts as input compressed digital audio and utilizes baseband signal processing techniques such as interleaving and forward error correction to increase the robustness of the signal in the transmission channel. This allows a high quality audio signal plus ancillary data to be transmitted using power levels and band segments selected to minimize interference to existing analog signals.

Figure 1 illustrates the three major subsystems of the IBOC digital radio system specified by NRSC-5 and how they relate to one another. The major subsystems are:

- RF/transmission subsystem
- Transport and service multiplex subsystem
- Audio and data input subsystems

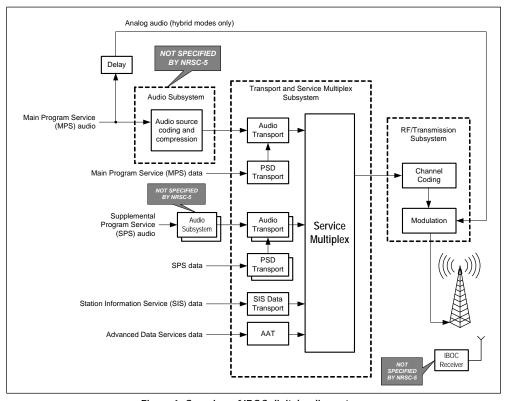


Figure 1: Overview of IBOC digital radio system

¹ This system, operating in modes MP1 (FM band) and MA1 (AM band) underwent an extensive evaluation by the NRSC. The NRSC has not tested or evaluated any modes other than these. These operating modes are described in normative references [1] and [2] for the FM and AM bands, respectively. The NRSC IBOC evaluation reports, entitled "Evaluation of the iBiquity Digital Corporation IBOC System, Part 1 – FM IBOC," National Radio Systems Committee DAB Subcommittee, November 29, 2001, and "Evaluation of the iBiquity Digital Corporation IBOC System, Part 2 – AM IBOC," National Radio Systems Committee DAB Subcommittee, April 6, 2002, are available on the NRSC web site at www.nrscstandards.org.

3.1 RF/transmission Subsystem

The RF/transmission subsystem shall comply with the requirements in normative references [1] and [6] for FM and [2] and [8] for AM. This subsystem takes the multiplexed bit stream and applies coding and interleaving that can be used by the receiver to reconstruct the transmitted data, even when the received signal does not exactly match the transmitted signal due to impairments in the channel. The multiplexed and coded bit stream is modulated onto orthogonal frequency division multiplexed (OFDM) subcarriers and up-converted to the AM or FM band. A description of the RF/transmission subsystem is given in Section 4; a detailed specification is given in normative references [1] and [6] (FM band) and [2] and [8] (AM band).

3.2 Transport and Service Multiplex Subsystem

The transport and service multiplex subsystem shall comply with the requirements in normative references [3], [4], [5], and [9]. This subsystem feeds the information to be transmitted to the RF/transmission subsystem. It takes the audio and data information that it receives, organizes it into packets, and multiplexes the packets into a single data stream. Each packet is uniquely identified as an audio or data packet. Certain data packets (*i.e.*, those containing program service data, which includes song title, artist, etc.) are added to the stream of packets carrying their associated audio information before they are fed into the multiplexer.

The transport stream is modeled loosely on the ISO 7498-1 standard. A description of the transport and service multiplex subsystem is given in Section 5; a detailed specification is given in normative references [3], [4], [5], and [9].

3.3 Audio and Data Input Subsystems

3.3.1 Audio Inputs

Source coding and compression of the main program service (MPS) and supplemental program service (SPS) audio must be performed before the audio information is fed into the audio transport subsystems. Each audio service (main program service and each individual supplemental program service) has its own source coding, compression and transport subsystem. NRSC-5 does not include specifications for audio source coding and compression. Suitable audio source coding and compression systems will use appropriate technologies (e.g., perceptual audio coding) to reduce the bit rate required for description of audio signals.

In hybrid modes the analog MPS audio is also modulated directly onto the RF carrier for reception by conventional analog receivers. The MPS analog audio does not pass through the audio transport subsystem, and is delayed so that it will arrive at the receiver close enough in time to the digital signal.² This will enable seamless switching from digital to analog reception when the received signal quality is not sufficient for digital audio reception or when digital packets in the MPS PDU are corrupted. This "blend" capability is also used for fast channel changes, allowing the receiver to demodulate and play out the analog stream first and then blend to the digital audio stream.

MPS audio is discussed further in Section 5, and some of the necessary characteristics for audio source coding are described in Section 6.

3.3.2 Data Inputs

There are three types of data inputs to the IBOC digital radio system. The first type is Program Service Data (PSD), which includes descriptive information associated with the transmitted audio programming such as song title and artist. The second type is Station Information Service (SIS) data, which contains

² This is referred to as "diversity delay," T_{dd}, and is specified in references [1] and [2].

information about the station and the signal that is not associated with an individual program stream. The third type of data is Advanced Data Services (ADS) data, which is referred to generally as other data.

3.3.2.1 Program Service Data Inputs

PSD inputs shall comply with the detailed requirements in normative reference [7]. The PSD fields include song title, artist, album, genre, comment, commercial and reference identifiers. A description of PSD is given in Section 5.

There are two classifications of PSD. The first is Main PSD (MPSD), which may be transmitted along with the main program audio and is intended to describe or complement the main program audio program heard by the radio listener.

The second classification of PSD is Supplemental PSD (SPSD), which may be transmitted with each Supplemental Program Service audio program. Each SPSD input is formatted in the same fashion as MPSD, but has content for and is associated with a specific Supplemental Program Service audio program.

3.3.2.2 Station Information Service Inputs

The second major type of data input to the IBOC digital radio system is Station Information Service (SIS) data. Station Information Service Data inputs shall comply with the requirements in normative reference [5]. SIS data provides more general information about the station, including technical information that is useful for non-program related applications. The SIS fields include station identification number (based in part on the FCC facility identification number), station call letters, station name, station location, two fields dedicated to station time, a field that permits the broadcaster to send an arbitrary text message, and two reserved fields. A description of SIS data is given in Section 5; a detailed specification is given in normative reference [5].

3.3.2.3 Other Data Inputs

Advanced Data Services (ADS) provide broadcasters with the ability to transmit information that may be unrelated to MPS, SPS or SIS. These services can carry any form and content that can be expressed as a data file or a data stream, including audio services. Examples of such services include (i) visual effects associated with MPS, SIS, or SPS services; (ii) multimedia presentations of stock, news, weather, and entertainment programming including audio, text and images; (iii) broadcast updates to in-vehicle systems; (iv) local storage of content for time shifting and later replay; (v) targeted advertising; and (vi) traffic updates and information for use with navigation systems. These services are incorporated onto the IBOC signal through the AAT protocol, and are discussed further in Section 5.

4 RF/TRANSMISSION SYSTEM CHARACTERISTICS

4.1 AM RF/transmission System Characteristics

This section specifies the RF portion of the NRSC-5 IBOC digital radio standard for AM band implementations. Figure 2 illustrates how the RF/transmission portion of the standard relates to the other parts. It is an overview of the entire AM band implementation of the NRSC-5 IBOC digital radio standard.

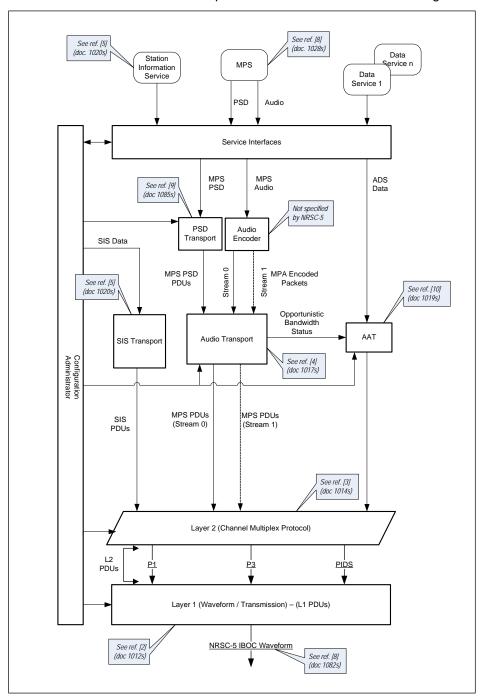


Figure 2: AM band implementation of NRSC-5 IBOC digital radio standard

4.1.1 Transmission Characteristics

This section includes a high-level description of each Layer 1 functional block and the associated signal flow. Figure 3 is a top level block diagram of the RF/transmission subsystem illustrating the order of processing therein. Figure 4 is a functional block diagram of Layer 1 processing. Audio and data are passed from the higher protocol layers to the physical layer, the modem, through the Layer 2 – Layer 1 interface.

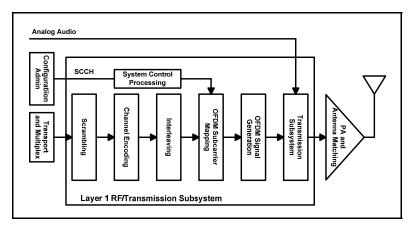


Figure 3: RF/transmission subsystem block diagram

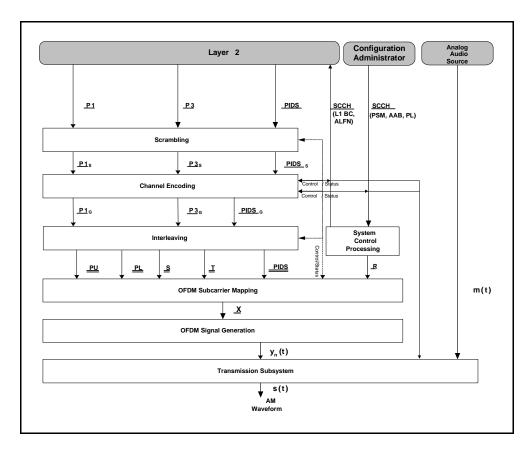


Figure 4: AM air interface Layer 1 functional block diagram with details of data flow illustrated

4.1.2 Layer 1 Interface

The layer 1 interface illustrates the access points between the channel multiplex and layer 1 of the system protocol stack. Data enters layer 1 as discrete transfer frames, with unique size and rate determined by the service mode, as specified in normative references [1] and [2]. Transfer frames that carry information from the channel multiplex are referred to as L1 PDUs.

4.1.3 Logical Channels

The concept of logical channels and their function is central to the transport and transmission of data through the IBOC system. A logical channel is a signal path that conducts Layer 1 PDUs through Layer 1 with a specified grade of service. Logical channels are specified in normative reference [2]. In Figure 4 the logical channels are denoted by symbols such as P1, PIDS, etc. The underscore indicates that the data in the logical channel is formatted as a vector.

4.1.4 Channel Coding

Channel coding comprises the functions of scrambling, channel encoding, and interleaving shown in Figure 3 and specified in normative references [1] and [2].

4.1.4.1 Scrambling

This function (specified in normative references [1] and [2]) randomizes the digital data in each logical channel to "whiten" and mitigate signal periodicities when the waveform is demodulated in a conventional analog AM demodulator. The bits in each logical channel are scrambled to randomize the time-domain data and aid in receiver synchronization. The inputs to the scramblers are the active logical channels from Layer 2, as selected by the service mode. The outputs of the scramblers are transfer frames of scrambled bits for each of the active logical channels. The scrambler generates a pseudorandom code which is modulo 2 summed with the input data vectors. The code generator is a linear feedback shift register.

4.1.4.2 Channel Encoding

Channel encoding improves system performance by increasing the robustness of the signal in the presence of channel impairments. The channel encoding process is characterized by punctured convolutional encoding.

Punctured convolutional encoding is applied to each logical channel in the RF/transmission subsystem for forward error correction. Several different encoding polynomials and puncture matrices are used. Different logical channels have different code rates. The specification of the forward error correction coding used for each logical channel and each service mode is contained in normative reference [2].

4.1.4.3 Interleaving

Interleaving is also applied to the logical channels in the RF/transmission subsystem. The interleaving process provides both time and frequency diversity. The manner in which diversity delay (time) is applied to these logical channels is specified in normative reference [2] for each service mode. The delay provides time diversity to the affected logical channels. If applied, the value of the diversity delay is a fixed value.

Interleaving is comprised of four primary operations: subframe generation, delay for diversity, transmit time alignment, and bit mapping. These operations are applied to logical channels P1, P3, and PIDS starting with subframe generation (see Figure 5). Subframe generation creates new logical channels in which the incoming information has been redistributed. Some subframes are passed to delay buffers, creating the diversity delay path which results in main and backup streams. The final step is for the bit mapper to disperse the sequential subframe data to specific non-sequential locations in the interleaver output matrices. This bit mapping results in a new set of logical channels that pass this now-interleaved

information to the OFDM subcarrier generation process. The interleaving processes for each service mode, and parameters for each block, are specified in Section 10 of normative reference [2].

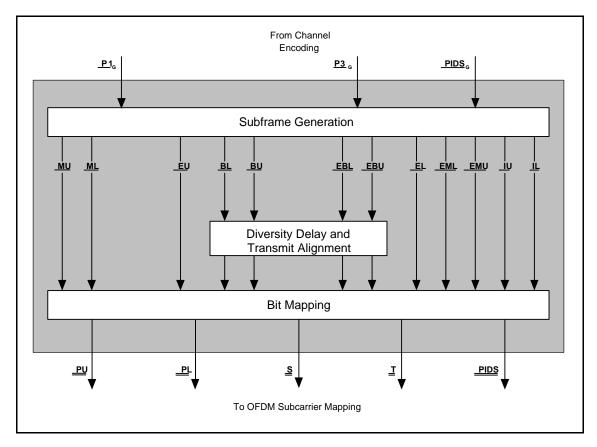


Figure 5: Interleaving conceptual block diagram

4.1.5 System Control Processing

As shown in Figure 3, the system control channel (SCCH) bypasses the channel coding. Under the direction of the system configuration settings, system control processing assembles and differentially encodes a sequence of bits (system control data sequence) destined for each reference subcarrier, as shown in Figure 6. There are 2 reference subcarriers at specific carrier offsets in the OFDM spectrum. This processing is specified in Section 11 of normative reference [2].

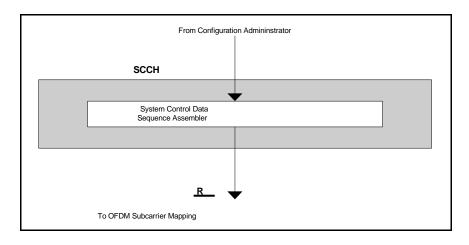


Figure 6: System control processing conceptual diagram

4.1.6 Subcarrier Mapping and Modulation

OFDM subcarrier mapping assigns interleaver partitions to frequency partitions. For each active interleaver matrix, OFDM subcarrier mapping assigns a row of bits from each interleaver to its respective frequency carrier and constellation value in the complex output vector \underline{X} . In addition, system control data sequence bits from a row of \underline{R} are mapped to the active reference subcarrier locations in \underline{X} . The service mode dictates which interleaver matrices and which elements of \underline{R} are active. Figure 7 shows the inputs, output, and component functions of OFDM subcarrier mapping.

The inputs to OFDM subcarrier mapping for each symbol are a row of bits from each active interleaver matrix and a row of bits from \underline{R} , the matrix of system control data sequences. The output from OFDM subcarrier mapping for each OFDM symbol is a single complex vector, X, of length 163.

The interleaver matrices carrying the user audio and data (<u>PU</u>, <u>PL</u>, <u>S</u>, <u>T</u>, <u>PIDS</u>) are mapped to scaled quadrature phase shift keying (QPSK), 16-QAM, or 64-QAM constellation points and to specific subcarriers. The R matrix is mapped to binary phase shift keying (BPSK) constellation points and the reference subcarriers. These phasors are then scaled in amplitude and then mapped to their assigned OFDM subcarriers. This process results in a vector, <u>X</u>, of phasors which are output to the OFDM signal generation function. This processing is specified in Section 12 of normative reference [2].

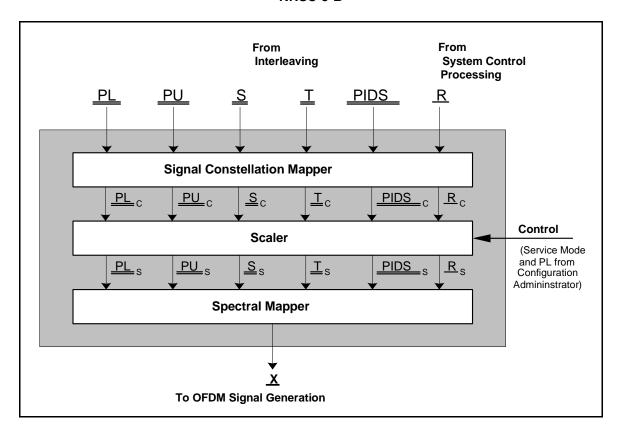


Figure 7: OFDM subcarrier mapping conceptual block diagram

4.1.7 Transmission

OFDM signal generation receives complex, frequency-domain OFDM symbols from OFDM subcarrier mapping, and outputs time-domain pulses representing the digital portion of the AM IBOC signal.

The input to OFDM signal generation for the n^{th} symbol is a complex vector \underline{X}_n of length L, representing the complex constellation values for each OFDM subcarrier in OFDM symbol n. For notational convenience, the output of OFDM subcarrier mapping described above did not use the subscript n. Rather, it referred to the vector \underline{X} as representing a single OFDM symbol. In this section, the subscript is appended to \underline{X} because of the significance of n to OFDM signal generation. The OFDM symbol is transformed to the time domain by a discrete Fourier transform and shaped to create one time domain symbol, $y_n(t)$. The output of OFDM signal generation is a complex, baseband, time-domain pulse $y_n(t)$, representing the digital portion of the AM IBOC signal for OFDM symbol n.

The $y_n(t)$ symbols are concatenated to form a continuous time domain waveform. This OFDM waveform is combined (summed) with the amplitude modulated (AM) waveform $a_n(t)$ (in the hybrid mode) to create $z_n(t)$. This waveform is upconverted to create the complete IBOC RF waveform for transmission. This is illustrated in Figure 8. The waveform is then spectrally mapped and frequency partitioned across the set of OFDM subcarriers. Normative reference [8] details the key transmission specifications for the AM IBOC RF waveform including carrier frequency and channel spacing, synchronization tolerances, analog host performance, spectral emission limits, digital sideband levels, analog modulation level, phase noise, error vector magnitude, gain flatness, amplitude and phase symmetry, and group delay flatness.

There are several issues of time alignment that the transmission system must address. For transmit facilities so equipped, every L1 frame transmitted must be properly aligned with GPS time. Also, the various logical channels must be properly aligned with each other and in some service modes some channels are purposely delayed by a fixed amount to accommodate diversity combining at the receiver.

Layer 1 provides for the time alignment of the transfer frames received from the channel multiplex. The higher protocol layers provide alignment of the contents of the transfer frames.

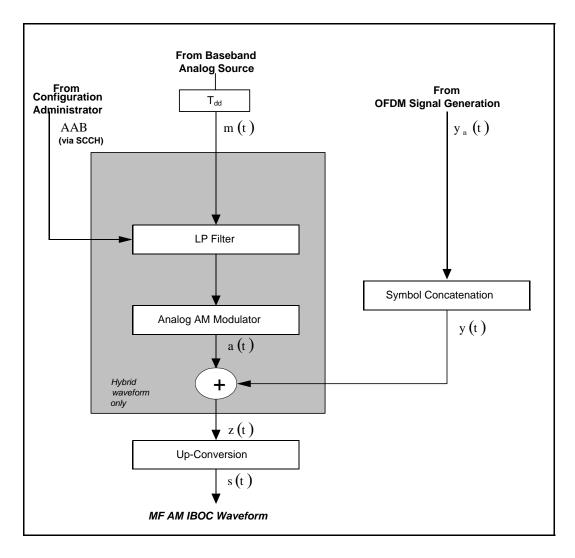


Figure 8: Hybrid/extended hybrid transmission subsystem functional block diagram

4.1.8 Spectrum Emissions Limits for AM IBOC

For hybrid transmissions utilizing 5 kHz mode analog bandwidth, noise and spuriously generated signals from all sources, including phase noise and intermodulation products, shall conform to the limits as described in Section 4.5.1 of normative reference [8] and shown in Figure 9.³

³ The requirements for noise and spurious emission limits illustrated in Figures 9-11 reflect acceptable performance criteria. In certain circumstances, additional measures (filtering, active emissions suppression, etc.) may be needed to reduce the spectral emissions below the limits given in this subsection in order to reduce mutual interference between broadcast stations.

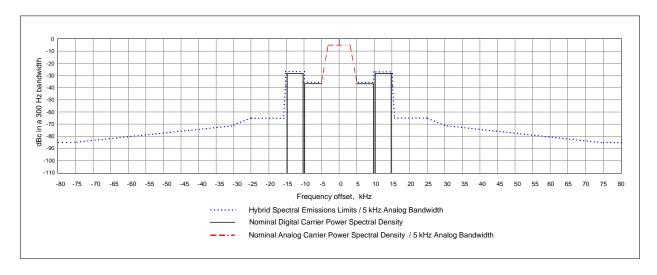


Figure 9. NRSC-5 AM hybrid waveform spectral emissions limits for 5 kHz mode analog bandwith

For hybrid transmissions utilizing 8 kHz mode analog bandwidth, noise and spuriously generated signals from all sources, including phase noise and intermodulation products, shall conform to the limits as described in Section 4.5.2 of normative reference [8] and shown in Figure 10.

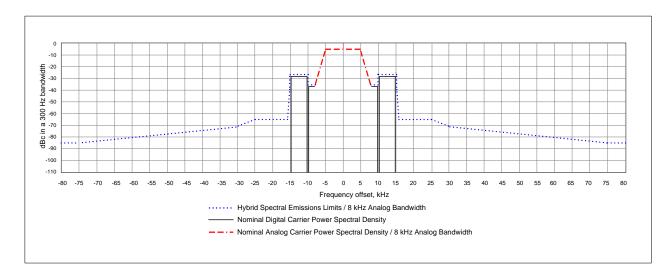


Figure 10. NRSC-5 AM hybrid waveform spectral emissions limits for 8 kHz mode analog bandwidth

For all-digital transmissions, noise and spuriously generated signals from all sources, including phase noise and intermodulation products, shall conform to the limits as described in Section 4.5.2 of normative reference [8] and shown in Figure 11.

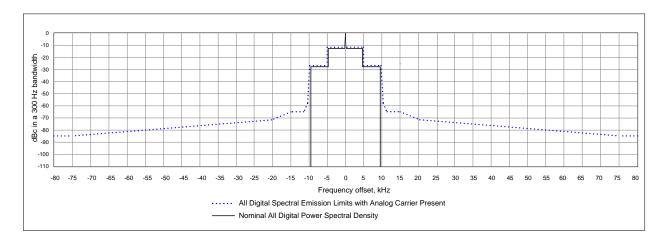


Figure 11. NRSC-5 AM all-digital waveform spectral emissions limits

4.1.8.1 Measurement of mask compliance for AM IBOC systems

Reference document [8] specifies a method for determining spectral emission mask compliance for AM hybrid (both 5 kHz mode and 8 kHz mode) and AM All Digital IBOC, in Sections 4.5.1, 4.5.2, and 4.5.3, respectively. For more detailed information on mask compliance measurements refer to NRSC Guideline G-201, "NRSC-5-B RF Mask Compliance: Measurement Methods and Practice." Included in this Guideline are detailed measurement procedures applicable to different types of measurements (e.g., factory test, in-service out-of-band emissions), as well as recommended locations for making measurements depending upon the specifics of a particular implementation.

4.2 FM RF/transmission System Characteristics

This section specifies the RF portion of the NRSC-5 IBOC digital radio standard for FM band implementations. Figure 12 illustrates how the standard is partitioned according to protocol layer and is annotated with the referenced documents that specify the associated detailed requirements. It is an overview of the entire FM band implementation of the NRSC-5 IBOC digital radio standard.

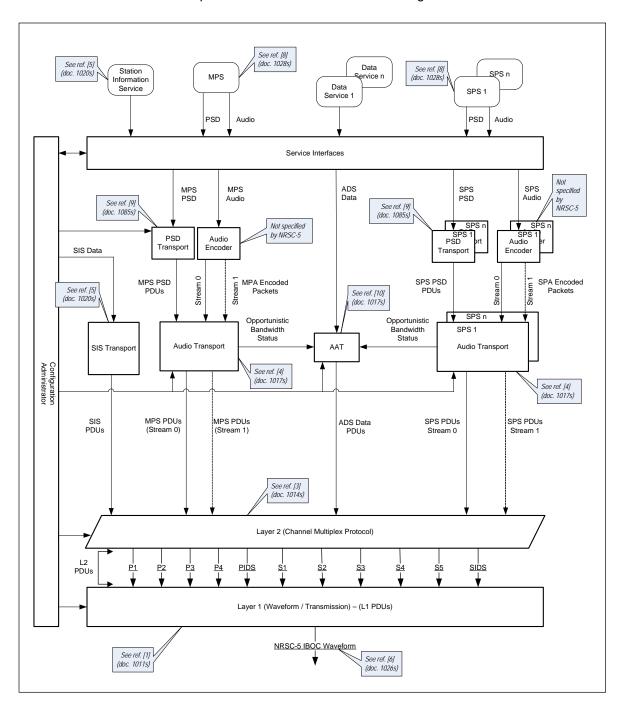


Figure 12: FM band implementation of NRSC-5 IBOC digital radio standard

4.2.1 Transmission Characteristics

This section includes a high-level description of each Layer 1 functional block and the associated signal flow. Figure 13 is a top level block diagram of the RF/transmission subsystem illustrating the order of processing therein. Figure 14 is a functional block diagram of Layer 1 processing. Audio and data are passed from the higher protocol layers to the physical layer, the modem, through the Layer 2 - Layer 1 interface.

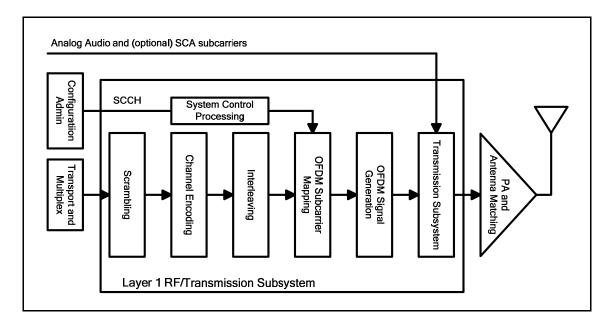


Figure 13: RF/transmission subsystem block diagram

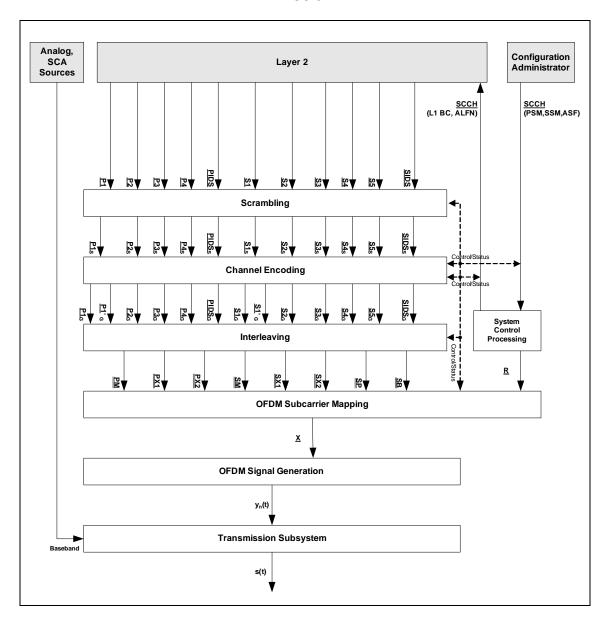


Figure 14: FM air interface Layer 1 functional block diagram with details of data flow illustrated

4.2.2 Layer 1 Interface

The L1 interface illustrates the access points between the channel multiplex and Layer 1 of the system protocol stack. Each channel enters Layer 1 in discrete transfer frames, with unique size and rate determined by the service mode. Transfer frames that carry information from the channel multiplex are referred to as L1 PDUs.

4.2.3 Logical Channels

The concept of logical channels and their function is central to the transport and transmission of data through the IBOC system. A logical channel is a signal path that conducts Layer 1 PDUs through Layer 1 with a specified grade of service. Logical channels are specified in normative reference [1]. In Figure 14 the logical channels are denoted by symbols such as P1, PIDS, S1, etc. The underscore indicates that the data in the logical channel is formatted as a vector.

4.2.4 Channel Coding

Channel coding (specified in normative references [1] and [2]) comprises the functions of scrambling, channel encoding, and interleaving shown in Figure 13.

4.2.4.1 Scrambling

This function randomizes the digital data in each logical channel to "whiten" and mitigate signal periodicities when the waveform is demodulated in a conventional analog FM demodulator. The bits in each logical channel are scrambled to randomize the time-domain data and aid in receiver synchronization. The inputs to the scramblers are the active logical channels from Layer 2, as selected by the service mode. The outputs of the scramblers are transfer frames of scrambled bits for each of the active logical channels. The scrambler generates a pseudorandom code which is modulo 2 summed with the input data vectors. The code generator is a linear feedback shift register.

4.2.4.2 Channel Encoding

Channel encoding improves system performance by increasing the robustness of the signal in the presence of channel impairments. The channel encoding process is characterized by two main operations: time delay (for diversity delay and transmit alignment) and convolutional encoding.

Depending on the service mode, logical channels P1 and S1 may be split into two channels and delayed as they enter the channel encoding process. The manner in which diversity delay is applied to these logical channels is specified in normative reference [1] for each service mode. The delay provides time diversity to the affected logical channels. If applied, the value of the diversity delay is a fixed value.

Punctured convolutional encoding is applied to each logical channel in the RF/transmission subsystem for forward error correction. Several different encoding polynomials and puncture matrices are used. Different logical channels have different code rates. The specification of the forward error correction coding used for each logical channel and each service mode is contained in normative reference [1].

4.2.4.3 Interleaving

Interleaving is also applied to the logical channels in the RF/transmission subsystem. Interleaving comprises six parallel interleaving processes (IPs): PM, PX, SM, SX, SP, and SB (see Figure 15). An IP contains one or more interleavers, and, in some cases, a transfer frame multiplexer. The service mode determines which inputs and IPs are active at any given time. The universe of inputs for interleaving are the channel-encoded transfer frames from the primary logical channels P1 through P4 and PIDS, and the secondary logical channels S1 through S5 and SIDS. The interleaver outputs are matrices. The interleaving processes for each service mode are specified in Section 10 of normative reference [1].

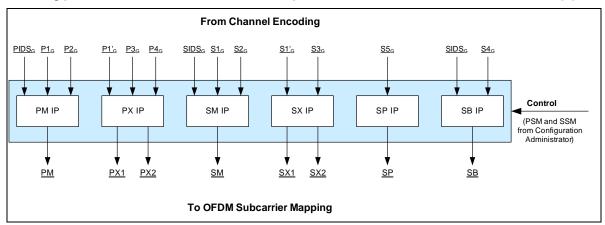


Figure 15: Interleaving conceptual block diagram

4.2.5 System Control Processing

As shown in Figure 14, the system control channel (SCCH) bypasses the channel coding. Under the direction of the system configuration settings, system control processing assembles and differentially encodes a sequence of bits (system control data sequence) destined for each reference subcarrier, as shown in Figure 16. There are up to 61 reference subcarriers, numbered 0 ... 60, distributed throughout the OFDM spectrum. The number of reference subcarriers broadcast in a given waveform depends on the service mode; however, system control processing always outputs all 61 system control data sequences, regardless of service mode. This processing is specified in Section 11 of normative reference [1].

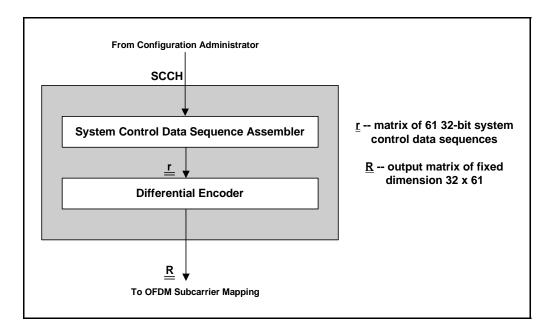


Figure 16: System control processing conceptual diagram

4.2.6 Subcarrier Mapping and Modulation

OFDM subcarrier mapping assigns interleaver partitions to frequency partitions. For each active interleaver matrix, OFDM subcarrier mapping assigns a row of bits from each interleaver partition to its respective frequency partition in the complex output vector \underline{X} . In addition, system control data sequence bits from a row of \underline{R} are mapped to the active reference subcarrier locations in \underline{X} . The service mode dictates which interleaver matrices and which elements of \underline{R} are active. Figure 17 shows the inputs, output, and component functions of OFDM subcarrier mapping.

The inputs to OFDM subcarrier mapping for each symbol are a row of bits from each active interleaver matrix and a row of bits from $\underline{\mathbb{R}}$, the matrix of system control data sequences. The output from OFDM subcarrier mapping for each OFDM symbol is a single complex vector, \underline{X} , of length 1093.

The interleaver matrices carrying the user audio and data (\underline{PM} , $\underline{PX1}$, ... \underline{SB}) are mapped to quadrature phase shift keying (QPSK) constellation points and to specific subcarriers. The R matrix is mapped to binary phase shift keying (BPSK) constellation points and the reference subcarriers. These phasors are then scaled in amplitude and then mapped to their assigned OFDM subcarriers. This process results in a vector, \underline{X} , of phasors which are output to the OFDM signal generation function. This processing is specified in Section 12 of normative reference [1].

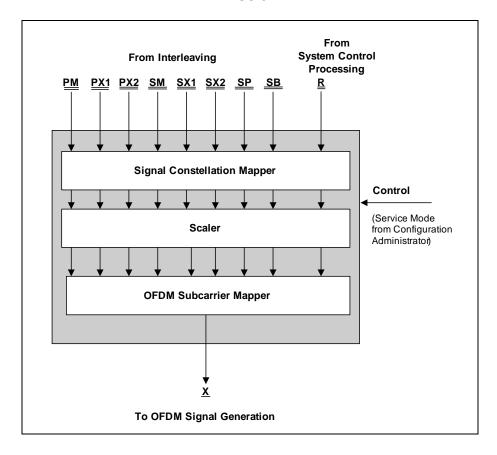


Figure 17: OFDM subcarrier mapping conceptual block diagram

4.2.7 Transmission

OFDM signal generation receives complex, frequency-domain OFDM symbols from OFDM subcarrier mapping, and outputs time-domain pulses representing the digital portion of the FM IBOC signal.

The input to OFDM signal generation for the n^{th} symbol is a complex vector \underline{X}_n of length L, representing the complex constellation values for each OFDM subcarrier in OFDM symbol n. For notational convenience, the output of OFDM subcarrier mapping described above did not use the subscript n. Rather, it referred to the vector \underline{X} as representing a single OFDM symbol. In this section, the subscript is appended to \underline{X} because of the significance of n to OFDM signal generation. The OFDM symbol is transformed to the time domain by a discrete Fourier transform and shaped to create one time domain symbol, $y_n(t)$. The output of OFDM signal generation is a complex, baseband, time-domain pulse $y_n(t)$, representing the digital portion of the FM IBOC signal for OFDM symbol n.

The $y_n(t)$ symbols are concatenated to form a continuous time domain waveform. This waveform is upconverted and combined with the analog modulated audio (in the hybrid and extended hybrid modes) to create the complete IBOC RF waveform for transmission. This is illustrated in Figure 18. The waveform is then spectrally mapped and frequency partitioned across the set of OFDM subcarriers. Normative reference [6] details the key transmission specifications for the FM IBOC RF waveform including carrier frequency and channel spacing, synchronization tolerances, spectral emission limits, digital sideband levels, phase noise, error vector magnitude, gain flatness, and group delay flatness.

There are several issues of time alignment that the transmission system must address. For transmit facilities so equipped, every L1 frame transmitted must be properly aligned with GPS time. Also, the various logical channels must be properly aligned with each other and in some service modes some channels are purposely delayed by a fixed amount to accommodate diversity combining at the receiver.

Layer 1 provides for the time alignment of the transfer frames received from the channel multiplex. The higher protocol layers provide alignment of the contents of the transfer frames.

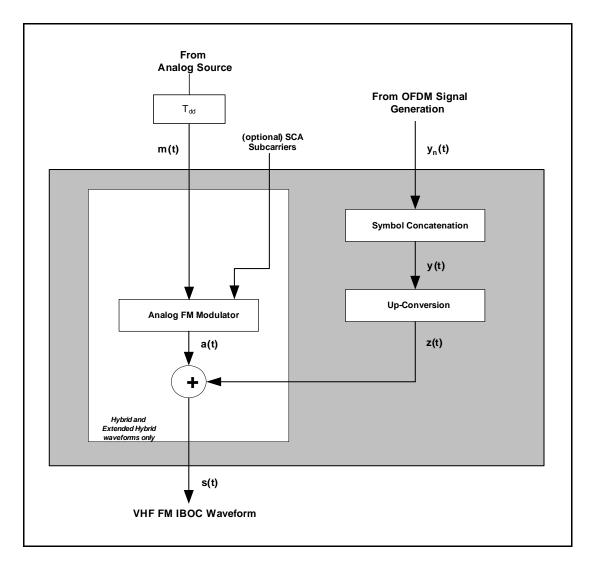


Figure 18: Hybrid/extended hybrid transmission subsystem functional block diagram

4.2.8 Spectrum Emissions Limits for FM IBOC

For hybrid transmissions, noise and spuriously generated signals from all sources, including phase noise and intermodulation products, shall conform to the limits as described in Section 4.4.1 of normative reference [6] and shown in Figure 19.⁴

⁴ The requirements for noise and spurious emission limits illustrated in figures 19 & 20 reflect acceptable performance criteria. In certain circumstances, additional measures (filtering, active emissions suppression, etc.) may be needed to reduce the spectral emissions below the limits given in this subsection in order to reduce mutual interference between broadcast stations.

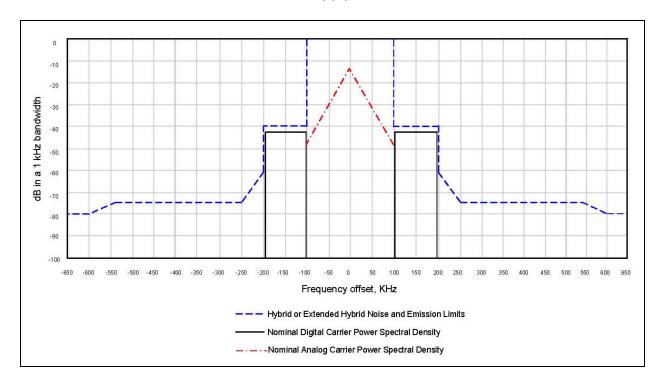


Figure 19. NRSC-5 FM hybrid waveform noise and emission limits

For all-digital transmissions, noise and spuriously generated signals from all sources, including phase noise and intermodulation products, shall conform to the limits as described in Section 4.4.2 of normative reference [6] and shown in Figure 20.

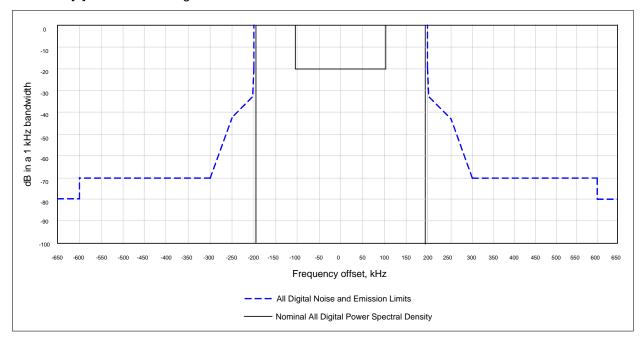


Figure 20. NRSC-5 all-digital waveform noise and emission limits

4.2.8.1 Measurement of mask compliance for FM IBOC systems

Reference document [6] specifies a method for determining spectral emission mask compliance for FM hybrid IBOC and All Digital IBOC, in Sections 4.4.1 and 4.4.2, respectively. For more detailed information on mask compliance measurements refer to NRSC Guideline G-201, "NRSC-5-B RF Mask Compliance: Measurement Methods and Practice." Included in this Guideline are detailed measurement procedures applicable to different types of measurements (e.g., factory test, in-service out-of-band emissions), as well as recommended locations for making measurements depending upon the specifics of a particular implementation.

5 TRANSPORT AND SERVICE MULTIPLEX CHARACTERISTICS

This section specifies how control signals and non-audio information are passed through the IBOC digital radio system up to, but not including the RF/transmission subsystem.

The IBOC digital radio broadcasting system specified by the NRSC-5 Standard allows a broadcast station to offer multiple services. A service can be thought of as a logical grouping of application data identified by the IBOC digital radio system. Services are grouped into one of three categories:

- 1. Core services:
 - a. Main Program Service (MPS), both audio (MPA) and data (MPSD)
 - b. Station Information Service (SIS)
- 2. Supplemental services:
 - a. Supplemental Program Service (SPS), both audio (SPA) and data (SPSD)
- 3. Advanced Data Services (ADS)

The flow of service content through the IBOC digital radio broadcast system is as follows:

- Service content enters the IBOC digital radio broadcast system via service interfaces;
- Content is assembled for transport using a specific protocol;
- 3. It is routed over logical channels via the channel multiplex;
- 4. And finally it is waveform modulated via the RF/transmission subsystem for over-the-air transmission.

Figure 2 is an overview of the AM band IBOC digital radio broadcast system and Figure 12 is an overview of the FM IBOC digital radio broadcast system. The following sections present a brief description of the IBOC digital radio core, supplemental, and advanced data services framework.

5.1 Core Services Overview

5.1.1 Main Program Service (MPS)

The main program service (specified in normative references [4] and [7]) is a direct extension of traditional analog radio. The IBOC format enables the transmission of existing analog radio programming in both analog and digital formats. This allows for a smooth transition from analog to digital radio. Radio receivers that are not IBOC digital radio enabled can continue to receive the traditional analog radio signal, while IBOC digital radio enabled receivers can receive both digital and analog signals via the same frequency band. In addition to digital audio, MPS includes digital data related to the audio programming. This is also referred to as MPS Data or Main Program Service Data (MPSD).

5.1.2 Station Information Service (SIS)

The Station Information Service (specified in normative reference [5]) provides the necessary radio station control and identification information, such as station call sign identification, time and location reference information. SIS can be considered a built-in service that is readily available on all IBOC digital radio stations.

5.2 Supplemental Services Overview

5.2.1 Supplemental Program Service (SPS)

The Supplemental Program Service (specified in normative references [4] and [7]) is a direct extension of MPS in FM IBOC digital radio transmissions. SPS allows the transmission of additional audio content in digital format. This allows for additional audio programs to be broadcast on the same carrier. Multiple (up to 7) SPS channels or programs may be transmitted simultaneously. The system allows broadcasters to reallocate capacity that could otherwise be used for MPS or data in order to allow for this configuration.

A station's broadcast of SPS will not affect a receiver's ability to receive traditional analog radio signals or MPS transmissions, even if the receiver is not SPS enabled. In addition to digital audio, SPS includes digital data related to the audio programming. This is also referred to as SPS Data or program service data (PSD).

5.3 Advanced Data Services (ADS)

Advanced data services provide broadcasters with the ability to transmit information that may be unrelated to MPS, SIS or SPS. These services can carry any form and content that can be expressed as a data file or a data stream, including audio services. Examples of such services include (i) visual effects associated with MPS, SIS, or SPS services; (ii) multimedia presentations of stock, news, weather, and entertainment programming including audio, text and images; (iii) broadcast updates to in-vehicle systems; (iv) local storage of content for time shifting and later replay; (v) targeted advertising; and (vi) traffic updates and information for use with navigation systems.

Advanced data services are carried on IBOC by the Advanced Application Services Transport (AAT). In addition to allowing multiple data applications to share the Waveform / Transmission medium, AAT provides a common transport mechanism. The detailed description of AAT is documented in normative reference [10].

The following sections describe the Advanced Data Services interfaces and transports.

5.3.1 Advanced Application Services Transport (AAT)

Figure 21 details the interface of the AAT for the NRSC-5 IBOC digital radio broadcasting system. The AAT is used in the transport of fixed and opportunistic data in the IBOC system. Various advanced data services use the Service Interfaces to interact with the IBOC system. During broadcast, the AAT receives ADS data from the Service Interfaces and then encodes and encapsulates this data to generate AAT PDUs. The AAT PDUs are then sent to Layer 2 via Bearer Channels for further processing [3]. The AAT PDUs are sent over different data channels which carry fixed data and opportunistic data packets. Fixed data bandwidth is established in advance by the station operator. The opportunistic data bandwidth depends on the audio content transmitted.

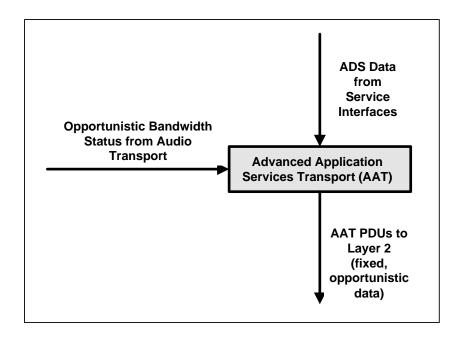


Figure 21. Advanced Data Services Protocol Interface Diagram

To send data over these channels, packets are encoded in a continuous byte stream. Successful packet delivery relies on the data channels to deliver the bytes in the same order that they were transmitted. Byte stream encoding consists of Packet Encapsulation with embedded error detection, and Forward Error Correction (FEC). The general structure of an encoded byte stream is shown below in Figure 22.

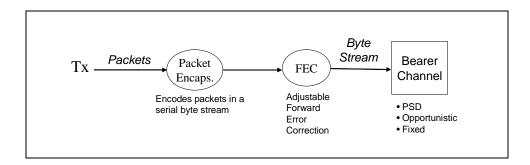


Figure 22. Byte Stream Encoding

5.3.1.1 Packet Encapsulation

The packet encapsulation used by the AAT follows the HDLC-like framing employed by the Point-to-Point Protocol (PPP) as standardized by the IETF in RFC-1662, "PPP in HDLC-Like Framing." The HDLC-like framing allows encapsulation of a packet referred to as an AAT PDU, within a byte stream, that may be sent in segments of arbitrary size (e.g., in each L1 frame). [1][2]

The contents of the encapsulated packet are described in Table 1.

Table 1. Description of AAT packet encapsulation

Flag	Delimiter to indicate start of next protocol data unit.	
Data Transport Packet Format	Identifier to define the format for the data packet.	
Port Identification	Identifier to indicate to the receiver which application is associated with the transmitted data.	
Sequence	An incremental field to track successive packets transmitted. Each port has an individual sequence.	
Payload	Application data. The payload may be of any size up to 8192 bytes.	
Frame Check Sequence	16-bit CRC used for error detection.	

5.3.1.2 Forward Error Correction (FEC)

Forward Error Correction allows for increased reliability in transmission of data applications.

5.3.1.2.1 Reed Solomon Coding

A Reed Solomon coder shall be included in the FEC to increase the robustness of the service by improving reliability of reception. The Reed Solomon coder shall have a codeword size of 255 bytes with no more than 64-bytes for parity. The amount of parity shall be determined by the desired reliability and managed by the Configuration Administrator in the broadcast system.

5.3.1.2.2 Interleaving

The FEC shall employ a convolutional byte interleaver. Interleaving randomized the occurrence of transmission errors resulting in improved reception. The byte interleaver shall map Reed Solomon code words (255 bytes) to an interleaver matrix. The depth of the matrix shall be no greater than 64 codewords, depending on desire transmission reliability and processing delay. The depth of the interleaver is managed by the Configuration Administrator in the broadcast system.

5.3.1.2.3 Block Synchronization

The FEC process operates on blocks of 255-bytes. A 4-byte block boundary marker (BBM) shall be regularly inserted into the stream to provide a method for receiver synchronization. The frequency of markers shall depend on the data transmission channel.

5.3.2 Transmission Channels

Encoded data packets (PDUs) shall be sent over different channels (Opportunistic or Fixed) to Layer 2 for further processing as shown in Figure 23.

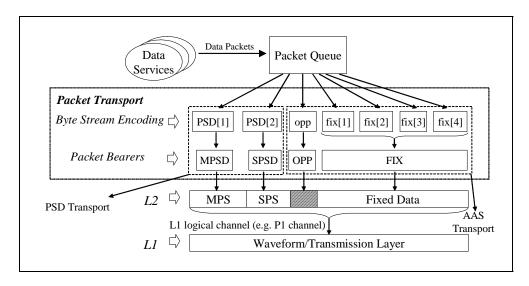


Figure 23. AAT Packet Transport Mechanism

Each transmission channel has different FEC configurations based on the level of reliability for the transmission as described in Table 2.

BBM Reed **Type Bearer Description** Interleaver Frequency Solomon Unused bytes allocated to audio programs. OPP 223:32 None 1:1 Variable capacity. 255:0 0 - 64Uses allocated segment(s) of L2 frame. **FIX** 1:4 "Infinitely" variable FEC. To 191:64 **Blocks**

Table 2. Description of transmission channel FEC configurations

5.3.2.1 Opportunistic Channel

The Opportunistic Channel may utilize unused capacity in the audio transport for data transmission. The size of the opportunistic payload is determined on the basis of whether the audio programs use their full allocated capacity.

A 5-byte delimiter shall be appended to the Opportunistic channel to mark the boundary between the data and the audio services.

The Reed Solomon encoder shall have a fixed parity of 32-bytes. The interleaver is not used for this channel.

5.3.2.2 Fixed Channel

When transmitted, fixed channel data shall be transmitted through any or all Layer 1 logical channels. Each fixed channel shall be subdivided into sub-channels (maximum 4). To allow for different levels of transmission reliability, sub-channels may have different FEC settings.

Reed Solomon parity shall be less than or equal to 64-bytes. Interleaver depth shall be less than or equal to 64 codewords. A block boundary marker shall occur every 4 codewords.

Each fixed channel shall contain synchronization and control information.

5.3.2.2.1 Synchronization

A single-byte synchronization word shall transmit the timing information and width of the Configuration Control.

5.3.2.2.2 Configuration Control

The Configuration Control shall send a repeating message describing the number, width, and FEC configuration of the fixed sub-channels. The message shall be encapsulated to provide framing and error detection.

5.4 Core Services Description

5.4.1 Main Program Service (MPS)

The following sections describe the MPS audio and data interfaces and transports.

5.4.1.1 MPS Audio

Figure 24 details the interface of the audio encoder for the IBOC digital radio system.⁵ The audio encoder receives input audio frames from the audio interface application and encodes the audio data. The encoded audio is combined with MPS data, and sent to the channel multiplex (Layer 2) as an audio codec protocol data unit (PDU). The audio codec PDU is comprised of compressed audio and PSD. This process is specified in normative reference [4].

⁵ This standard does not specify an encoder. In order to determine the system's viability the NRSC evaluated it using the HD Codec developed by iBiquity and Coding Technologies.

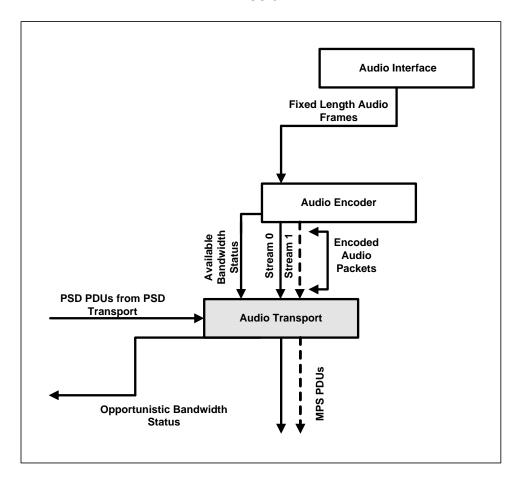


Figure 24: Audio codec protocol interface diagram

The audio frame, packet and PDU are defined in Table 3.

Table 3. Description of MPS audio frame, packet and PDU

Audio frame	The unit of information payload exchanged from the audio interface and the audio codec protocol layer. Audio frames are typically comprised of 1024 audio samples at a sampling rate of 44.1 kHz using 16 bit pulse code modulation (PCM).
Audio encoded packet	Compressed audio frames output from the audio encoder. These may be divided into one to three output streams depending on the audio encoder mode.
Audio codec PDU	This is the output of the audio transport process. An audio codec PDU consists of protocol information followed by a sequence of audio encoded packets. Audio codec PDUs may be output on from one to three streams depending on the audio encoder mode.

5.4.1.2 MPS Data Interface and Transport

The MPS allows data to be transmitted in tandem with program audio. The data is intended to describe or complement the audio program that the radio listener is hearing.

MPS data (fully specified in normative reference [7]) defines a specific set of data fields (e.g., artist, title, etc.). The fields can be used for all forms of audio programming. For example, the "title" field may immediately seem to apply only to song titles. However, it also applies to titles for commercials, announcements, and talk programs.

The core fields are described in Table 4.

Table 4. Core MPS data fields

Title	Equivalent to a one-line title name	
Artist	Performer, originator, author, sponsor, show host.	
Album	Content source , such as album name, show name, sponsor name	
Genre	Categorization of content. This is an enumerated field of predefined types, such as Jazz, Rock, Speech, etc.	
Comment Title	One-line title for comment description	
Comment Description	Detail description, user call-back or other information, such as talk show call in number or web-site URL.	
Commercial	Collection of fields that support detailed product advertisement and purchasing, including: - Price of merchandise - Expiration data for transaction - Transaction method - URL which could be used to initiate purchase transaction via an external return channel - Advertisement description - Seller identification	
Reference Identifiers	Identifiers that can be used to uniquely identify the MPS data message	

The MPS Data is transmitted within the audio transport. The SPS transport is identical to the MPS transport. Both services contain identical header and data structures. See normative references [7] and [9] for further details.

5.4.2 Station Information Service (SIS)

A description of the SIS interface and transport is provided in the following sections.

5.4.2.1 SIS Interface

The station information service interface allows broadcasters to transmit the information listed in Table 5.

Table 5. Description of information transmitted by SIS

Station Name (Short Format)	Station Name (Short Format) identifies the 4-alpha-character station call sign plus an optional extension.
Station Name (Long Format)	Station Name (Long Format) identifies the station call sign. For example, station WXXX or WYYY-FM. In addition, to traditional call sign identification, broadcasters can also transmit long station names such as MD SMOOTH JAZZ or WXYX ROCK N ROLL.
Station ID Number	Station ID Number is a unique numerical identifier that identifies the Federal Communications Commission (FCC) approved broadcast station. The Station ID Number is made up of the following information: i) Country Code, and ii) FCC facility ID.
Station Location	Station Location provides a 3-dimensional geographic station location, consisting of latitude, longitude, and altitude. It can, for example, be used for receiver position determination in advanced applications.
Absolute Layer 1 Frame Number (ALFN)	System Time can be derived from the ALFN. The system time can be used to provide accurate time clock synchronization for receivers. For IBOC digital radio stations synchronized to the Global Positioning System (GPS), ALFN reflects global GPS time.
SIS Parameter Message	SIS Parameter Message provides additional data to aid the receiver with position and time applications. This message includes data fields for leap second timing (GPS) and local time of day (time zone, daylight savings time setting, etc). IBOC digital radio receivers enabled with time-of-day applications use these parameters in combination with ALFN to compute the correct local time.
Station Message	Station Message allows a station to send an arbitrary text message.

5.4.2.2 SIS Transport

The SIS is sent through the IBOC digital radio system via a dedicated logical channel. The channel multiplex routes SIS content to a dedicated SIS logical channel in the RF/transmission subsystem. For more information on the SIS transport, see normative reference [5].

5.5 Channel Multiplex

The channel multiplex allows the IBOC digital radio system to support independent transports for the following services:

- 1. Main program service audio and MPS data
- 2. Supplemental program service audio and SPS data
- 3. Advanced application services
- 4. Station information service

The channel multiplex is aware of the audio transport configuration requirements (e.g., channel mapping and bandwidth requirements). This allows for a high-level of synchronization between the analog and digital program audio streams. The channel multiplex has the flexibility to dynamically allocate unused MPS/SPS bandwidth for advanced data services.

The station information service passes through the channel multiplex without any additional processing.

5.5.1 Interface to RF/transmission subsystem

The RF/transmission subsystem interface provides a group of logical channels. Each channel is distinguished by the following characteristics:

- Channel identifier
- Transfer frame size
- Transfer frame rate
- Channel robustness
- Channel latency

Depending on the RF/transmission service mode, the number of logical channels will vary. The channel multiplex is synchronized to the RF/transmission clock rate. The channel multiplex is signaled at each channel transmission opportunity by the RF /transmission subsystem. A complete transfer frame for each logical channel is delivered to the RF/transmission system for broadcast transmission.

For more information on the channel multiplex, see normative reference [3].

5.5.2 Configuration Administrator

Control of the IBOC digital radio system is handled by the "configuration administrator" function as shown in Figure 2 and Figure 12. The AM or FM modem mode (e.g., MP1, MP2 etc.), bandwidth allocations, and specific information being sent across the logical channels (e.g., P1, P2, etc.) are controlled by the configuration administrator. This function represents the processes for communicating conditions and settings to and among the various transports and functional blocks, and will vary from implementation to implementation—its structure and detail are not specified by NRSC-5.

6 AUDIO ENCODER CHARACTERISTICS

This section specifies some of the characteristics of audio codecs designed for use with the NRSC-5 IBOC digital radio system. As noted above, NRSC-5 does not include specifications for audio source coding and compression. Suitable audio source coding and compression systems will use perceptual audio coding or other appropriate technologies to reduce the bit rate required for description of audio signals.

6.1 Coding rates

Table 6 shows the default audio bit rates for each audio codec mode. Minimum bit rates are shown in Table 7.

Table 6. Default audio bit-rate as a function of audio codec mode6

Audio Codec Mode	Typical use	Stream Type	Nominal Bit Rate (kbps)
0000	FM hybrid, extended hybrid (MP1-MP3, MP11)	Core	96
0001	EM all digital (MDS)	Core	48
0001	FM all-digital (MP6)	Enhanced	48
0010	ANA but sid (NAAA)	Core	20
	AM hybrid (MA1)	Enhanced	16
	AM all digital (MA2)	Core	20
	AM all-digital (MA3)	Enhanced	20
0011	EM all digital (MDS)	Core	24
	FM all-digital (MP5)	Enhanced	72
1010	EM Supplemental Brogram Service	Core	22
	FM Supplemental Program Service	Enhanced	24
1101	FM Supplemental Program Service	Core	24

Table 7. Nominal and minimum bit rates as a function of audio codec mode⁷

Audio Codec Mode	Typical use	Nominal bit rate (kbps)	Minimum bit rate (kbps)†
0000	FM hybrid	96	24
0001	FM all-digital	96	24
0010	AM hybrid	36	20
0010	AM all-digital	40	20
0011	FM all-digital	96	24
1010	FM SPS	46	22
1101	FM SPS	24	12

†These values reflect useful rates. "0" rate could also be considered but that reflects an inactive code.

6.2 Opportunistic Data

When an audio encoder does not use all of the bytes allocated to its use of the layer 2 PDU, the unused capacity can be made available as "opportunistic data" capacity. Opportunistic data is allocated independently on each PDU, such that there is no guarantee of a delivery rate in a series of transmitted PDU's. The effective data rate of opportunistic data is highly dependent on the characteristics of the audio program and the resulting quantity of data required to represent the coded audio. Effective opportunistic data rates typically range from zero to several kilobits per second. The relationship between

⁶ From reference [4], section 5.2.1.1, Table 5-2.

⁷ From reference [4], section 5.2.1.1, Table 5-3.

fixed, opportunistic, and other data capacities on the layer 2 channel multiplex is described in detail in normative reference [3].

* * *

NRSC Document Improvement Proposal

If in the review or use of this document a potential change appears needed for safety, health or technical reasons, please fill in the appropriate information below and email, mail or fax to:

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